Problems in and among industries for the prompt realization of IoT and safety considerations
draft-baba-iot-problems-03

Abstract

This document contains opinions gathered from enterprises engaging in the IoT business as stated in the preceding version hereof, and also examines the possibilities of new social problems in the IoT era. Recognition of the importance of information security has grown in step with the rising use of the Internet. Closer examination reveals that the IoT era may see a new direct physical threat to users. For instance, the situation at a smart house may lead it to judge that the owner has only temporarily stepped out, causing it to unlock the front door, which in turn makes it easier for thieves to enter. These kinds of scenarios may occur without identity fraud, hacking, and other means of compromising information security. Therefore, for the purpose of this document, this issue shall be referred to as "IoT Safety" to distinguish it from Information Security.

We believe that it is necessary to deepen our understanding of these new IoT-related threats through discussion and ensure there are measures to address these threats in the future. At the same time, we must also coordinate these measures with the solutions to the problems described in the previous version of this document.

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1. Introduction

Many activities are progressing in various fields, such as the proposal of standards for creating an IoT world. There are also many reports that analyze and predict the benefits that IoT can bring to the economy and society. These developments remind us of the end of the 20th century, when the effect and impact of the Internet was actively debated.
The authors tried using the following approach to clarify the issues for the prompt realization of IoT. First, the players were conveniently divided into two groups: ICT industry players and Things industry players. Next, we met major players in the ICT industry and Things industry and asked about the challenges they faced and the challenges the other side faced in creating IoT.

The ICT industry players mentioned here include communication carriers, ICT equipment vendors, the Internet service providers, application vendors, and software houses. The Things industry players include home and housing equipment manufacturers, infrastructure providers such as railways companies and power companies, and manufacturers of home appliances such as air conditioners and refrigerators, which are also the ICT users.

This paper is principally a summary of the meetings results, and a presentation of the micro case studies about the challenges for realizing IoT services. It is not an overview of the IoT world or a macro-proposal intended to promote the benefits of IoT.

In addition, the revised version includes an examination of the possibilities of new direct physical threats in the IoT era that have not yet been seen. These threats should affect the safety of our bodies, lives, and "things," which includes property. For this reason, this issue shall be referred to as "IoT Safety" to distinguish it from Information Security for the purpose of this document.

2. Technical Challenges

2.1. Safety, Security and Privacy

2.1.1. Challenges in protecting lives and property from IoT-related threats (IoT Safety)

The introduction of IoT may generate threats to "Safety" through the actual operation of mechanical devices, in addition to the Information Security problems discussed in Section 2.1.2 below. For example, the spread of applications for visualizing electric power consumption allows for mischief in device operation without the use of identity fraud or hacking. In addition, there is the potential for problems to arise in the normal operation of individual devices that are not caused by abnormal current or voltage, another troubling aspect of the introduction of IoT. These issues cannot be resolved with ordinary information security measures for Network Layer 4 or lower. In another case, a command to an IoT device is proper by itself, but it may conflict with the other commands or its environmental status. Therefore, the authors consider it necessary
to have a system for interpreting the details of operations of many appliances and preventing operations according to the necessity in Layer 7 (what the authors tentatively call "Sekisyo").

These threats are categorized into three types: threat to physical safety; the threat of the failure or destruction of equipment and property; and the threat of impeding the proper performance of equipment. The following section introduces examples of the different threats.

2.1.1.1. Safety of body/life

Information on things such as the use of faucets and housing equipment, the locking of the front doors and windows, and the state of electric power consumption based on the smart meter is used by smart houses to regulate homes. This information is used to determine whether anyone is at home, and the electronic lock of the front door and windows is unlocked and a notice of absence is issued to a thief.

2.1.1.2. Safety of equipment

Air conditioners and other equipment that normally are not expected to be frequently started or stopped each a day can be caused to break down by repeatedly turning them on and off as many as hundreds of times a day.

2.1.1.3. Proper performance of equipment

Water heaters containing a hot well can be caused to operate erratically. This is done by frequently transmitting signals from the mischief application instead of operation panel to tell the water heater that only 10% of the normal amount of hot water is needed, leaving the water heater perpetually low on water.

2.1.2. Information Security

We have confirmed two viewpoints regarding the information security of services using IoT equipment and devices. The first is tangible information security involving the critical infrastructure. The second concerns the information security of individuals and homes.

In regards to information security involving the critical infrastructure, the basic policy in the past was to stay physically disconnected from an external network, such as the Internet, to ensure information security. However, because of the advance in the systems from proprietary communication protocols to open IP protocols to detect symptoms of problems and to remotely maintain a large
number of facilities spread over a wide area, connecting to an external network will become unavoidable to achieve various goals. In addition, it is clear that isolated networks are also subject to the same kind of risks, even though it is not directly connected to the outside. There is no major difference in the information security risks because isolated networks are already the target of international cyber terrorism, with internal crimes and targeted attacks occurring more frequently. Based on these reasons, the ICT security of the social infrastructure requires an extremely high level of information security.

Looking at the information security of micro units, such as individuals and homes, the improved convenience provided by the introduction of IoT will lead to greater risks. For example, there is a product available for connecting the entrance door to the network. In ICT security technology, increasing the key length of the encryption makes it much harder to break. But even if the latest information security technology is used when it is installed, the information security technology will become obsolete and even pose a risk about halfway through the twenty- to thirty-year lifetime of the entrance door. As has been explained in other items, the ICT sense of time is completely different from that of Things.

2.1.3. Privacy in acquiring data

The problem of privacy in handling acquired data is a huge challenge for companies promoting IoT. In addition, the ownership of this data poses yet another challenge.

For example, railway companies have installed many cameras for station security and for marketing beverage vending machines. This creates problems for personal identification and privacy. At the present time, the companies are processing the images in real time and do not store the images to avoid the problems.

Another huge challenge is the ownership of data. Up until now, there has been a divided debate on whether data belonged to the company or to the users. Likewise, the relationship inside a small user group is also extremely diverse and complicated. One specific example is of a company that had obtained permission from the head of the household to use the data when it carried out an HEMS trial. Later on, the spouse of the head of the household disagreed and as a result permission to use the data was withdrawn.
2.2. Challenges posed by data acquisition, data distribution, data management and data quantity

2.2.1. Traffic patterns

The manner in which data is acquired from and distributed to IoT equipment/devices differs immensely from the traffic patterns of the present Internet. The present form of the Internet focuses on distributing information, and its systems focus on effectively delivering contents to the users. On the other hand, routinely or temporarily sending or receiving data through a huge number of various sensors and devices presents a very different kind of Internet traffic. However, questions such as how much traffic will come from what kind of Things, and how will they superimpose each other have not been sufficiently studied. There is no concrete explanation about the backbone design and operation of traffic, and there have been many cases in which the unclear specifications for IoT traffic made the design difficult on the communication company side. There are many challenges related to the set up and management of IoT equipment. We have heard from the construction companies that the configuration of IoT equipment with a large number of sensors involves a lot of hard work.

2.2.2. Acquired mass data

It is necessary to develop a management method to reuse acquired data safely and effectively. Even now, there are occasional instances of the theft and leakage of social data (such as IDs) that can be used to identify individuals. In the IoT era, there will be mass data that can lead to Things, and the Things in turn will lead to individuals. There are IoT industry players who do not invest as much in ICT systems as government agencies and large companies do, and thus a management system to safely and effectively reuse the acquired data needs to be developed. The laws and regulations related to ID management differ vastly by country and region. These issues related to society and individuals are largely affected by differences in common sense, and therefore need to be localized.

2.2.3. Explosive increase and diversity of data

In the future IoT era, there are concerns about the explosive increase in data quantity and the diversity of data sent from sensors and IoT equipment. On the other hand, M2M communication does not require mass data like images, and an extraordinary increase in traffic will be unlikely despite the increase in the number of sensors.
If data is sent from all Things, there will be an infinite number of
different kinds of data. In addition, with the present form of
Internet traffic, data is received by people, and most of it consists
of video or image downloads. The download traffic is several times
greater than that of the upload traffic. If there is a tremendous
increase in the use of IoT, such as M2M communication, the difference
between upload and download traffic will probably not be that much.
It might be necessary to fundamentally review the network and in
particular the last mile characteristics. The importance of this
issue is not yet widely recognized.

2.3. Mapping of the physical world and the virtual world

2.3.1. Physically handling acquired data

The acquired data simply represents certain kinds of digital value,
and it is important to uncover the meaning of this data. As
described previously, configuration of IoT equipment, such as the
large number of installed sensors, requires a lot of hard work. An
even greater amount of effort will be needed to determine the meaning
of the data and connect it to the physical world.

In energy management experiments, data is mapped manually. This is a
time consuming process, and one that is prone to human error. Cases
that rely on the use of human hands require the configuration of
automated setting systems to reduce labor, costs, and human errors to
introduce IoT

2.3.2. Data calibration

Another important thing is calibration. This involves properly
linking the data sent from Things to the Things concerned, and
correctly indicating the operating conditions.

It may be necessary to have a tool to treat this problem concerning
continuation of operation and the one pertaining to introduction of
IoT described previously as a package.

2.4. Product lifetime, generation management, and the cost of equipment
updates

2.4.1. Product lifetime

The life of most ICT equipment is about 5 years or less, while the
life of IoT equipment and devices is at least 10 years. There is a
clear gap between these two types of equipment.
In the example of the entrance door connected to the network mentioned earlier, the door is often used for about twenty to thirty years after installed. If it is connected to a network, the communication technology and communication service will most likely have undergone numerous generation changes in that twenty- to thirty-year time span. This presents a large gap between the ICT industry and the Things industry.

A solution to this problem that was reached during the meeting with the housing equipment manufacturers is that with the automatic control of multiple shutters in a building, the portion between the controller and the multiple shutters, the so-called mature technology, can be placed under the control of the shutter manufacturers, while the controller connected to the network will deal with the generation changes of the communication service.

2.4.2. Introducing IoT equipment into commodity equipment

It costs a lot to make the many different types of commodity equipment popular around the world usable as IoT equipment and devices. There are two ways to change commodity equipment into IoT equipment. One way is to convert it to IoT compatible equipment. The other way involves adding devices to commodity equipment. There are costs in both cases, and it will take a long time to introduce IoT unless different incentives are offered to help to overcome the burden of cost.

2.5. Too many related standards and the speed of standardization

2.5.1. Too many related standards

There are many standards related to IoT equipment and devices. There are multiple standards, technologies and services for communication technology, such as Bluetooth, Wi-Fi, NFC, and LTE, and it is difficult to choose which to apply.

The Things industry players do not always have the communication technology professionals needed for IoT. In the meeting, we learned that many companies were uncertain and hesitant about fields outside their own area of expertise. On the other hand, technological competition will improve quality as well as the level of completion, and thus will be beneficial for users.

In the future, a consulting business for clarifying ICT technology for the Things industry players may emerge. If there is a system that can interconnect multiple standards, it will accelerate the Things industry to enter IoT.
2.5.2. Speed of standardization

The concept of product life in ICT industry is completely different from that of the Things industry, and as a result the concept of standardization also varies greatly. Before standardization occurs in the ICT industry, many different proposals are made, from which the best are selected. The final decision often changes, and products have to be updated in order to follow the changes in standards. But in the Things industry, the standards have to remain unchanged for as long as possible because of the long product lifetimes. Therefore, it takes a long time to determine when a particular standard has become mature. When the Things industry goes to implement a standard from the ICT industry, it feels that the standard is incredibly fluid and seemingly undecided. Furthermore, the standardization process of the two industries is very different, and making it difficult to work on the other side when trying to determine a standard.

2.6. Interoperability, fault isolation, and total quality assurance

2.6.1. Interoperability

The verification of interoperability poses a major challenge because of the configuration used by multi-vendors. In addition to interoperability between equipment, the ability to ensure backward compatibility is also important for bringing about the IoT world.

If these capabilities cannot be provided, it will be very difficult to create an IoT world in which past products can function.

2.6.2. Fault isolation

The method for fault isolation that may occur presents another challenge.

Many PC users have experienced various kinds of problems. When their PC experiences a problem, they have to isolate the faults by themselves, with no one available to lend a helping hand.

In the IoT world, these issues become more difficult and complicated. For example, a smart home is equipped with air conditioners, kitchen supplies, and doors connected to the Internet. A problem that occurs in the smart home poses a much more serious problem to end users than an e-mail failure or problem with a PC.

If users are left to isolate the fault on their own, they may not know which manufacturer they contact for repairs if they are unable to isolate the fault on their own, or the manufacturer may refuse to
perform repairs because they fall outside the scope of their responsibility. As can be seen, the issue is an important challenge that will determine whether the B2C specific IoT world can be established.

2.6.3. Quality assurance

The quality assurance of individual pieces of IoT equipment does not guarantee the total quality of IoT. Since IoT involves connecting multiple Things and communication, it is natural to assume that the total service quality will depend on the quality of the IoT equipment and devices, which can sometimes become bottleneck. However, users are not aware of this.

As was mentioned previously in Section 2.6 issues that are not directly related to the quality of an individual component can be important factors in determining the quality of the service. In this way, the quality of IoT is not decided by each individual Thing, but needs to be considered as a service spread across the network.

2.7. Product design policy

2.7.1. Changes in design policy

The design policy has to be changed from placing emphasis on the high functionality of a single product to stressing the singular function of individual products as well as how they work in coordination with other products. For many years, the Things industry has focused on producing high functionality products with added value. But in the IoT era, the implicit assumption is to confine Things to their basic function and enhance the level of coordination between Things, rather than focusing on the added value. Simplified Things must be able to be controlled with an external application that can also be used by the Things of cross manufacturers.

Given this situation, the Things industry faces the challenge of adopting a completely different policy. During the meeting with the manufacturing industries, we could sense their difficulty in understanding and recognizing the need to change the policy.

2.8. Various technology restrictions within actual usage

2.8.1. Using radio waves

There are many cases that have provided us with insight about issues related to the use of radio waves in IoT (such as the wave traveling range and whether or not it travels further than stated in assumptions available). The suppliers or providers who configure IoT
are not always wave communication technology experts. People who are unfamiliar with radio waves seem to think that waves travel from antenna to antenna in a straight line, and that they can be blocked by obstacles. As a result, they often ask questions about how many meters radio waves can travel or whether radio waves can actually travel. Few people understand the fact that the emitted radio waves are reflected from various locations and are superimposed at the reception point where they are received, or that depending on how waves are reflected a change in the reception signal intensity, called fading, may occur. The lack of engineers who can advise on specialized matters such as these poses a major obstacle.

2.8.2. Batteries

The power capacity and lifetime of batteries represent another set of challenges similar in nature to the issue of radio waves traveling distance. There are questions such as the difference between the real and catalog specifications, as well as factors that affect the battery power capacity. The IoT providers, who are also users of IoT, have to solve these issues, while these are difficult problems even for experts.

2.8.3. Wiring

The incredible amount of wiring and its complexity (power lines and communication lines) pose major challenges. The complexity of wiring—such as the large number of sensors and equipment, the power lines that drive them, and the communication lines that connect them to the network for acquiring information—is to the point that people doing IoT installation work will start wishing for a wire harness. In addition, the installation of cables and electric work are often done by different engineers. This makes the issue even more complicated.

2.8.4. Being open

A single company alone cannot make all the commodities for IoT. The IoT world needs to be open, and this can only be achieved with the cooperation of many different industries. Up until now, companies in the Things industry have developed products in a closed loop process, seeking to capture users with their company’s own products. For this reason, they lack an open design concept of interoperability. Today, an entirely new design concept is needed to design products that can interconnect with the products of other companies.
3. Non-technical Challenges

3.1. Changing the product paradigm

3.1.1. Ecosystems

While the goal of setting up IoT is to generate new value, it may actually lead to the destruction of the ecosystems in which industries operate. In the IoT era, the traditional vertically integrated way of producing Things in manufacturing industries will consume too much time and cost. This approach also makes it difficult to incorporate the ideas of other cultures. The need for paradigm shift is easy to understand, but difficult to implement. Promoting this shift will pose a management challenge that requires a considerable amount of skill and effort to overcome.

3.1.2. Coordination and significant changes in strategy

It will become necessary to run businesses jointly with new partners, as well as cooperate and work in coordination with other industries and competitors. This issue—even when it is fully understood—will be very difficult to address and put into practice.

We have seen instances in which only a limited amount of information was given when parties exchanged opinions. There have also been instances in which communication was difficult because of differences in terminology and culture.

3.1.3. Competition with existing industries

The issue of competition with existing industries often arises when attempts are made to change or reform a business model change or reform. This issue can also be viewed as the reorganization of industries, rather than competition between existing industries. However, this realignment of industries is difficult to move forward in the absence of supervisors.

3.2. Benefits

3.2.1. Rising costs and monetization

Introducing IoT within products will cause costs to go up, and yet the benefits it provides are unclear. There is no specific killer application available, and the number of users will not rise immediately. Therefore, finding a way to make the business profitable will be very difficult. This issue is especially difficult for businesses and products that rely on cost reductions to deliver low prices that make them competitive.
3.3. Information security and privacy of social systems

3.3.1. Classification of ownership, location, and the usage of data

There are many questions regarding the wide variety of data gathered from IoT equipment, including questions related to ownership, storage location, and the authorization to grant a license to use data. These need to be addressed so that the system and equipment can be accepted by society.

For example, if a company installs a door in a house that gathers data on the opening and closing of the door, questions about the data will arise. Does it belong to the users or the company? Can another company use this data?

3.4. Disclosure of data

3.4.1. Side effects and malicious use potentially caused by the disclosure of data

For example, it has been shown that the electricity smart meter can lead to burglary because it shows when electricity is used and not used, providing an indication of the time when no one is home. This particular example demonstrates the importance of ensuring information security and privacy.

3.5. Preparing social support

3.5.1. Regulations

Systems of laws and regulations are important for ensuring the safety of the conventional products, but they can also be a barrier for innovation.

IoT can be affected by laws and regulations at home and abroad, and can also be influenced by regulations that extend across multiple countries. Regulatory authorities need to monitor IoT carefully and adjust the regulations and laws they oversee in a way that does not negatively impact the global competition environment.

3.5.2. Corporate social responsibility

In addition to pursuing profit, companies that promote IoT also need to improve the benefits offered to users and society.
3.5.3. Customization for individual customers

There is an ongoing shift in demand away from general products to customized products for individual customers. This could also be viewed as a shift away from manufacturing businesses to service businesses. IoT will play an important role in this shift.

Instead of manufacturing Things through mass production, it will be easier to customize a product by moving some of the functions to an application. Likewise, the manufacturing business also needs to move forward with the previously mentioned paradigm shift in order to achieve customization.

3.5.4. IoT literacy of the users

Because Things are connected to the network, apps will need to be created. Some of these will serve as the interface with which people interact with IoT.

In the IoT era of the future, users will need to possess a certain amount of knowledge about IoT apps.

3.5.5. Individual vs. family

The issue of whether the data of Things in the house belongs to the family or the individual will largely affect data analysis and the handling of privacy.

As was mentioned in Section 2.1.2, the spouse could later object to the head of the household granting authorization to use data.

4. Information Security Considerations

Meetings with the players in various IoT fields provided insight into information security issues. These issues are described in the following sections.

- Section 2.1.2 Physical damper of devices
- Section 2.1.2 Product lifetime and encryption strength

For details, please see the corresponding text.

5. Privacy Considerations

Similarly, issues regarding privacy are described in the following sections.
6. Acknowledgments

We would like to thank the foundation the promotion of industrial science and its RC-88 member companies for their cooperation.

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This document contains opinions gathered from enterprises engaging in the IoT business as stated in the preceding version hereof, and also examines the possibilities of new social problems in the IoT era. Recognition of the importance of information security has grown in step with the rising use of the Internet. Closer examination reveals that the IoT era may see a new direct physical threat to users. For instance, the situation at a smart house may lead it to judge that the owner has only temporarily stepped out, causing it to unlock the front door, which in turn makes it easier for thieves to enter. These kinds of scenarios may occur without identity fraud, hacking, and other means of compromising information security. Therefore, for the purpose of this document, this issue shall be referred to as "IoT Safety" to distinguish it from Information Security.

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1. Introduction

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In addition, the revised version includes an examination of the possibilities of new direct physical threats in the IoT era that have not yet been seen. These threats should affect the safety of our bodies, lives, and "things," which includes property. For this reason, this issue shall be referred to as "IoT Safety" to distinguish it from Information Security for the purpose of this document.

For the past few years, we got new findings through COMMA House, the experimental smart house owned by The University of Tokyo. Therefore, we will add new topics to the next version.

2. Technical Challenges

2.1. Safety, Security and Privacy

2.1.1. Challenges in protecting lives and property from IoT-related threats (IoT Safety)

The introduction of IoT may generate threats to "Safety" through the actual operation of mechanical devices, in addition to the Information Security problems discussed in Section 2.1.2 below. For example, the spread of applications for visualizing electric power consumption allows for mischief in device operation without the use of identity fraud or hacking. In addition, there is the potential for problems to arise in the normal operation of individual devices that are not caused by abnormal current or voltage, another troubling aspect of the introduction of IoT. These issues cannot be resolved
with ordinary information security measures for Network Layer 4 or lower. In another case, a command to an IoT device is proper by itself, but it may conflict with the other commands or its environmental status. Therefore, the authors consider it necessary to have a system for interpreting the details of operations of many appliances and preventing operations according to the necessity in Layer 7 (what the authors tentatively call "Sekisyo").

These threats are categorized into three types: threat to physical safety; the threat of the failure or destruction of equipment and property; and the threat of impeding the proper performance of equipment. The following section introduces examples of the different threats.

2.1.1.1. Safety of body/life

Information on things such as the use of faucets and housing equipment, the locking of the front doors and windows, and the state of electric power consumption based on the smart meter is used by smart houses to regulate homes. This information is used to determine whether anyone is at home, and the electronic lock of the front door and windows is unlocked and a notice of absence is issued to a thief.

2.1.1.2. Safety of equipment

Air conditioners and other equipment that normally are not expected to be frequently started or stopped each a day can be caused to break down by repeatedly turning them on and off as many as hundreds of times a day.

2.1.1.3. Proper performance of equipment

Water heaters containing a hot well can be caused to operate erratically. This is done by frequently transmitting signals from the mischief application instead of operation panel to tell the water heater that only 10% of the normal amount of hot water is needed, leaving the water heater perpetually low on water.

2.1.2. Information Security

We have confirmed two viewpoints regarding the information security of services using IoT equipment and devices. The first is tangible information security involving the critical infrastructure. The second concerns the information security of individuals and homes.

In regards to information security involving the critical infrastructure, the basic policy in the past was to stay physically
disconnected from an external network, such as the Internet, to ensure information security. However, because of the advance in the systems from proprietary communication protocols to open IP protocols to detect symptoms of problems and to remotely maintain a large number of facilities spread over a wide area, connecting to an external network will become unavoidable to achieve various goals. In addition, it is clear that isolated networks are also subject to the same kind of risks, even though it is not directly connected to the outside. There is no major difference in the information security risks because isolated networks are already the target of international cyber terrorism, with internal crimes and targeted attacks occurring more frequently. Based on these reasons, the ICT security of the social infrastructure requires an extremely high level of information security.

Looking at the information security of micro units, such as individuals and homes, the improved convenience provided by the introduction of IoT will lead to greater risks. For example, there is a product available for connecting the entrance door to the network. In ICT security technology, increasing the key length of the encryption makes it much harder to break. But even if the latest information security technology is used when it is installed, the information security technology will become obsolete and even pose a risk about halfway through the twenty- to thirty-year lifetime of the entrance door. As has been explained in other items, the ICT sense of time is completely different from that of Things.

2.1.3. Privacy in acquiring data

The problem of privacy in handling acquired data is a huge challenge for companies promoting IoT. In addition, the ownership of this data poses yet another challenge.

For example, railway companies have installed many cameras for station security and for marketing beverage vending machines. This creates problems for personal identification and privacy. At the present time, the companies are processing the images in real time and do not store the images to avoid the problems.

Another huge challenge is the ownership of data. Up until now, there has been a divided debate on whether data belonged to the company or to the users. Likewise, the relationship inside a small user group is also extremely diverse and complicated. One specific example is of a company that had obtained permission from the head of the household to use the data when it carried out an HEMS trial. Later on, the spouse of the head of the household disagreed and as a result permission to use the data was withdrawn.
2.2. Challenges posed by data acquisition, data distribution, data management and data quantity

2.2.1. Traffic patterns

The manner in which data is acquired from and distributed to IoT equipment/devices differs immensely from the traffic patterns of the present Internet. The present form of the Internet focuses on distributing information, and its systems focus on effectively delivering contents to the users. On the other hand, routinely or temporarily sending or receiving data through a huge number of various sensors and devices presents a very different kind of Internet traffic. However, questions such as how much traffic will come from what kind of Things, and how will they superimpose each other have not been sufficiently studied. There is no concrete explanation about the backbone design and operation of traffic, and there have been many cases in which the unclear specifications for IoT traffic made the design difficult on the communication company side. There are many challenges related to the set up and management of IoT equipment. We have heard from the construction companies that the configuration of IoT equipment with a large number of sensors involves a lot of hard work.

2.2.2. Acquired mass data

It is necessary to develop a management method to reuse acquired data safely and effectively. Even now, there are occasional instances of the theft and leakage of social data (such as IDs) that can be used to identify individuals. In the IoT era, there will be mass data that can lead to Things, and the Things in turn will lead to individuals. There are IoT industry players who do not invest as much in ICT systems as government agencies and large companies do, and thus a management system to safely and effectively reuse the acquired data needs to be developed. The laws and regulations related to ID management differ vastly by country and region. These issues related to society and individuals are largely affected by differences in common sense, and therefore need to be localized.

2.2.3. Explosive increase and diversity of data

In the future IoT era, there are concerns about the explosive increase in data quantity and the diversity of data sent from sensors and IoT equipment. On the other hand, M2M communication does not require mass data like images, and an extraordinary increase in traffic will be unlikely despite the increase in the number of sensors.
If data is sent from all Things, there will be an infinite number of different kinds of data. In addition, with the present form of Internet traffic, data is received by people, and most of it consists of video or image downloads. The download traffic is several times greater than that of the upload traffic. If there is a tremendous increase in the use of IoT, such as M2M communication, the difference between upload and download traffic will probably not be that much. It might be necessary to fundamentally review the network and in particular the last mile characteristics. The importance of this issue is not yet widely recognized.

2.3. Mapping of the physical world and the virtual world

2.3.1. Physically handling acquired data

The acquired data simply represents certain kinds of digital value, and it is important to uncover the meaning of this data. As described previously, configuration of IoT equipment, such as the large number of installed sensors, requires a lot of hard work. An even greater amount of effort will be needed to determine the meaning of the data and connect it to the physical world.

In energy management experiments, data is mapped manually. This is a time consuming process, and one that is prone to human error. Cases that rely on the use of human hands require the configuration of automated setting systems to reduce labor, costs, and human errors to introduce IoT.

2.3.2. Data calibration

Another important thing is calibration. This involves properly linking the data sent from Things to the Things concerned, and correctly indicating the operating conditions.

It may be necessary to have a tool to treat this problem concerning continuation of operation and the one pertaining to introduction of IoT described previously as a package.

2.4. Product lifetime, generation management, and the cost of equipment updates

2.4.1. Product lifetime

The life of most ICT equipment is about 5 years or less, while the life of IoT equipment and devices is at least 10 years. There is a clear gap between these two types of equipment.
In the example of the entrance door connected to the network mentioned earlier, the door is often used for about twenty to thirty years after installed. If is connected to a network, the communication technology and communication service will most likely have undergone numerous generation changes in that twenty- to thirty-year time span. This presents a large gap between the ICT industry and the Things industry.

A solution to this problem that was reached during the meeting with the housing equipment manufacturers is that with the automatic control of multiple shutters in a building, the portion between the controller and the multiple shutters, the so-called mature technology, can be placed under the control of the shutter manufacturers, while the controller connected to the network will deal with the generation changes of the communication service.

### 2.4.2. Introducing IoT equipment into commodity equipment

It costs a lot to make the many different types of commodity equipment popular around the world usable as IoT equipment and devices. There are two ways to change commodity equipment into IoT equipment. One way is to convert it to IoT compatible equipment. The other way involves adding devices to commodity equipment. There are costs in both cases, and it will take a long time to introduce IoT unless different incentives are offered to help to overcome the burden of cost.

### 2.5. Too many related standards and the speed of standardization

#### 2.5.1. Too many related standards

There are many standards related to IoT equipment and devices. There are multiple standards, technologies and services for communication technology, such as Bluetooth, Wi-Fi, NFC, and LTE, and it is difficult to choose which to apply.

The Things industry players do not always have the communication technology professionals needed for IoT. In the meeting, we learned that many companies were uncertain and hesitant about fields outside their own area of expertise. On the other hand, technological competition will improve quality as well as the level of completion, and thus will be beneficial for users.

In the future, a consulting business for clarifying ICT technology for the Things industry players may emerge. If there is a system that can interconnect multiple standards, it will accelerate the Things industry to enter IoT.
2.5.2. Speed of standardization

The concept of product life in ICT industry is completely different from that of the Things industry, and as a result the concept of standardization also varies greatly. Before standardization occurs in the ICT industry, many different proposals are made, from which the best is selected. The final decision often changes, and products have to be updated in order to follow the changes in standards. But in the Things industry, the standards have to remain unchanged for as long as possible because of the long product lifetimes. Therefore, it takes a long time to determine when a particular standard has become mature. When the Things industry goes to implement a standard from the ICT industry, it feels that the standard is incredibly fluid and seemingly undecided. Furthermore, the standardization process of the two industries is very different, and making it difficult to work on the other side when trying to determine a standard.

2.6. Interoperability, fault isolation, and total quality assurance

2.6.1. Interoperability

The verification of interoperability poses a major challenge because of the configuration used by multi-vendors. In addition to interoperability between equipment, the ability to ensure backward compatibility is also important for bringing about the IoT world.

If these capabilities cannot be provided, it will be very difficult to create an IoT world in which past products can function.

2.6.2. Fault isolation

The method for fault isolation that may occur presents another challenge.

Many PC users have experienced various kinds of problems. When their PC experiences a problem, they have to isolate the faults by themselves, with no one available to lend a helping hand.

In the IoT world, these issues become more difficult and complicated. For example, a smart home is equipped with air conditioners, kitchen supplies, and doors connected to the Internet. A problem that occurs in the smart home poses a much more serious problem to end users than an e-mail failure or problem with a PC.

If users are left to isolate the fault on their own, they may not know which manufacturer they contact for repairs if they are unable to isolate the fault on their own, or the manufacturer may refuse to perform repairs because they fall outside the scope of their
responsibility. As can be seen, the issue is an important challenge that will determine whether the B2C specific IoT world can be established.

2.6.3. Quality assurance

The quality assurance of individual pieces of IoT equipment does not guarantee the total quality of IoT. Since IoT involves connecting multiple Things and communication, it is natural to assume that the total service quality will depend on the quality of the IoT equipment and devices, which can sometimes become bottleneck. However, users are not aware of this.

As was mentioned previously in Section 2.6 issues that are not directly related to the quality of an individual component can be important factors in determining the quality of the service. In this way, the quality of IoT is not decided by each individual Thing, but needs to be considered as a service spread across the network.

2.7. Product design policy

2.7.1. Changes in design policy

The design policy has to be changed from placing emphasis on the high functionality of a single product to stressing the singular function of individual products as well as how they work in coordination with other products. For many years, the Things industry has focused on producing high functionality products with added value. But in the IoT era, the implicit assumption is to confine Things to their basic function and enhance the level of coordination between Things, rather than focusing on the added value. Simplified Things must be able to be controlled with an external application that can also be used by the Things of cross manufacturers.

Given this situation, the Things industry faces the challenge of adopting a completely different policy. During the meeting with the manufacturing industries, we could sense their difficulty in understanding and recognizing the need to change the policy.

2.8. Various technology restrictions within actual usage

2.8.1. Using radio waves

There are many cases that have provided us with insight about issues related to the use of radio waves in IoT (such as the wave traveling range and whether or not it travels further than stated in assumptions available). The suppliers or providers who configure IoT are not always wave communication technology experts. People who are
unfamiliar with radio waves seem to think that waves travel from antenna to antenna in a straight line, and that they can be blocked by obstacles. As a result, they often ask questions about how many meters radio waves can travel or whether radio waves can actually travel. Few people understand the fact that the emitted radio waves are reflected from various locations and are superimposed at the reception point where they are received, or that depending on how waves are reflected a change in the reception signal intensity, called fading, may occur. The lack of engineers who can advise on specialized matters such as these poses a major obstacle.

2.8.2. Batteries

The power capacity and lifetime of batteries represent another set of challenges similar in nature to the issue of radio waves traveling distance. There are questions such as the difference between the real and catalog specifications, as well as factors that affect the battery power capacity. The IoT providers, who are also users of IoT, have to solve these issues, while these are difficult problems even for experts.

2.8.3. Wiring

The incredible amount of wiring and its complexity (power lines and communication lines) pose major challenges. The complexity of wiring—such as the large number of sensors and equipment, the power lines that drive them, and the communication lines that connect them to the network for acquiring information—is to the point that people doing IoT installation work will start wishing for a wire harness. In addition, the installation of cables and electric work are often done by different engineers. This make the issue even more complicated.

2.8.4. Being open

A single company alone cannot make all the commodities for IoT. The IoT world needs to be open, and this can only be achieved with the cooperation of many different industries. Up until now, companies in the Things industry have developed products in a closed loop process, seeking to capture users with their company’s own products. For this reason, they lack an open design concept of interoperability. Today, an entirely new design concept is needed to design products that can interconnect with the products of other companies.
3. Non-technical Challenges

3.1. Changing the product paradigm

3.1.1. Ecosystems

While the goal of setting up IoT is to generate new value, it may actually lead to the destruction of the ecosystems in which industries operate. In the IoT era, the traditional vertically integrated way of producing Things in manufacturing industries will consume too much time and cost. This approach also makes it difficult to incorporate the ideas of other cultures. The need for paradigm shift is easy to understand, but difficult to implement. Promoting this shift will pose a management challenge that requires a considerable amount of skill and effort to overcome.

3.1.2. Coordination and significant changes in strategy

It will become necessary to run businesses jointly with new partners, as well as cooperate and work in coordination with other industries and competitors. This issue—even when it is fully understood—will be very difficult to address and put into practice.

We have seen instances in which only a limited amount of information was given when parties exchanged opinions. There have also been instances in which communication was difficult because of differences in terminology and culture.

3.1.3. Competition with existing industries

The issue of competition with existing industries often arises when attempts are made to change or reform a business model change or reform. This issue can also be viewed as the reorganization of industries, rather than competition between existing industries. However, this realignment of industries is difficult to move forward in the absence of supervisors.

3.2. Benefits

3.2.1. Rising costs and monetization

Introducing IoT within products will cause costs to go up, and yet the benefits it provides are unclear. There is no specific killer application available, and the number of users will not rise immediately. Therefore, finding a way to make the business profitable will be very difficult. This issue is especially difficult for businesses and products that rely on cost reductions to deliver low prices that make them competitive.
3.3. Information security and privacy of social systems

3.3.1. Classification of ownership, location, and the usage of data

There are many questions regarding the wide variety of data gathered from IoT equipment, including questions related to ownership, storage location, and the authorization to grant a license to use data. These need to be addressed so that the system and equipment can be accepted by society.

For example, if a company installs a door in a house that gathers data on the opening and closing of the door, questions about the data will arise. Does it belong to the users or the company? Can another company use this data?

3.4. Disclosure of data

3.4.1. Side effects and malicious use potentially caused by the disclosure of data

For example, it has been shown that the electricity smart meter can lead to burglary because it shows when electricity is used and not used, providing an indication of the time when no one is home. This particular example demonstrates the importance of ensuring information security and privacy.

3.5. Preparing social support

3.5.1. Regulations

Systems of laws and regulations are important for ensuring the safety of the conventional products, but they can also be a barrier for innovation.

IoT can be affected by laws and regulations at home and abroad, and can also be influenced by regulations that extend across multiple countries. Regulatory authorities need to monitor IoT carefully and adjust the regulations and laws they oversee in a way that does not negatively impact the global competition environment.

3.5.2. Corporate social responsibility

In addition to pursuing profit, companies that promote IoT also need to improve the benefits offered to users and society.
3.5.3. Customization for individual customers

There is an ongoing shift in demand away from general products to customized products for individual customers. This could also be viewed as a shift away from manufacturing businesses to service businesses. IoT will play an important role in this shift.

Instead of manufacturing Things through mass production, it will be easier to customize a product by moving some of the functions to an application. Likewise, the manufacturing business also needs to move forward with the previously mentioned paradigm shift in order to achieve customization.

3.5.4. IoT literacy of the users

Because Things are connected to the network, apps will need to be created. Some of these will serve as the interface with which people interact with IoT.

In the IoT era of the future, users will need to possess a certain amount of knowledge about IoT apps.

3.5.5. Individual vs. family

The issue of whether the data of Things in the house belongs to the family or the individual will largely affect data analysis and the handling of privacy.

As was mentioned in Section 2.1.2, the spouse could later object to the head of the household granting authorization to use data.

4. Information Security Considerations

Meetings with the players in various IoT fields provided insight into information security issues. These issues are described in the following sections.

- Section 2.1.2 Physical damper of devices
- Section 2.1.2 Product lifetime and encryption strength

For details, please see the corresponding text.

5. Privacy Considerations

Similarly, issues regarding privacy are described in the following sections.
Section 2.1.2, Section 3.3.1 Ownership of the data
Section 3.4.1 Data disclosure and malicious use
Section 3.5.5 Individual vs. family

For details, please see the corresponding text.

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The Impact of Transport Header Encryption on Operation and Evolution of the Internet
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Abstract

This document describes implications of applying end-to-end encryption at the transport layer. It identifies some in-network uses of transport layer header information that can be used with a transport header integrity check. It reviews the implication of developing encrypted end-to-end transport protocols and examines the implication of developing and deploying encrypted end-to-end transport protocols. Since transport measurement and analysis of the impact of network characteristics have been important to the design of current transport protocols, it also considers some anticipated implications on transport and application evolution.

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1. Introduction

This document discusses the implications of end-to-end encryption applied at the transport layer, and examines the impact on transport protocol design, transport use, and network operations and management. It also considers some anticipated implications on transport and application evolution.

The transport layer is the first end-to-end layer in the network stack. Despite headers having end-to-end meaning, some transport headers have come to be used in various ways within the Internet. In response to pervasive monitoring [RFC7624] revelations and the IETF consensus that "Pervasive Monitoring is an Attack" [RFC7258], efforts are underway to increase encryption of Internet traffic, which would prevent visibility of transport headers. This has implications on how network protocols are designed and used [I-D.mm-wg-effect-encrypt].

Transport information that is sent without end-to-end integrity check could be modified by "middleboxes" - defined as any intermediary box performing functions apart from normal, standard functions of an IP router on the data path between a source host and destination host [RFC3234]. When transport headers are modified by network devices on the path, this can change the end-to-end protocol transport protocol behaviour in a way that may have benefits (e.g., to user performance/cost) or may hinder (e.g., disrupting application experience). Whatever the outcome, modification of packets by a middlebox was not
usually intended when the protocol was specified and is usually not known by the sending or receiving endpoints.

Middleboxes have been deployed for a variety of reasons [RFC3234], including protocol enhancement, proxies such as Protocol Enhancing Proxies (PEPs) [RFC3135], TCP acknowledgement (ACK) enhancement [RFC3449], use by application protocol caches [I-D.mm-wg-effect-encrypt], application layer gateways [I-D.mm-wg-effect-encrypt], etc. [I-D.dolson-plus-middlebox-benefits] summarizes some of the functions provided by such middleboxes, and benefits that may arise when used in specific deployment scenarios. Such methods, which involve in-network modification of transport headers, are not further discussed.

Transport protocols can be designed to encrypt or authenticate transport header fields. Authentication methods at the transport layer can detect any changes to an immutable header field that were made by a network device along a path. These methods do not require encryption of the header fields, and hence authenticated fields may remain visible to network devices. A receiving transport endpoint can use an integrity check to avoid accepting modified protocol headers. This document therefore does not consider the case where there is undetected modification of the transport header fields as a packet traverses the network path. The intentional modification of transport headers by middleboxes (such as Network Address Translation with Protocol Translation, NAT-P) is not considered.

Authentication methods (that provide integrity checks of protocols fields) have also been specified at the network layer, and this also protects transport header fields. The network layer itself carries protocol header fields that are increasingly used to help forwarding decisions reflect the need of transport protocols, such as the IPv6 Flow Label [RFC6437], the Differentiated Services Code Point (DSCP) [RFC2474] and Explicit Congestion Notification (ECN) [RFC3168].

Encryption methods can hide information from an eavesdropper in the network. Encryption can also help protect the privacy of a user, by hiding data relating to user/device identity or location. Neither an integrity check nor encryption methods prevent traffic analysis, and usage needs to reflect that profiling of users and fingerprinting of behaviour can take place even on encrypted traffic flows.

This document seeks to identify the implications of various approaches to transport protocol authentication and encryption.
2. Internet Transports and Pervasive Encryption

End-to-end encryption can be applied at various protocol layers. It can be applied above the transport to encrypt the transport payload. One motive to use encryption is a response to perceptions that the network has become ossified by over-reliance on middleboxes that prevent new protocols and mechanisms from being deployed. This has lead to a common perception that there is too much "manipulation" of protocol headers within the network, and that designing to deploy in such networks is preventing transport evolution. In the light of this, a method that authenticates transport headers may help improve the pace of transport development, by eliminating the need to always consider deployed middleboxes [I-D.trammell-plus-abstract-mech], or potentially to only explicitly enable middlebox use for particular paths with particular middleboxes that are deliberately deployed to realise a useful function for the network and/or users[RFC3135].

Another perspective stems from increased concerns about privacy and surveillance. Some Internet users have valued the ability to protect identity and defend against traffic analysis, and have used methods such as IPsec ESP and Tor [Tor]. Revelations about the use of pervasive surveillance [RFC7624] have, to some extent, eroded trust in the service offered by network operators, and following the Snowden revelation in the USA in 2013 has led to an increased desire for people to employ encryption to avoid unwanted "eavesdropping" on their communications. Whatever the reasons, there are now activities in the IETF to design new protocols that may include some form of transport header encryption (e.g., QUIC [I-D.ietf-quic-transport]).

The use of transport layer authentication and encryption exposes a tussle between middlebox vendors, operators, applications developers and users.

- On the one hand, future Internet protocols that enable large-scale encryption assist in the restoration of the end-to-end nature of the Internet by returning complex processing to the endpoints, since middleboxes cannot modify what they cannot see.
- On the other hand, encryption of transport layer header information has implications for people who are responsible for operating networks and researchers and analysts seeking to understand the dynamics of protocols and traffic patterns.

Whatever the motives, a decision to use pervasive of transport header encryption will have implications on the way in which design and evaluation is performed, and which can in turn impact the direction of evolution of the TCP/IP stack.
The next subsections briefly review some security design options for transport protocols.

2.1. Authenticating the Transport Protocol Header

Transport layer header information can be authenticated. An integrity check that protects the immutable transport header fields, but can still expose the transport protocol header information in the clear, allowing in-network devices to observe these fields. An integrity check cannot prevent in-network modification, but can avoid accepting changes and avoid impact on the transport protocol operation.

An example transport authentication mechanism is TCP-Authentication (TCP-AO) [RFC5925]. This TCP option authenticates TCP segments, including the IP pseudo header, TCP header, and TCP data. TCP-AO protects the transport layer, preventing attacks from disabling the TCP connection itself. TCP-AO may interact with middleboxes, depending on their behavior [RFC3234].

The IPSec Authentication Header (AH) [RFC4302] works at the network layer and authenticates the IP payload. This therefore also authenticates all transport headers, and verifies their integrity at the receiver, preventing in-network modification.

2.2. Encrypting the Transport Payload

The transport layer payload can be encrypted to protect the content of transport segments. This leaves transport protocol header information in the clear. The integrity of immutable transport header fields could be protected by combining this with an integrity check (Section 2.1).

Examples of encrypting the payload include Transport Layer Security (TLS) over TCP [RFC5246] [RFC7525] or Datagram TLS (DTLS) over UDP [RFC6347] [RFC7525].

2.3. Encrypting the Transport Header

The network layer payload could be encrypted (including the entire transport header and payload). This method does not expose any transport information to devices in the network, which also prevents modification along the network path.

The IPSec Encapsulating Security Payload (ESP) [RFC4303] is an example of encryption at the network layer, it encrypts and authenticates all transport headers, preventing visibility of the
headers by in-network devices. Some Virtual Private Network (VPN) methods also encrypt these headers.

2.4. Authenticating Transport Information and Selectively Encrypting the Transport Header

A transport protocol design can encrypt selected header fields, while also choosing to authenticate fields in the transport header. This allows specific transport header fields to be made observable by network devices. End-to-end integrity checks can prevent an endpoint from undetected modification of the immutable transport headers.

The choice of which fields to expose and which to encrypt is a design choice for the transport protocol. Any selective encryption method requires trading two conflicting goals for a transport protocol designer to decide which header fields to encrypt. On the one hand, security work typically employs a design technique that seeks to expose only what is needed. On the other hand, there may be performance and operational benefits in exposing selected information to network tools.

Mutable fields in the transport header provide opportunities for middleboxes to modify the transport behaviour (e.g., the extended headers described in [I-D.trammell-plus-abstract-mech]). This considers only immutable fields in the transport headers, that is, fields that may be authenticated end-to-end across a path.

An example of a method that encrypts some, but not all, transport information is GRE-in-UDP [RFC8086] when used with GRE encryption.

2.5. Adding Transport Information to Network-Layer Protocol Headers

The transport information can be made visible in a network-layer header. This has the advantage that this information can then be observed by in-network devices. This has the advantage that a single header can support all transport protocols, but there may also be less desirable implications of separating the operation of the transport protocol from the measurement framework.

Some measurements may be made by adding additional protocol headers carrying operations, administration and management (OAM) information to packets at the ingress to a maintenance domain (e.g., an Ethernet protocol header with timestamps and sequence number information using a method such as 802.11ag) and removing the additional header at the egress of the maintenance domain. This approach enables some types of measurements, but does not cover the entire range of measurements described in this document.
Another example of a network-layer approach is the IPv6 Performance and Diagnostic Metrics (PDM) Destination Option [I-D.ietf-ippm-6man-pdm-option]. This allows a sender to optionally include a destination option that carries header fields that can be used to observe timestamps and packet sequence numbers. This information could be authenticated by receiving transport endpoints when the information is added at the sender and visible at the receiving endpoint, although methods to do this have not currently been proposed. This method needs to be explicitly enabled at the sender.

A drawback of using extension headers is that IPv4 network options are often not supported (or are carried on a slower processing path) and some IPv6 networks are also known to drop packets that set an IPv6 header extension. Another disadvantage is that protocols that separately expose header information do not necessarily have an advantage to expose the information that is utilised by the protocol itself, and could manipulate this header information to gain an advantage from the network.

3. Use of Transport Headers in the Network

This section identifies ways that actors can benefit by observing (non-encrypted) transport header fields at devices in the network. The list of actors who perform measurements include:

- Protocol developers and implementors of TCP/IP stacks;
- Researchers working on new mechanisms;
- Use of new applications using existing applications;
- Analysis researching the impact of mechanisms on network equipment or specific network topologies;
- Staff supporting operation of a network.

One approach is to use active measurement using dedicated tools to generate and measure test traffic. To test a transport path, such active tools need to be run from an endpoint, and most operators do not have access to user equipment. There may also be costs associated with running such tests (e.g., the implications of bandwidth tests in a mobile network are obvious). Some active measurements (e.g., response under load or particular workloads) may perturb other traffic, and could require dedicated access to the network segment. An alternative approach is to use in-network techniques that observe transport packet headers in operational networks to make the measurements.

Transport layer information can help identify whether the link/network tuning is effective and alert to potential problems that can be hard to derive from link or device measurements alone. The design
Trade offs for radio networks are often very different to those of wired networks. A radio-based network (e.g., cellular mobile, enterprise WiFi, satellite access/backhaul, point-to-point radio) has the complexity of a subsystem that performs radio resource management — with direct impact on the available capacity, and potentially loss/reordering of packets. The impact of the pattern of loss and congestion, differs for different traffic types, correlation with propagation and interference can all have significant impact on the cost and performance of a provided service. The need for this type of information is expected to increase as operators bring together heterogeneous types of network equipment and seek to deploy opportunistic methods to access radio spectrum.

In-network observation of transport protocol headers requires knowledge of the format of the transport header:

- Flows, need to be identified at the level required for monitoring;
- The protocol and version of the header that is being used. As protocols evolve over time and there may be a need to introduce new transport headers. This may require interpretation of protocol version information or connection setup information;
- The position and syntax of any transport headers that need to be observed. IETF transport protocols specify this information.

The following subsections describe various ways that observable transport information may be utilised.

3.1. Use to Identify Flows and Packet Formats

Transport protocol header information can identify a flow and the connection state of the flow, together with the protocol options being used. In some usages, a low-numbered (well-known) port that can identify a protocol (although port information alone is not sufficient to guarantee identification of a protocol). Transport protocols, such as TCP and SCTP specify a standard base header that includes sequence number information and other data, with the possibility to negotiate additional headers at connection setup and identified by an option number in the transport header. UDP-based protocols sometimes do not use well-known ports but also can instead be identified by signalling protocols or through the use of magic numbers placed in the first byte(s) of the datagram payload.

3.2. Measurements derived from Transport Header Information

Some actors have a need to characterise the performance of link/network segments. Passive monitoring uses observed traffic to make inferences from transport headers to derive measurements. A variety of open source and commercial tools can utilise this information.
Transport fields in the Real Time Protocol (RTP) header [RFC3550] [RFC4585] can be observed to derive traffic volume measurements and provide information on the progress and quality of a session using RTP. Key performance indicators are retransmission rate, packet drop rate, sector utilization level, a measure of reordering, peak rate, the CE-marking rate, etc. Metadata is often important to understand the context under which the data was collected, including the time, observation point, and way in which metrics were accumulated.

Some Internet transports report summary performance data that is observable in the network (e.g., RTCP feedback [RFC3550]). A user of summary measurement data needs to trust the source of this data and the method used to generate the summary information.

When encryption conceals information in packet headers, measurements need to rely on pattern inferences and other heuristics grows, and accuracy suffers [I-D.mm-wg-effect-encrypt].

3.2.1. Use to Characterise Traffic Rate and Volume

Transport headers may be observed to derive volume measures per-application, to characterise the traffic using a network segment and pattern of network usage. This may be measured per endpoint or aggregate of endpoint (e.g., by an operator to assess subscriber usage). This can also be used to trigger measurement-based traffic shaping and to implement QoS support within the network and lower layers. Volume measures can be valuable for capacity planning (providing detail of trends rather than the volume per subscriber).

3.2.2. Measuring Loss Rate and Loss Pattern

Flow loss rate is often used as a metric for performance assessment and to characterise the transport behaviour. Understanding the root cause of loss can help an operator determine whether this requires corrective action.

There are various cause of loss, including: corruption on a link (e.g., interference on a radio link), buffer overflow (e.g., due to congestion), policing (traffic management), buffer management (e.g., Active Queue Management (AQM)). Loss can be monitored at the interface level by devices in the network. It is often important to understand the conditions under which packet loss occurs, which usually means relating loss to the traffic flowing on the network segment at the time of loss. Understanding flow loss rate requires either maintaining per flow packet counters or by observing sequence numbers in transport headers.
Observation of transport feedback information (observing loss reports, e.g., RTCP, TCP SACK) can increase understanding of the impact of loss and help identify cases where loss may have been wrongly identified, or the transport did not require the lost packet. It is sometimes more important to understand the pattern of loss, than the loss rate – since losses can often occur as bursts, rather than randomly timed events.

3.2.3. Measuring Throughput and Goodput

The throughput observed by a flow can be determined even when a flow is encrypted, providing the individual flow can be identified. Goodput [RFC7928] is a measure of useful data exchanged (the ratio of useful/total volume of traffic sent by a flow), which requires ability to differentiate loss and retransmission of packets (e.g., by observing packet sequence numbers in TCP or RTP).

3.2.4. Measuring Latency (Network Transit Delay and Jitter)

Latency is a key performance metric that impacts application response time and user-perceived response time. It also often indirectly impacts throughput and flow completion time. Latency determines the reaction time of the transport protocol itself, impacting flow setup, congestion control, loss recovery, and other transport mechanisms. The observed latency can have many components [Latency]. Of these, unnecessary/unwanted queuing in network buffers has often been observed as a significant factor. Once the cause of unwanted latency has been identified, this can often be eliminated, and determining latency metrics is a key driver in the deployment of AQM [RFC7567], DiffServ [RFC2474], and ECN [RFC3168] [RFC8087].

To measure latency across a part of the path, an observation point can measure the experienced round trip time (RTT) using packet sequence numbers, and acknowledgements, or by observing header timestamp information. Such information allows an observation point in the network to determine not only the path RTT, but also to measure the upstream and downstream contribution to the RTT. This may be used to locate a source of latency, e.g., by observing cases where the ratio of median to minimum RTT is large for a part of a path.

An example usage of this method could be to identify excessive buffers and to deploy or configure Active Queue Management (AQM) [RFC7567] [RFC7928]. Operators deploying such tools can effectively eliminate unnecessary queuing in routers and other devices. AQM methods need to be deployed at the capacity bottleneck, but are often deployed in combination with other techniques, such as scheduling [RFC7567] [I-D.ietf-aqm-fq-codel] and although parameter-less methods
are desired [RFC7567], current methods [I-D.ietf-aqm-fq-codel] [I-D.ietf-aqm-codel] [I-D.ietf-aqm-pie] often cannot scale across all possible deployment scenarios. The service offered by operators can therefore benefit from latency information to understand the impact of deployment and tune deployed services.

Some network applications are sensitive to packet jitter, and it can be necessary to measure the jitter observed along a portion of the path. The requirements to measure jitter resemble those for the measurement of latency.

3.2.5. Measuring Flow Reordering

Significant flow reordering can impact time-critical applications and can be interpreted as loss by reliable transports. Many transport protocol techniques are impacted by reordering (e.g., triggering TCP retransmission, or rebuffering of real-time applications). Packet reordering can occur for many reasons (from equipment design to misconfiguration of forwarding rules).

As in the drive to reduce network latency, there is a need for operational tools to be able to detect misordered packet flows and quantify the degree or reordering. Techniques for measuring reordering typically observe packet sequence numbers. Metrics have been defined that evaluate whether a network has maintained packet order on a packet-by-packet basis [RFC4737] and [RFC5236].

There has been initiatives in the IETF transport area to reduce the impact of reordering within a transport flow, possibly leading to reduced the requirements for ordering. These have promise to simplify network equipment design as well as the potential to improve robustness of the transport service. Measurements of reordering can help understand the level of reordering within deployed infrastructure, and inform decisions about how to progress such mechanisms.

3.3. Measurements derived from Network-Transport Information

This section describes transport information that is already observable in network-layer header fields.

3.3.1. Use of IPv6 Network-Layer Flow Label

Endpoints should expose flow information in the IPv6 Flow Label field of the network-layer header (e.g., [RFC8085]). This can be used to inform network-layer queuing, forwarding (e.g., for equal cost multi-path (ECMP) routing, and Link Aggregation (LAG)). This can provide useful information to assign packets to flows in the data collected.
by measurement campaigns. Although important to characterising a path, it does not directly provide any performance data.

3.3.2. Use Network-Layer Differentiated Services Code Point

Application can expose their delivery expectations to the network, by setting the Differentiated Services Code Point (DSCP) field of IPv4 and IPv6 packets. This can be used to inform network-layer queuing and forwarding, and can also provide information on the relative importance of packet information collected by measurement campaigns, but does not directly provide any performance data.

This field provides explicit information that can be used in place of inferring traffic requirements (e.g., by inferring QoS requirements from port information via a multi-field classifier). The DSCP value can therefore impact the quality of experience for a flow. Observations of service performance need to consider this field when a network path has support for differentiated service treatment.

3.3.3. Use of Explicit Congestion Marking

Explicit Congestion Notification (ECN)[RFC3168] uses a codepoint in the network-layer header. Use of ECN can offer gains in terms of increased throughput, reduced delay, and other benefits when used over a path that includes equipment that supports an AQM method that performs Congestion Experienced (CE) marking of IP packets [RFC8087].

This exposes the presence of congestion on a network path to the transport and network layer. The reception of Congestion Experienced (CE) marked packets can therefore be used to monitor the presence and estimate the level of incipient congestion on the upstream portion of the path from the point of observation (Section 2.5 of [RFC8087]). Because ECN marks carried in the IP protocol header, measuring ECN can be much easier than metering packet loss. However, interpreting the marking behaviour (i.e., assessing congestion and diagnosing faults) requires context from the transport layer (path RTT, visibility of loss - that could be due to queue overflow, congestion response, etc)[RFC7567].

Some ECN-capable network devices can provide richer (more frequent and fine-grained) indication of their congestion state. Setting congestion marks proportional to the level of congestion (e.g., DCTP [I-D.ietf-tcpm-dctcp], and L4S [I-D.ietf-tpsv-wg-l4s-arch]).

AQM and ECN offer a range of algorithms and configuration options, it is therefore important for tools to be available to network operators and researchers to understand the implication of configuration choices and transport behaviour as use of ECN increases and new
methods emerge [RFC7567] [RFC8087]. ECN-monitoring is expected to become important as AQM is deployed that supports ECN [RFC8087].

Section 5.6 describes the transport layer feedback information that accompanies the use of ECN.

4. Transport Measurement

The common language between network operators and application/content providers/users is packet transfer performance at a layer that all can view and analyze. For most packets, this has been transport layer, until the emergence of QUIC, with the obvious exception of VPNs and IPsec. When encryption conceals more layers in a packet, people seeking understanding of the network operation need to rely more on pattern inferences and other heuristics. The accuracy of measurements therefore suffers, as does the ability to investigate and troubleshoot interactions between different anomalies. For example, the traffic patterns between server and browser are dependent on browser supplier and version, even when the sessions use the same server application (e.g., web e-mail access). Even when measurement datasets are made available (e.g., from endpoints) additional metadata, such as the state of the network, is often required to interpret the data. Collecting and coordinating such metadata is more difficult when the observation point is at a different location to the bottleneck/device under evaluation.

Packet sampling techniques can be used to scale processing involved in observing packets on high rate links. This only exports the packet header information of (randomly) selected packets. The utility of these measurements depends on the type of bearer and number of mechanisms used by network devices. Simple routers are relatively easy to manage, a device with more complexity demands understanding of the choice of many system parameters. This level of complexity exists when several network methods are combined.

This section discusses topics concerning transport measurement.

4.1. Point of Measurement

Often measurements can only be understood in the context of the other flows that share a bottleneck. A simple example is the monitoring of AQM. For example, FQ-CODEL [I-D.ietf-aqm-fq-codel], combines sub queues (statistically assigned per flow), management of the queue length (CODEL), flow-scheduling, and a starvation prevention mechanism. Usually such algorithms are designed to be self-tuning, but current methods typically employ heuristics that can result in more loss under certain path conditions (e.g., large RTT, effects of multiple bottlenecks [RFC7567]).
In-network measurements that can distinguish between upstream and downstream metrics with respect to the measurement point. They are particularly useful for locating the source of problems or to assess the performance of a network segment or a particular device configuration.

4.2. Use by Operators to Plan and Provision Networks

Traffic measurements (e.g., traffic volume, loss, latency) is used by operators to help plan deployment of new equipment and configurations in their networks. Data is also important to equipment vendors who need to understand traffic trends traffic and patterns of usage as inputs to decisions about planning products and provisioning for new deployments. This measurement information can also be correlated with billing information when this is also collected by an operator.

A network operator supporting traffic that uses transport header encryption may not have access to per-flow measurement data. Trends in aggregate traffic can be observed and can be related this to the endpoint addresses being used, but it may not be possible to correlate patterns in measurements with changes in transport protocols (e.g., the impact of changes in introducing a new transport protocol mechanism). This increases the dependency on other indirect sources of information to inform planning and provisioning.

4.3. Service Performance Measurement

Traffic measurements (e.g., traffic volume, loss, latency) can be used by various actors to help understand the performance available to users of a network segment. While active measurements may be used in-network passive measurements can have advantages in terms of eliminating unproductive traffic, reducing the influence of test traffic on the overall traffic mix, and the ability to choose the point of measurement Section 4.1.

4.4. Use for Network Diagnostics and Troubleshooting

Transport header information is useful for a variety of operational tasks [I-D.mm-wg-effect-encrypt]: to diagnose network problems, assess performance, capacity planning, management of denial of service threats, and responding to user performance questions. These tasks seldom involve the need to determine the contents of the transport payload, or other application details.

A network operator supporting traffic that uses transport header encryption can see only encrypted transport headers. This prevents deployment of performance measurement tools that rely on transport protocol information. Choosing to encrypt all information may be
expected to reduce the ability for networks to "help" (e.g., in response to tracing issues, making appropriate Quality of Service, QoS, decisions). For some this will be blessing, for others it may be a curse. For example, operational performance data about encrypted flows needs to be determined by traffic pattern analysis, rather than relying on traditional tools. This can impact the ability of the operator to respond to faults, it could require reliance on endpoint diagnostic tools or user involvement in diagnosing and troubleshooting unusual use cases or non-trivial problems. Although many network operators utilise transport information as a part of their operational practice, the network will not break because transport headers are encrypted.

4.5. Acceptable Response to Congestion

Congestion control is a key transport function. Many network operators implicitly accept that TCP traffic to comply with a behaviour that is acceptable for use in the shared Internet. TCP algorithms have been continuously improved over decades, and they have reached a level of efficiency and correctness that custom application-layer mechanisms will struggle to easily duplicate [RFC8085]. A standards-compliant TCP stack provides congestion control that is therefore judged safe for use across the Internet. Applications developed on top of well-designed transports can be expected to appropriately control their network usage, reacting when the network experiences congestion, by back-off and reduce the load placed on the network. This is the normal expected behaviour for TCP and other IETF-defined transports.

Tools exist that can interpret the transport protocol header information to help understand the impact of specific transport protocols (or protocol mechanisms) on other traffic that shares their network. An observation in the network can gain understanding of the dynamics of a flow and its congestion control behaviour. Analysing observed packet sequence numbers can be used to help build confidence that an application flow backs-off its share of the network load in the face of persistent congestion, and hence to understand whether the behaviour is appropriate for sharing limited network capacity. For example, it is common to visualise plots of TCP sequence numbers versus time for a flow to understand how a flow shares available capacity, deduce its dynamics in response to congestion, etc.

4.5.1. Measuring Compliance of UDP Traffic

UDP provides a minimal message-passing transport that has no inherent congestion control mechanisms. Because congestion control is critical to the stable operation of the Internet, applications and other protocols that choose to use UDP as an Internet transport must
employ mechanisms to prevent congestion collapse, avoid unacceptable contributions to jitter/latency, and to establish an acceptable share of capacity with concurrent traffic [RFC8085].

A network operator has no way of knowing the specific methods used by a UDP application, unless the header format can be determined. Tools are needed to understand if UDP flows comply with congestion control expectations and therefore whether there is a need to deploy methods such as rate-limiters, transport circuit breakers or other methods to enforce acceptable usage. UDP flows that expose a well-known header by specifying the format of header fields can allow information to be observed that gains understanding of the dynamics of a flow and its congestion control behaviour. For example, tools exist to monitor various aspects of the RTP and RTCP header information of real-time flows (see Section 3.2).

4.5.2. Measuring Transport to Support Network Operations

By correlating observations at multiple points along the path (e.g., at the ingress and egress of a network segment), an observer can determine the contribution of a portion of the path to an observed metric (to locate a source of delay, jitter, loss, reordering, congestion marking, etc).

Information provided by tools can help determine whether mechanisms are needed in the network to prevent flows from acquiring excessive network capacity. Operators can manage traffic flows (e.g., to prevent flows from acquiring excessive network capacity under severe congestion) by deploying rate-limiters, traffic shaping or network transport circuit breakers [RFC8084].

5. Observing Transport Flows with Encrypted Transport Header Fields

This section examines implications of encrypting specific transport header information.

5.1. Transport Information at the Network Layer

Some transport information is made visible in the network-layer protocol header. These header fields are not encrypted and can be used to make flow observations. Endpoints should expose flow information in the IPv6 Flow Label Section 3.3.1 in the network-layer header. This can be used to inform network-layer queuing, forwarding (e.g., for equal cost multi-path (ECMP) routing, and Link Aggregation (LAG)). For transport measurement, this can provide useful information to assign packets to flows in the data collected by measurement campaigns, but does not directly provide any performance data. Similarly the Differentiated Services CodePoint (DCSP)
indicates expected forwarding treatment Section 3.3.2. The ECN field provides observable congestion data and can help inform measurement of flow congestion Section 3.3.3.

5.2. An Observable Transport Flow Identifier

To measure and analyse a transport protocol, a measurement tool needs to be able to identify traffic flows. Aggregation of sessions, and persistent use of established transport flows by multiple sessions means that a flow at the transport layer is not necessarily the same as a flow seen at the application layer. This is usually not a consequence. Data is measured for the aggregate transport flow.

Some measurement methods sample traffic, rather than collecting all packets passing through a measurement point. These methods still require a way to determine the presence, size and position of any observable header fields - but may need to do this without observing a protocol exchange for a connection setup.

5.2.1. A Method to Determine Header Format

If flow information is observed from transport headers, then there needs to be a way to identify the format of the header Section 3.1. Some IETF transport protocols are identified by an IP protocol number (e.g., TCP, SCTP, UDP). All IETF-defined transport protocols include a transport port field in their transport header. Higher layer protocols (e.g., HTTP) can be sometimes be observed by a well-known port value, which can be indicative of the protocol being encapsulated, but there is no way to enforce this usage. This can be used to configure decapsulation, alternatives include a "magic" number placed at the start of each UDP datagram.

Once the protocol has been determined, the transport header can be determined from a published specification. If multiple formats are permitted, this may also require observing the protocol version being used and possibly parameter negotiation at connection setup.

5.2.2. Use of a Transport as a Substrate

When a transport is used as a substrate, the transport provides an encapsulation that allows another transport flow to be within the payload of a transport flow. The transported protocol header may provide additional information for multiplexing multiple flows over the same 5-tuple. The UDP Guidelines [RFC8085] provides some guidance on using UDP as a substrate protocol. If there is no additional information about the protocol transported by the substrate, this may be viewed as an opaque traffic aggregate, and prevents transport measurement in the network. Examples include GRE-
in-UDP [RFC8086], SCTP-in-UDP. The GRE-in-UDP encapsulation may
eencrypt the payload, but does not encrypt the GRE protocol header.

5.2.3. Support for Mobility and Flow Migration

With the proliferation of mobile connected devices, there is a stated
need for connection-oriented protocols to maintain connections after
a network migration by an endpoint. The ability and desirability of
in-network devices to track such migration depends on the context.
On the one hand, a load-balancer device in front of server may find
it useful to map a migrated connection to the same server endpoint.
On the other hand, a user performing migration to avoid detection may
prefer the network not to be able to correlate the different parts of
a migrating session. Care must then be exercised to make sure that
the information encoded by the endpoints is not sufficient to
identify unique flows and facilitate a persistent surveillance attack
vector [I-D.mm-wg-effect-encrypt].

The impact of flow migration on measurement activities depends on the
data being measured, rate of migration and level of encryption that
is employed. Requirements for load balancing and mobility can lead
to complex protocol interactions.

5.2.4. Flow Start and Stop

Transports can expose that start and end of flows in a transport
header field (e.g., TCP SYN, FIN, RST). This can also help
measurement devices identify the start of flows, or to remove stale
flow information. This information is supplemental – flows can start
and end at any time, the Internet network layer provides only a best
effort service that allows alternate routing, reordering, loss, etc,
so a network measurement tool can not rely upon observing these
indicators. The time to complete a protocol connection and/or
session setup can be reported.

Flow information can provide in-network devices to manage their
forwarding state [I-D.trammell-plus-statefulness]. It can assist a
firewall in deciding which flows are permitted through a security
gateway, or to help maintain the network address translation (NAT)
bindings in a NAPT or application layer gateway. This information
may also find use in load balancers, where visibility of the 5-tuple
could assist in selecting a server [I-D.mm-wg-effect-encrypt].

Access to flow information and an observable start/stop indication
[I-D.trammell-plus-statefulness] can avoid stateful middleboxes
relying on timeouts to remove old state. Without this, middleboxes
are unaware when a particular flow ceases to be used by an
application[RFC8085]. This can lead to the state table entries keeping state for less time for flows that are not identifiable.

5.3. Observable Transport Sequence Number

The TCP or RTP sequence number can be observed in one direction (the direction that carries data segments). An authenticated header prevents this field being modified or terminated/split [RFC3135] by a network device, but allows this still to be used to observe progress of the network flow.

An incrementing sequence number enables detection of loss (either by correlating ingress and egress value, or when assuming that all packets follow a single path), duplication and reordering (with understanding that not necessarily all packets of a flow follow the same path, and reordering can complicate processing of observations). Tools are widely available to interpret RTP and TCP sequence numbers, ranging from open source tools to dedicated commercial packages. As for TCP, use by in-network measurement devices needs to account for the impact of load-balancing of flows, changes in forwarding behaviour, measurement loss (rather than observed packet loss), etc.

5.4. Observable Transport Reception

Acknowledgement (ACK) data provides information about the path from the network device to the remote endpoint. The information can help identify packet loss (or the point of loss), RTT, and other network-related performance parameters (e.g., throughput, jitter, reordering). Unless this information is correlated with other data there is no way to disambiguate the cause of impairments (congestion loss, link transmission loss, equipment failure).

An in-network device must not modify the flow of end-to-end ACK data when using an authenticated protocol. That is, must not use the in-network methods described in [RFC3449]. This can impact the performance and/or efficiency (e.g., cost) of using paths where the return capacity is limited or has implications on the overall design (e.g., using TCP with cellular mobile uplinks, DOCSIS uplinks).

The TCP stream can be observed by correlating the stream of TCP ACKs that flow from a receiver in the return direction. Although these ACKs are cumulative, and are not necessarily sent on the same path as the forward data, when visible, their sequence can confirm successful transmission and the path RTT. In the case of TCP they may also indicate packet loss (duplicate ACKs).

An RTP session can provide reception information [RFC3550] [RFC4585] feedback using the RTCP framework. This reception information and
can be observed by in-network measurement devices and can be interpreted to provide a variety of quality of experience information for the related RTP flow, as well as basic network performance data (RTT, loss, jitter, etc).

5.5. Observable Transport Timestamps

The use of timestamps for latency and jitter measurements Section 3.2.4 is discussed in other sections of the current version of the document.

5.6. Observable ECN Transport Feedback Information

Transport protocols that use ECN Section 3.3.3 need to provide ECN feedback information in the transport header to inform the sender whether packets have been received with an ECN CE-mark [RFC3819]. This information can be in the form of feedback once each RTT [RFC3819] or more frequent. The latter may involve sending a detailed list of all ECN-marked packets (e.g., [I-D.ietf-tcpm-accurate-ecn] and [RFC6679]). The detailed information can provide detail about the pattern and rate of marking. The information provided in these protocol headers can help a network operator to understand the congestion status of the forward path and the impact of marking algorithms on the traffic that is carried [RFC8087].

IETF specifications for Congestion Exposure (CONVEX) [RFC7713] is an example of a framework that monitors reception reports for CE-marked packets to support network operations.

5.7. Other Observable Transport Fields

This section is not complete - later revision may determine other fields or remove this section.

5.8. Interpretation of Transport Header Fields

Understanding and analysing transport protocol behaviour typically demands tracking changes to the protocol state at the transport endpoints. Although protocols communicate state information in their protocol headers, a protocol implementation typically also contains internal state that is not directly visible from observing transport protocol headers. Effective measurement tools need to consider that not all packets may be observed (due to drops at the capture tap or because packets take an alternate route that does not pass the tap). Some flows of packets may also be encapsulated within a maintenance domain in other protocols, which further complicates analysis.
Some examples of using network measurements of transport headers to infer internal TCP transport state information include:

- The TCP congestion window (cwnd) and slow start threshold (ssthresh). Tools for analysing in-network performance of TCP may observe sequence number to infer the current congestion controller state.
- The TCP RTT estimator and TCP Retransmission Time Out (RTO) value. This can be estimated by correlating sequence and acknowledgement numbers, or possibly by observing TCP timestamp options.
- Use of pacing (and pacing rate) and use of methods such as Proportional Rate Reduction (PRR) and Congestion Window validation (CWV). This may be estimated from observing timing of segments with TCP sequence numbers. This is important to some congestion control mechanisms and can be important for applications that are rate limited or send traffic bursts.
- Receiver window and flow control state. This may be inferred from information in TCP ACK segments. It is important to applications where the remote endpoint is resource constrained, or the path exhibits a large RTT.
- Retransmission state and receiver buffer. This may be inferred from information in TCP ACK segments (especially when SACK blocks are provided), this can be important to the performance of applications that send traffic bursts.
- Use of ACK delay and Nagle algorithm. This may be estimated from observing timing of segments with TCP sequence numbers, and is important to the performance of thin application flows.

5.9. Requirements for Transport Measurement

Transport measurement and analysis of the impact of network characteristics have been important to the design of current transport protocols. Transport measurement introduces the following requirements to identify the observable information:

- Observable protocol type and version information is needed to identify the protocol being used when characterising the traffic, and to enable further observation of the flow.
- Observable format information is needed to allow an observer to determine the presence of any observable header fields.
- A published specification is needed to allow an observer to determine the size and position of any observable header fields so that these fields may be decoded by a measurement tool.
- Observable flow start/stop information can assist some forms of measurement and has utility for middleboxes that track state.
The need for in-network transport measurement introduces the following requirements for observable information in transport header fields:

- Observable transport information to determine the progress of flows for each direction of communication. This requires observable packet numbers.
- Observable transport information to determine loss, and understand the response to congestion for a network segment. This requires observable reception information (e.g., packet acknowledgment information).
- Observable transport information is needed for more advanced measurement of latency, jitter, etc. This requires an observable field and a method to correlating return information with the observed field. This could utilise a packet number and/or transmission timestamp information. This information needs to be available in both directions of transmission.
- Exposure of Transport ECN feedback provides a powerful tool to understand ECN-enabled AQM-based networks. (Forward ECN information is already observable in the network header).

6. The Effect of Encrypting Transport Header Fields

This section explores key implications of working with encrypted transport protocols.

6.1. Independent Measurement

Independent observation by multiple actors is important for scientific analysis. Encrypting transport header encryption changes the ability for other actors to collect and independently analyse data. Internet transport protocols employ a set of mechanisms. Some of these need to work in cooperation with the network layer - loss detection and recovery, congestion detection and congestion control, some of these need to work only end-to-end (e.g., parameter negotiation, flow-control).

When encryption conceals information in the transport header, it could be possible for an applications to provide summary data on performance and usage of the network. This data could be made available to other actors. However, this data needs to contain sufficient detail to understand (and possibly reconstruct the network traffic pattern for further testing) and to be correlated with the configuration of the network paths being measured. Sharing information between actors needs also to consider the privacy of the user and the incentives for providing accurate and detailed information. Protocols that expose the state information used by the transport protocol in their header information (e.g., timestamps used
to calculate RTT, packet numbers used to assess congestion and requests for retransmission) provide an incentive for the sending endpoint to provide correct information, increasing confidence that the observer understands the transport interaction with the network. This becomes important when considering changes to transport protocols, changes in network infrastructure, or the emergence of new traffic patterns.

6.2. Characterising "Unknown" Network Traffic

If "unknown" or "uncharacterised" traffic patterns form a small part of the traffic aggregate passing through a network device or segment of the network the path, the dynamics of the uncharacterised traffic may not have a significant collateral impact on the performance of other traffic that shares this network segment. Once the proportion of this traffic increases, the need to monitor the traffic and determine if appropriate safety measures need to be put in place.

Tracking the impact of new mechanisms and protocols requires traffic volume to be measured and new transport behaviours to be identified. This is especially true of protocols operating over a UDP substrate. The level and style of encryption needs to be considered in determining how this activity is performed. On a shorter timescale, information may also need to be collected to manage denial of service attacks against the infrastructure.

6.3. Accountability and Internet Transport Portocols

Attention therefore needs to be paid to the expected scale of deployment of new protocols and protocol mechanisms. Whatever the mechanism, experience has shown that it is often difficult to correctly implement combination of mechanisms [RFC8085]. These mechanisms therefore typically evolve as a protocol matures, or in response to changes in network conditions, changes in network traffic or changes to application usage.

The growth and diversity of applications and protocols using the Internet continues to expand - and there has been recent interest in a wide range of new transport methods, e.g., Larger Initial Window, Proportional Rate Reduction (PRR), congestion control methods based on measuring bottleneck bandwidth and round-trip propagation time, the introduction of AQM techniques and new forms of ECN response (e.g., Data Centre TCP, DCTP [I-D.ietf-tcpm-dctcp], and methods proposed for Low Latency Low Loss Scalable throughput, L4S). For each new method it is desirable to build a body of data reflecting its behaviour under a wide range of deployment scenarios, traffic load, and interactions with other deployed/candidate methods.
Measurement therefore has a critical role in the design of transport protocol mechanisms and their acceptance by the wider community (e.g., as a method to judge the safety for Internet deployment. Open standards suggest that such evaluation needs to include independent observation and evaluation of performance data.

7. Implications on Evolution of the Internet Transport

The transport layer provides the first end-to-end interactions across the Internet. Transport protocols are layered directly over the network service and are sent in the payload of network-layer packets. However, this simple architectural view hides one of the core functions of the transport – to discover and adapt to the properties of the Internet path that is currently being used. The design of Internet transport protocols is as much about trying to avoid the unwanted side effects of congestion on a flow and other capacity-sharing flows, avoiding congestion collapse, adapting to changes in the path characteristics, etc., as it is about end-to-end feature negotiation, flow control and optimising for performance of a specific application.

To achieve stable Internet operations the IETF transport community, has to date, relied heavily on measurement and insight provided from the wider community to understand the trade-offs and to inform selection of select appropriate mechanisms to ensure a safe, reliable and robust Internet since the 1990’s.

There are many motivations for deploying encrypted transports, and encryption of transport payloads. The increasing public concerns about the interference with Internet traffic have led to a rapidly expanding deployment of encryption to protect end-user privacy, in protocols like QUIC. At the same time, network operators and access providers, especially in mobile networks, have come to rely on the in-network functionality provided by middleboxes both to enhance performance and support network operations.

This document has expanded upon the expected implications on operational practices when working with encrypted transport protocols, and offers insight into the potential benefit of authentication, encryption and techniques that require in-network devices to interpret specific protocol header fields. It presents a need for architectural changes and consideration of approaches to the way network transport protocols are designed when using encryption.

The use of encryption at the transport layer comes with implications that need to be considered:
Troubleshooting and diagnostics: Encrypting all transport information eliminates the incentive for operators to troubleshoot what they cannot interpret: one flow experiencing packet loss looks like any other. When transport header encryption prevents decoding the transport header (if sequence numbers and flow ID are obscured), and hence understanding the impact on a particular flow or flows that share a common network segment. Encrypted traffic therefore implies "don’t touch", and a likely first response will be "can’t help, no trouble found", or the need to add complexity that comes with an additional operational cost [I-D.mm-wg-effect-encrypt].

Open verifyable data: The use of transport header encryption may reduce the range of actors who can capture useful measurement data. This may in future restrict the information sources available to the Internet community to understand the operation of the network and transport protocols, necessary to inform standardisation and design decisions for new protocols, equipment and operational practices. There are dangers in a model where transport information is only observable at endpoints: i.e., at user devices and within service platforms and a need for independently captured data to develop open standards and stimulate research into new methods.

Operational practice: Published transport specifications allow operators to check compliance. This can bring assurance to those operating networks, often avoiding the need to deploy complex techniques that routinely monitor and manage TCP/IP traffic flows (e.g. Avoiding the capital and operational costs of deploying flow rate-limiting and network circuit-breaker methods). This should continue when encrypted transport headers are used, but methods need to confirm that the traffic produced conforms to the expectations of the operator or developer.

Traffic analysis: The use of encryption could make it harder to determine which transport methods are being used across a network segment and the trends in usage. This could impact the ability for an operator to anticipate the need for network upgrades and roll-out. It can also impact on-going traffic engineering activities. Although the impact in many case may be small, there are cases where operators directly support services (e.g., in radio links, or to troubleshoot QoS-related issues). The more complex the underlying infrastructure the more important this impact.
Interactions between mechanisms: An appropriate vantage point, coupled with timing information for the flow (fine-grained timestamps) is a valuable tool in benchmarking equipment/configurations and understanding non-trivial interactions. Encryption restricts the ability to explore interactions between functions at different protocol layers. This is a side-effect of not allowing a choice of the vantage point from which this information is observed. This can be important (e.g., in examining collateral impact of flows sharing a bottleneck, or where the intention is to understand the interaction between a layer 2 function (e.g., radio resource management policy, a channel impairment, an AQM configuration, a Per Hop Behaviour (PHB) or scheduling method, and a transport protocol).

Common specifications: Since the introduction of congestion control, TCP has continued to be the predominate transport, with a consistent approach to avoiding congestion collapse. There is a risk that the diversity of transport mechanisms could also increase, with incentives to use a wide range of methods, this is not in itself a problem, nor is this a direct result of encryption. Encryption of all headers places the onus on validation in the hands of developers. While there is little to doubt that developers will seek to produce high quality code for their target use, it is not clear whether there is sufficient incentive to ensure good practice that benefits the wide diversity of requirements from the Internet community as a whole. The use of encryption needs to be weighed against the reduced visibility of the interactions between traffic, the network and the mechanisms. Especially, if a development cycle could focus on specific protocols/applications and then offer incentives for optimisations that could prove suboptimal for users or operators that utilise a network segments with different characteristics than targeted by the developer.

Restricting research and development: The use of encryption may impede independent research into new mechanisms, measurement of behaviour, and development initiatives. Experience shows that transport protocols are complicated to design and complex to deploy, and that individual mechanisms need to be evaluated while considering other mechanism, across a broad range of network topologies and with attention to the impact on traffic sharing the capacity. Adopting pervasive encryption of transport information could eliminate the independent self-checks that have previously been in place from research and academic contributors (e.g., the role of the IRTF ICCRG, and research publications in reviewing new
transport mechanisms and assessing the impact of their experimental deployment).

Pervasive use of transport header encryption can impact the ways that future protocols are designed and deployed. The choice of whether candidate transport designs should encrypt their protocol headers therefore needs to be taken based not just on security considerations, but also on the impact on operating networks and the constrictions this may place on evolution of Internet protocols. While encryption of all transport information can help reduce ossification of the transport layer, it could result in ossification of the network service. There can be advantages in providing a level of ossification of the header in terms of providing a set of open specified header fields that are observable from in-network devices.

8. Acknowledgements

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9. IANA Considerations

XX RFC ED - PLEASE REMOVE THIS SECTION XXX

This memo includes no request to IANA.

10. Security Considerations

This document is about design and deployment considerations for transport protocols. Authentication, confidentiality protection, and integrity protection are identified as Transport Features by RFC8095*. As currently deployed in the Internet, these features are generally provided by a protocol or layer on top of the transport protocol; no current full-featured standards-track transport protocol provides these features on its own. Therefore, these features are not considered in this document, with the exception of native authentication capabilities of TCP and SCTP for which the security considerations in RFC4895.

Like congestion control mechanisms, security mechanisms are difficult to design and implement correctly. It is hence recommended that
applications employ well-known standard security mechanisms such as DTLS, TLS or IPsec, rather than inventing their own.

11. References

11.1. Normative References


11.2. Informative References


Appendix A. Revision information

-00 This is an individual draft for the IETF community

-01 This draft was a result of walking away from the text for a few days and then reorganising the content.
-02 This draft fixes textual errors.

-03 This draft follows feedback from people reading this draft.

Comments from the community are welcome on the text and recommendations.

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The Impact of Transport Header Encryption on Operation and Evolution of
the Internet
draft-fairhurst-tsvwg-transport-encrypt-04

Abstract

This document describes implications of applying end-to-end
encryption at the transport layer. It identifies some in-network
uses of transport layer header information that can be used with a
transport header integrity check. It reviews the implication of
developing encrypted end-to-end transport protocols and examines the
implication of developing and deploying encrypted end-to-end
transport protocols. Since transport measurement and analysis of the
impact of network characteristics have been important to the design
of current transport protocols, it also considers some anticipated
implications on transport and application evolution.

Status of this Memo

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1. Introduction

This document discusses the implications of end-to-end encryption applied at the transport layer, and examines the impact on transport protocol design, usage, and network operations and management. It also considers anticipated implications on transport and application evolution.
The transport layer provides the first end-to-end interactions across the Internet. Transport protocols layer directly over the network-layer service and are sent in the payload of network-layer packets. They support end-to-end communication between applications, supported by higher-layer protocols, running on the end systems (or transport endpoint). This simple architectural view hides one of the core functions of the transport, however — to discover and adapt to the properties of the Internet path that is currently being used. The design of Internet transport protocols is as much about trying to avoid the unwanted side effects of congestion on a flow and other capacity-sharing flows, avoiding congestion collapse, adapting to changes in the path characteristics, etc., as it is about end-to-end feature negotiation, flow control and optimising for performance of a specific application.

To achieve stable Internet operations the IETF transport community has to date relied heavily on measurement and insights of the network operations community to understand the trade-offs, and to inform selection of select appropriate mechanisms, to ensure a safe, reliable and robust Internet. In turn, the network operations community relies on being able to understand the traffic passing over the Internet, both in aggregate and at the flow level — inspecting transport layer headers to help understand traffic dynamics.

There are many motivations for deploying encrypted transports, and encryption of transport payloads. The increasing public concerns about the interference with Internet traffic have led to a rapidly expanding deployment of encryption to protect end-user privacy, in protocols like QUIC. At the same time, network operators and access providers, especially in mobile networks, have come to rely on the in-network measurement of transport properties and the functionality provided by middleboxes to both support network operations and enhance performance.

This document considers some implications of working with encrypted transport protocols, and discusses trade-offs around authentication, encryption of transport protocol headers. It describes some of the architectural challenges and considerations in the way transport protocols are designed when using encryption [Measure].

Encryption of the transport layer brings some well-known privacy and security benefits, but also introduces various costs that need to be considered. Specifically, it can impact the following activities that rely on measurement and analysis of traffic flows:
- Network Operations and Research: Observable transport headers enable operators and the research community to measure and analyse protocol performance, network anomalies, and failure pathologies. This information can help inform capacity planning, and assist in determining the need for equipment and/or configuration changes by network operators. This data also can inform Internet engineering research, and help the development of new protocols and procedures. Encryption of the entire transport protocol, including header information, will restrict the availability of data, and might lead to the development of alternative, and potentially more intrusive, methods to acquire the needed data. Encrypting the transport payload, but leaving some, or all, of the transport headers unencrypted but authenticated can provide the majority of the privacy and security benefits while allowing some measurement.

- Network Troubleshooting and diagnostics: Encrypting transport header information eliminates the incentive for operators to troubleshoot what they cannot interpret. A flow experiencing packet loss looks like an unaffected flow when only observing network layer headers (if transport sequence numbers and flow identifiers are obscured). This limits understanding of the impact of packet loss on the flows that share a network segment. Encrypted traffic therefore implies "don’t touch", and a likely trouble-shooting response will be "can’t help, no trouble found". The additional mechanisms that will need to be introduced to help reconstruct transport-level metrics add complexity and operational costs [I-D.mm-wg-effect-encrypt].

- Network Traffic Analysis: The use of encryption can make it harder to determine which transport protocols and features are being used across a network segment. The trends in usage. This could impact the ability for an operator to anticipate the need for network upgrades and roll-out. It can also impact the on-going traffic engineering activities performed by operators. While the impact may, in many cases, be small there are scenarios where operators directly support particular services (e.g., in radio links, or to troubleshoot issues relating to Quality of Service, QoS). The more complex the underlying infrastructure the more important this impact.
- Open and Verifiable Network Data: The use of transport header encryption reduces the range of actors that can capture useful measurement data. This is, of course, its goal. Doing so, however, limits the information sources available to the Internet community to understand the operation of transport protocols, so preventing access to the information necessary to inform design decisions and standards for new protocols and related operational practices. There are dangers in a model where only endpoints (i.e., at user devices and within service platforms) can observe performance, and this cannot be independently verified. To ensure the health of the standards and research communities, we need independently captured data to develop on the behaviour of the transports. Independently verifiable performance metrics might also important in order to demonstrate regulatory compliance in some jurisdictions.

The last point leads us to consider the impact of encrypting all the transport headers the specification and development of protocols and standards. It has potential impact on:

- Understanding Feature Interactions: An appropriate vantage point, coupled with timing information about traffic flows, provides a valuable tool for benchmarking equipment and/or configurations, and to understand complex feature interactions. Transport header encryption limits the ability to diagnose and explore interactions between features at different protocol layers, a side-effect of not allowing a choice of vantage point from which this information is observed.

- Supporting Common Specifications: The Transmission Control Protocol (TCP) is the predominant transport protocol. Its many variants have broadly consistent approaches to avoiding congestion collapse, and to ensuring the stability of the network. Increased use of transport layer encryption can overcome ossification, allowing deployment of new transports with different types of congestion control. This flexibility can be beneficial, but it comes at the cost of fragmenting the ecosystem. There’s little doubt that developers will try to produce high quality transports for their target uses, but it is not clear there are sufficient incentives to ensure good practice that benefits the wide diversity of requirements for the Internet community as a whole. Increased diversity, and the ability to innovate without public scrutiny, risks point solutions that optimise for specific needs, but accidentally disrupt operations of/in different parts of the network. The social compact that maintains the stability of the network relies on accepting common specifications, and on the ability to verify that others also conform.
Operational practice: Published transport specifications allow operators to check compliance. This can bring assurance to those operating networks, often avoiding the need to deploy complex techniques that routinely monitor and manage TCP/IP traffic flows (e.g., avoiding the capital and operational costs of deploying flow rate-limiting and network circuit-breaker methods). This should continue when encrypted transport headers are used, but methods need to confirm that the traffic produced conforms to the expectations of the operator or developer.

Restricting research and development: The use of encryption may impede independent research into new mechanisms, measurement of behaviour, and development initiatives. Experience shows that transport protocols are complicated to design and complex to deploy, and that individual mechanisms need to be evaluated while considering other mechanism, across a broad range of network topologies and with attention to the impact on traffic sharing the capacity. Adopting pervasive encryption of transport information could eliminate the independent self-checks that have previously been in place from research and academic contributors (e.g., the role of the IRTF ICCRG, and research publications in reviewing new transport mechanisms and assessing the impact of their experimental deployment).

Pervasive use of transport header encryption can impact the ways that protocols are designed, standardised, deployed, and operated. The choice of whether future transport protocols encrypt their protocol headers therefore needs to be taken based not solely on security and privacy considerations, but also taking into account the impact on operations, standards, and research. A network that is secure but unusable due to persistent congestion collapse is not an improvement, and while that would be an extreme outcome proposals that impose high costs for very limited benefits need to be considered carefully, to ensure the benefits outweigh the costs.

1.1. Current uses of Transport Headers within the Network

The transport layer is the first end-to-end layer in the network stack. Despite headers having end-to-end meaning, some transport headers have come to be used in various ways within the Internet. In response to pervasive monitoring [RFC7624] revelations and the IETF consensus that "Pervasive Monitoring is an Attack" [RFC7258], efforts are underway to increase encryption of Internet traffic, which would prevent visibility of transport headers. This affects on how network protocols are designed and used [I-D.mm-wg-effect-encrypt]. To understand these implications, it is first necessary to understand how transport layer headers are currently observed and/or modified by middleboxes within the network.
Transport protocols can be designed to encrypt or authenticate transport header fields. Authentication methods at the transport layer can be sued to detect any changes to an immutable header field that were made by a network device along a path. The intentional modification of transport headers by middleboxes (such as Network Address Translation with Protocol Translation, NAT-PT, or Firewalls) is not considered.

1.1.1. Observing Transport Information in the Network

In-network observation of transport protocol headers requires knowledge of the format of the transport header:

- Flows need to be identified at the level required for monitoring;
- The protocol and version of the header need to be observable. As protocols evolve over time and there may be a need to introduce new transport headers. This may require interpretation of protocol version information or connection setup information;
- Location and syntax of any transport headers to be observed. IETF transport protocols specify this information.

The following subsections describe various ways that observable transport information may be utilised.

1.1.1.1. Flow Identification

Transport protocol header information can identify a flow and the connection state of the flow, together with the protocol options being used. In some usages, a low-numbered (well-known) port can identify a protocol (although port information alone is not sufficient to guarantee identification of a protocol). Transport protocols, such as TCP and Stream Control Transport Protocol (SCTP) specify a standard base header that includes sequence number information and other data, with the possibility to negotiate additional headers at connection setup, identified by an option number in the transport header. UDP-based protocols can use, but sometimes do not use, well-known ports. Some can instead be identified by signalling protocols or through the use of magic numbers placed in the first byte(s) of the datagram payload.

1.1.1.2. Metrics derived from Transport Layer Headers

Some actors have a need to characterise the performance of link/network segments. Passive monitoring uses observed traffic to make inferences from transport headers to derive these measurements. A variety of open source and commercial tools have been deployed that utilise this information. The following metrics can be derived from transport header information:
Traffic Rate and Volume: Header information may allow derivation of volume measures per-application, to characterise the traffic that uses a network segment or the pattern of network usage. This may be measured per endpoint or aggregate of endpoint (e.g., by an operator to assess subscriber usage). It can also be used to trigger measurement-based traffic shaping and to implement QoS support within the network and lower layers. Volume measures can be valuable for capacity planning (providing detail of trends rather than the volume per subscriber).

Loss Rate and Loss Pattern: Flow loss rate may be derived and is often used as a metric for performance assessment and to characterise transport behaviour. Understanding the root cause of loss can help an operator determine whether this requires corrective action.

There are various causes of loss, including: corruption on a link (e.g., interference on a radio link), buffer overflow (e.g., due to congestion), policing (traffic management), buffer management (e.g., Active Queue Management, AQM). Understanding flow loss rate requires either maintaining per flow packet counters or by observing sequence numbers in transport headers. Loss can be monitored at the interface level by devices in the network. It is often important to understand the conditions under which packet loss occurs. This usually requires relating loss to the traffic flowing on the network segment at the time of loss.

Observation of transport feedback information (observing loss reports, e.g., RTP Control Protocol (RTCP), TCP SACK) can increase understanding of the impact of loss and help identify cases where loss may have been wrongly identified, or the transport did not require the lost packet. It is sometimes more important to understand the pattern of loss, than the loss rate – since losses can often occur as bursts, rather than randomly-timed events.

Throughput and Goodput: The throughput observed by a flow can be determined even when a flow is encrypted, providing the individual flow can be identified. Goodput [RFC7928] is a measure of useful data exchanged (the ratio of useful/total volume of traffic sent by a flow), which requires ability to differentiate loss and retransmission of packets (e.g., by observing packet sequence numbers in the TCP or the Real Time Protocol, RTP, headers.
Latency: Latency is a key performance metric that impacts application response time and user-perceived response time. It often indirectly impacts throughput and flow completion time. Latency determines the reaction time of the transport protocol itself, impacting flow setup, congestion control, loss recovery, and other transport mechanisms. The observed latency can have many components [Latency]. Of these, unnecessary/unwanted queuing in network buffers has often been observed as a significant factor. Once the cause of unwanted latency has been identified, this can often be eliminated, and determining latency metrics is a key driver in the deployment of AQM [RFC7567], DiffServ [RFC2474], and Explicit Congestion Notification (ECN) [RFC3168] [RFC8087].

To measure latency across a part of the path, an observation point can measure the experienced round trip time (RTT) using packet sequence numbers, and acknowledgements, or by observing header timestamp information. Such information allows an observation point in the network to determine not only the path RTT, but also to measure the upstream and downstream contribution to the RTT. This may be used to locate a source of latency, e.g., by observing cases where the ratio of median to minimum RTT is large for a part of a path.

An example usage of this method could identify excessive buffers to help deploy or configure AQM [RFC7567] [RFC7928] to effectively eliminate unnecessary queuing in routers and other devices. AQM methods need to be deployed at the capacity bottleneck, but are often deployed in combination with other techniques, such as scheduling [RFC7567] [I-D.ietf-aqm-fq-codel] and although parameter-less methods are desired [RFC7567], current methods [I-D.ietf-aqm-fq-codel] [I-D.ietf-aqm-codel] [I-D.ietf-aqm-pie] often cannot scale across all possible deployment scenarios. The service offered by operators can therefore benefit from latency information to understand the impact of deployment and tune deployed services.

Jitter: Some network applications are sensitive to changes in packet timing. For such applications, it can be necessary to measure the jitter observed along a portion of the path. The requirements to measure jitter resemble those for the measurement of latency.

Flow Reordering: Significant flow reordering can impact time-critical applications and can be interpreted as loss by reliable transports. Many transport protocol techniques are impacted by reordering (e.g., triggering TCP retransmission, or re-buffering...
of real-time applications). Packet reordering can occur for many reasons (from equipment design to misconfiguration of forwarding rules).

As in the drive to reduce network latency, there is a need for operational tools to detect mis-ordered packet flows and quantify the degree of reordering. Techniques for measuring reordering typically observe packet sequence numbers. Metrics have been defined that evaluate whether a network has maintained packet order on a packet-by-packet basis [RFC4737] and [RFC5236].

There has been initiatives in the IETF transport area to reduce the impact of reordering within a transport flow, possibly leading to reduced the requirements for ordering. These have promise to simplify network equipment design as well as the potential to improve robustness of the transport service. Measurements of reordering can help understand the level of reordering within deployed infrastructure, and inform decisions about how to progress such mechanisms.

Some protocols provide in-built monitoring and reporting functions. Transport fields in the RTP header [RFC3550] [RFC4585] can be observed to derive traffic volume measurements and provide information on the progress and quality of a session using RTP. Key performance indicators are retransmission rate, packet drop rate, sector utilization level, a measure of reordering, peak rate, the CE-marking rate, etc. Metadata is often important to understand the context under which the data was collected, including the time, observation point, and way in which metrics were accumulated. The RTCP protocol directly reports some of this information in a form that can be directly visible in the network. A user of summary measurement data needs to trust the source of this data and the method used to generate the summary information.

When encryption conceals information in packet headers, measurements need to rely on pattern inferences and other heuristics grows, and accuracy suffers [I-D.mm-wg-effect-encrypt].

1.1.1.3. Metrics derived from Network Layer Headers

Some transport information is made visible in the network-layer protocol header. These header fields are not encrypted and can be used to make flow observations.

Use of IPv6 Network-Layer Flow Label: Endpoints are encouraged expose flow information in the IPv6 Flow Label field of the network-layer header (e.g. [RFC8085]). This can be used to inform network-layer queuing, forwarding (e.g., for equal cost multi-path (ECMP) routing, and Link Aggregation, LAG). This can provide useful information to assign packets to flows in the data collected by measurement campaigns. Although important to characterising a path, it does not directly provide any performance data.
Use Network-Layer Differentiated Services Code Point: Application can expose their delivery expectations to the network by setting the Differentiated Services Code Point (DSCP) field of IPv4 and IPv6 packets. This can be used to inform network-layer queuing and forwarding, and can also provide information on the relative importance of packet information collected by measurement campaigns, but does not directly provide any performance data.

This field provides explicit information that can be used in place of inferring traffic requirements (e.g., by inferring QoS requirements from port information via a multi-field classifier). The DSCP value can therefore impact the quality of experience for a flow. Observations of service performance need to consider this field when a network path has support for differentiated service treatment.

Use of Explicit Congestion Marking: ECN[RFC3168] is an optional transport mechanism that uses a code point in the network-layer header. Use of ECN can offer gains in terms of increased throughput, reduced delay, and other benefits when used over a path that includes equipment that supports an AQM method that performs Congestion Experienced (CE) marking of IP packets [RFC8087].

ECN exposes the presence of congestion on a network path to the transport and network layer. The reception of CE-marked packets can therefore be used to monitor the presence and estimate the level of incipient congestion on the upstream portion of the path from the point of observation (Section 2.5 of [RFC8087]). Because ECN marks carried in the IP protocol header, it is much easier to measure ECN than metering packet loss. However, interpreting the marking behaviour (i.e., assessing congestion and diagnosing faults) requires context from the transport layer (path RTT, visibility of loss - that could be due to queue overflow, congestion response, etc) [RFC7567].

Some ECN-capable network devices can provide richer (more frequent and fine-grained) indication of their congestion state. Setting congestion marks proportional to the level of congestion (e.g., Data Center TCP, DCTP [I-D.ietf-tcpm-dctcp], and Low Latency Low Loss Scalable throughput, L4S, [I-D.ietf-tsvwg-l4s-arch].

Use of ECN requires feedback a transport to feed back reception information on the path towards the data sender. Exposure of this Transport ECN feedback provides an additional powerful tool to understand ECN-enabled AQM-based networks [RFC8087].
AQM and ECN offer a range of algorithms and configuration options, it is therefore important for tools to be available to network operators and researchers to understand the implication of configuration choices and transport behaviour as use of ECN increases and new methods emerge [RFC7567] [RFC8087]. ECN-monitoring is expected to become important as AQM is deployed that supports ECN [RFC8087].

1.1.2. Transport Measurement

The common language between network operators and application/content providers/users is packet transfer performance at a layer that all can view and analyse. For most packets, this has been transport layer, until the emergence of QUIC, with the obvious exception of VPNs and IPsec. When encryption conceals more layers in a packet, people seeking understanding of the network operation need to rely more on pattern inferences and other heuristics. The accuracy of measurements therefore suffers, as does the ability to investigate and troubleshoot interactions between different anomalies. For example, the traffic patterns between a web server and a browser are dependent on browser supplier and version, even use of the application (e.g., web e-mail access). Even when measurement datasets are made available (e.g., from endpoints) additional metadata, such as the state of the network, is often required to interpret the data. Collecting and coordinating such metadata is more difficult when the observation point is at a different location to the bottleneck/device under evaluation.

Packet sampling techniques can be used to scale the processing involved in observing packets on high rate links. This exports only the packet header information of (randomly) selected packets. The utility of these measurements depends on the type of bearer and number of mechanisms used by network devices. Simple routers are relatively easy to manage, a device with more complexity demands understanding of the choice of many system parameters. This level of complexity exists when several network methods are combined.

This section discusses topics concerning observation of transport flows, with a focus on transport measurement.
1.1.2.1. Point of Measurement

Often measurements can only be understood in the context of the other flows that share a bottleneck. A simple example is monitoring of AQM. For example, FQ-CODEL [I-D.ietf-aqm-fq-codel], combines sub queues (statistically assigned per flow), management of the queue length (CODEL), flow-scheduling, and a starvation prevention mechanism. Usually such algorithms are designed to be self-tuning, but current methods typically employ heuristics that can result in more loss under certain path conditions (e.g., large RTT, effects of multiple bottlenecks [RFC7567]).

In-network measurements can distinguish between upstream and downstream metrics with respect to the measurement point. These are particularly useful for locating the source of problems or to assess the performance of a network segment or a particular device configuration.

By correlating observations at multiple points along the path (e.g., at the ingress and egress of a network segment), an observer can determine the contribution of a portion of the path to an observed metric (to locate a source of delay, jitter, loss, reordering, congestion marking, etc.).

1.1.2.2. Use by Operators to Plan and Provision Networks

Traffic measurements (e.g., traffic volume, loss, latency) is used by operators to help plan deployment of new equipment and configurations in their networks. Data is also important to equipment vendors who need to understand traffic trends, traffic and patterns of usage as inputs to decisions about planning products and provisioning for new deployments. This measurement information can also be correlated with billing information when this is also collected by an operator.

A network operator supporting traffic that uses transport header encryption may not have access to per-flow measurement data. Trends in aggregate traffic can be observed and can be related this to the endpoint addresses being used, but it may not be possible to correlate patterns in measurements with changes in transport protocols (e.g., the impact of changes in introducing a new transport protocol mechanism). This increases the dependency on other indirect sources of information to inform planning and provisioning.

1.1.2.3. Service Performance Measurement

Traffic measurements (e.g., traffic volume, loss, latency) can be used by various actors to help analyse the performance available to users of a network segment, and inform operational practice. While active measurements may be used in-network passive measurements can have advantages in terms of eliminating unproductive traffic, reducing the influence of test traffic on the overall traffic mix, and the ability to choose the point of measurement Section 1.1.2.1.

1.1.2.4. Measuring Transport to Support Network Operations
Information provided by tools observing transport headers can help determine whether mechanisms are needed in the network to prevent flows from acquiring excessive network capacity. Operators can implement operational practices to manage traffic flows (e.g., to prevent flows from acquiring excessive network capacity under severe congestion) by deploying rate-limiters, traffic shaping or network transport circuit breakers [RFC8084].

Congestion Control Compliance of Traffic: Congestion control is a key transport function. Many network operators implicitly accept that TCP traffic to comply with a behaviour that is acceptable for use in the shared Internet. TCP algorithms have been continuously improved over decades, and they have reached a level of efficiency and correctness that custom application-layer mechanisms will struggle to easily duplicate [RFC8085].

A standards-compliant TCP stack provides congestion control may therefore be judged safe for use across the Internet. Applications developed on top of well-designed transports can be expected to appropriately control their network usage, reacting when the network experiences congestion, by back-off and reduce the load placed on the network. This is the normal expected behaviour for TCP and SCTP.

However when anomalies are detected, tools can interpret the transport protocol header information to help understand the impact of specific transport protocols (or protocol mechanisms) on the other traffic that shares a network. An observation in the network can gain understanding of the dynamics of a flow and its congestion control behaviour. Analysing observed packet sequence numbers can be used to help build confidence that an application flow backs-off its share of the network load in the face of persistent congestion, and hence to understand whether the behaviour is appropriate for sharing limited network capacity. For example, it is common to visualise plots of TCP sequence numbers versus time for a flow to understand how a flow shares available capacity, deduce its dynamics in response to congestion, etc.

Congestion Control Compliance for UDP Traffic: UDP provides a minimal message-passing transport that has no inherent congestion control mechanisms. Because congestion control is critical to the stable operation of the Internet, applications and other protocols that choose to use UDP as an Internet transport are required to employ mechanisms to prevent congestion collapse, avoid unacceptable contributions to jitter/latency, and to establish an acceptable share of capacity with concurrent traffic [RFC8085].

A network operator needs tools to understand if UDP flows comply with congestion control expectations and therefore whether there...
is a need to deploy methods such as rate-limiters, transport
circuit breakers or other methods to enforce acceptable usage for
the offered service.

UDP flows that expose a well-known header by specifying the format
of header fields can allow information to be observed to gain
understanding of the dynamics of a flow and its congestion control
behaviour. For example, tools exist to monitor various aspects of
the RTP and RTCP header information of real-time flows (see
Section 1.1.1.2.

1.1.3. Use for Network Diagnostics and Troubleshooting

Transport header information is useful for a variety of operational
tasks [I-D.mm-wg-effect-encrypt]: to diagnose network problems,
assess performance, capacity planning, management of denial of
service threats, and responding to user performance questions. These
tasks seldom involve the need to determine the contents of the
transport payload, or other application details.

A network operator supporting traffic that uses transport header
encryption can see only encrypted transport headers. This prevents
deployment of performance measurement tools that rely on transport
protocol information. Choosing to encrypt all information may be
expected to reduce the ability for networks to "help" (e.g., in
response to tracing issues, making appropriate Quality of Service,
QoS, decisions). For some this will be blessing, for others it may be
a curse. For example, operational performance data about encrypted
flows needs to be determined by traffic pattern analysis, rather than
relying on traditional tools. This can impact the ability of the
operator to respond to faults, it could require reliance on endpoint
diagnostic tools or user involvement in diagnosing and
troubleshooting unusual use cases or non-trivial problems. A key
need here is that tools need to provide useful information during
network anomalies (e.g., significant reordering, high or intermittent
loss). Although many network operators utilise transport information
as a part of their operational practice, the network will not break
because transport headers are encrypted.

1.1.4. Observing Headers to Implement Network Policy

Information from the transport protocol can be used by a multi-field
classifier as a part of policy framework. Policies are commonly used
for QoS management for resource-constrained networks and by firewalls
that use the information to implement access rules. Traffic that
cannot be classified, will typically receive a default treatment.

2. Encryption and Authentication of Transport Headers
End-to-end encryption can be applied at various protocol layers. It can be applied above the transport to encrypt the transport payload. Encryption methods can hide information from an eavesdropper in the network. Encryption can also help protect the privacy of a user, by hiding data relating to user/device identity or location. Neither an integrity check nor encryption methods prevent traffic analysis, and usage needs to reflect that profiling of users, identification of location and fingerprinting of behaviour can take place even on encrypted traffic flows.

One motive to use encryption is a response to perceptions that the network has become ossified by over-reliance on middleboxes that prevent new protocols and mechanisms from being deployed. This has lead to a common perception that there is too much "manipulation" of protocol headers within the network, and that designing to deploy in such networks is preventing transport evolution. In the light of this, a method that authenticates transport headers may help improve the pace of transport development, by eliminating the need to always consider deployed middleboxes [I-D.trammell-plus-abstract-mech], or potentially to only explicitly enable middlebox use for particular paths with particular middleboxes that are deliberately deployed to realise a useful function for the network and/or users[RFC3135].

Another motivation stems from increased concerns about privacy and surveillance. Some Internet users have valued the ability to protect identity, user location, and defend against traffic analysis, and have used methods such as IPsec ESP and Tor [Tor]. Revelations about the use of pervasive surveillance [RFC7624] have, to some extent, eroded trust in the service offered by network operators, and following the Snowden revelation in the USA in 2013 has led to an increased desire for people to employ encryption to avoid unwanted "eavesdropping" on their communications. Whatever the reasons, there are now activities in the IETF to design new protocols that may include some form of transport header encryption (e.g., QUIC [I-D.ietf-quic-transport]).

Authentication methods (that provide integrity checks of protocols fields) have also been specified at the network layer, and this also protects transport header fields. The network layer itself carries protocol header fields that are increasingly used to help forwarding decisions reflect the need of transport protocols, such the IPv6 Flow Label [RFC6437], the Differentiated Services Code Point (DSCP) [RFC2474] and Explicit Congestion Notification (ECN) [RFC3168].

The use of transport layer authentication and encryption exposes a tussle between middlebox vendors, operators, applications developers and users.

- On the one hand, future Internet protocols that enable large-scale encryption assist in the restoration of the end-to-end nature of the Internet by returning complex processing to the endpoints, since middleboxes cannot modify what they cannot see.
o On the other hand, encryption of transport layer header information has implications for people who are responsible for operating networks and researchers and analysts seeking to understand the dynamics of protocols and traffic patterns.

Whatever the motives, a decision to use pervasive of transport header encryption will have implications on the way in which design and evaluation is performed, and which can in turn impact the direction of evolution of the TCP/IP stack.

The next subsections briefly review some security design options for transport protocols.

2.1. Authenticating the Transport Protocol Header

Transport layer header information can be authenticated. An integrity check that protects the immutable transport header fields, but can still expose the transport protocol header information in the clear, allowing in-network devices to observes these fields. An integrity check can not prevent in-network modification, but can avoid a receiving accepting changes and avoid impact on the transport protocol operation.

An example transport authentication mechanism is TCP-Authentication (TCP-AO) [RFC5925]. This TCP option authenticates TCP segments, including the IP pseudo header, TCP header, and TCP data. TCP-AO protects the transport layer, preventing attacks from disabling the TCP connection itself. TCP-AO may interact with middleboxes, depending on their behaviour [RFC3234].

The IPsec Authentication Header (AH) [RFC4302] works at the network layer and authenticates the IP payload. This therefore also authenticates all transport headers, and verifies their integrity at the receiver, preventing in-network modification.

2.2. Encrypting the Transport Payload

The transport layer payload can be encrypted to protect the content of transport segments. This leaves transport protocol header information in the clear. The integrity of immutable transport
header fields could be protected by combining this with an integrity check (Section 2.1).

Examples of encrypting the payload include Transport Layer Security (TLS) over TCP [RFC5246] [RFC7525] or Datagram TLS (DTLS) over UDP [RFC6347] [RFC7525].

2.3. Encrypting the Transport Header

The network layer payload could be encrypted (including the entire transport header and payload). This method does not expose any transport information to devices in the network, which also prevents modification along the network path.

The IPsec Encapsulating Security Payload (ESP) [RFC4303] is an example of encryption at the network layer, it encrypts and authenticates all transport headers, preventing visibility of the headers by in-network devices. Some Virtual Private Network (VPN) methods also encrypt these headers.

2.4. Authenticating Transport Information and Selectively Encrypting the Transport Header

A transport protocol design can encrypt selected header fields, while also choosing to authenticate fields in the transport header. This allows specific transport header fields to be made observable by network devices. End-to-end integrity checks can prevent an endpoint from undetected modification of the immutable transport headers.

The choice of which fields to expose and which to encrypt is a design choice for the transport protocol. Any selective encryption method requires trading two conflicting goals for a transport protocol designer to decide which header fields to encrypt. On the one hand, security work typically employs a design technique that seeks to expose only what is needed. On the other hand, there may be performance and operational benefits in exposing selected information to network tools.

Mutable fields in the transport header provide opportunities for middleboxes to modify the transport behaviour (e.g., the extended headers described in [I-D.trammell-plus-abstract-mech]). This considers only immutable fields in the transport headers, that is, fields that may be authenticated end-to-end across a path.

An example of a method that encrypts some, but not all, transport information is GRE-in-UDP [RFC8086] when used with GRE encryption.

2.5. Adding Transport Information to Network-Layer Protocol Headers
The transport information can be made visible in a network-layer header. This has the advantage that this information can then be observed by in-network devices. This has the advantage that a single header can support all transport protocols, but there may also be less desirable implications of separating the operation of the transport protocol from the measurement framework.

Some measurements may be made by adding additional protocol headers carrying operations, administration and management (OAM) information to packets at the ingress to a maintenance domain (e.g., an Ethernet protocol header with timestamps and sequence number information using a method such as 802.11ag) and removing the additional header at the egress of the maintenance domain. This approach enables some types of measurements, but does not cover the entire range of measurements described in this document.

Another example of a network-layer approach is the IPv6 Performance and Diagnostic Metrics (PDM) Destination Option [I-D.ietf-ippm-6man-pdm-option]. This allows a sender to optionally include a destination option that carries header fields that can be used to observe timestamps and packet sequence numbers. This information could be authenticated by receiving transport endpoints when the information is added at the sender and visible at the receiving endpoint, although methods to do this have not currently been proposed. This method needs to be explicitly enabled at the sender.

A drawback of using extension headers is that IPv4 network options are often not supported (or are carried on a slower processing path) and some IPv6 networks are also known to drop packets that set an IPv6 header extension. Another disadvantage is that protocols that separately expose header information do not necessarily have an advantage to expose the information that is utilised by the protocol itself, and could manipulate this header information to gain an advantage from the network.

3. Implications of Protecting the Transport Headers

This section explores key implications of working with encrypted transport protocols.

3.1. Independent Measurement

Independent observation by multiple actors is important for scientific analysis. Encrypting transport header encryption changes the ability for other actors to collect and independently analyse data. Internet transport protocols employ a set of mechanisms. Some
of these need to work in cooperation with the network layer - loss detection and recovery, congestion detection and congestion control, some of these need to work only end-to-end (e.g., parameter negotiation, flow-control).

When encryption conceals information in the transport header, it could be possible for an applications to provide summary data on performance and usage of the network. This data could be made available to other actors. However, this data needs to contain sufficient detail to understand (and possibly reconstruct the network traffic pattern for further testing) and to be correlated with the configuration of the network paths being measured. Sharing information between actors needs also to consider the privacy of the user and the incentives for providing accurate and detailed information. Protocols that expose the state information used by the transport protocol in their header information (e.g., timestamps used to calculate the RTT, packet numbers used to assess congestion and requests for retransmission) provide an incentive for the sending endpoint to provide correct information, increasing confidence that the observer understands the transport interaction with the network. This becomes important when considering changes to transport protocols, changes in network infrastructure, or the emergence of new traffic patterns.

3.2. Characterising "Unknown" Network Traffic

The patterns and types of traffic that share Internet capacity changes with time as networked applications, usage patterns and protocols continue to evolve.

If "unknown" or "uncharacterised" traffic patterns form a small part of the traffic aggregate passing through a network device or segment of the network the path, the dynamics of the uncharacterised traffic may not have a significant collateral impact on the performance of other traffic that shares this network segment. Once the proportion of this traffic increases, the need to monitor the traffic and determine if appropriate safety measures need to be put in place.

Tracking the impact of new mechanisms and protocols requires traffic volume to be measured and new transport behaviours to be identified. This is especially true of protocols operating over a UDP substrate. The level and style of encryption needs to be considered in determining how this activity is performed. On a shorter timescale, information may also need to be collected to manage denial of service attacks against the infrastructure.

3.3. Accountability and Internet Transport Protocols
Information provided by tools observing transport headers can help determine whether mechanisms are needed in the network to prevent flows from acquiring excessive network capacity, and where needed to deploy appropriate tools Section 1.1.2.4. Obfuscating or hiding this information using encryption is expected to lead operators and maintainers of middleboxes (firewalls, etc.) to seek other methods to classify and mechanisms to condition network traffic. A lack of data seems likely to reduce the level of precision with which these mechanisms are applied, and this needs to be considered when evaluating the impact of designs for transport encryption.

3.4. Impact on Research, Development and Deployment

Measurement data is increasingly being used to inform design decisions in networking research, during development of new mechanisms and protocols and in standardisation. Measurement has a critical role in the design of transport protocol mechanisms and their acceptance by the wider community (e.g., as a method to judge the safety for Internet deployment). Observation of pathologies are also important in understanding the interactions between cooperating protocols and network mechanism, the implications of sharing capacity with other traffic and the impact of different patterns of usage.

Attention needs to be paid to the expected scale of deployment of new protocols and protocol mechanisms. Whatever the mechanism, experience has shown that it is often difficult to correctly implement combination of mechanisms [RFC8085]. These mechanisms therefore typically evolve as a protocol matures, or in response to changes in network conditions, changes in network traffic or changes to application usage.

The growth and diversity of applications and protocols using the Internet continues to expand - and there has been recent interest in a wide range of new transport methods, e.g., Larger Initial Window, Proportional Rate Reduction (PRR), congestion control methods based on measuring bottleneck bandwidth and round-trip propagation time, the introduction of AQM techniques and new forms of ECN response (e.g., Data Centre TCP, DCTP [I-D.ietf-tcpm-dctcp], and methods proposed for Low Latency Low Loss Scalable throughput, L4S). For each new method it is desirable to build a body of data reflecting its behaviour under a wide range of deployment scenarios, traffic load, and interactions with other deployed/candidate methods.

Open standards motivate a desire for this evaluation to include independent observation and evaluation of performance data, which in turn suggests control over where and when measurement samples are collected. This requires consideration of the appropriate balance between encrypting all and no transport information.

4. Acknowledgements

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5. Security Considerations

This document is about design and deployment considerations for transport protocols. Authentication, confidentiality protection, and integrity protection are identified as Transport Features by RFC8095. As currently deployed in the Internet, these features are generally provided by a protocol or layer on top of the transport protocol; no current full-featured standards-track transport protocol provides these features on its own. Therefore, these features are not considered in this document, with the exception of native authentication capabilities of TCP and SCTP for which the security considerations in RFC4895.

Open data, and accessibility to tools that can help understand trends in application deployment, network traffic and usage patterns can all contribute to understanding security challenges. Standard protocols and understanding of the interactions between mechanisms and traffic patterns can also provide valuable insight into appropriate security design. Like congestion control mechanisms, security mechanisms are difficult to design and implement correctly. It is hence recommended that applications employ well-known standard security mechanisms such as DTLS, TLS or IPsec, rather than inventing their own.

6. IANA Considerations

XX RFC ED - PLEASE REMOVE THIS SECTION XXX

This memo includes no request to IANA.

7. References

7.1. Normative References


7.2. Informative References


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[I-D.ietf-tcpm-dctcp]

[I-D.ietf-tsvwg-l4s-arch]

[I-D.mm-wg-effect-encrypt]

[I-D.trammell-plus-abstract-mech]

[I-D.trammell-plus-statefulness]


Fairhurst & Perkins Expires March 29, 2018 [Page 25]


This is an individual draft for the IETF community.

This draft was a result of walking away from the text for a few days and then reorganising the content.

This draft fixes textual errors.

This draft follows feedback from people reading this draft.

This adds an additional contributor and includes significant reworking to ready this for review by the wider IETF community Colin Perkins joined the author list.

Comments from the community are welcome on the text and recommendations.

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Abstract

This memo specifies a component-based architecture for manufacturer usage descriptions (MUD). The goal of MUD is to provide a means for Things to signal to the network what sort of access and network functionality they require to properly function. The initial focus is on access control. Later work can delve into other aspects.

This memo specifies two YANG modules, IPv4 and IPv6 DHCP options, an LLDP TLV, a URL suffix specification, an X.509 certificate extension and a means to sign and verify the descriptions.

Status of This Memo

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1. Introduction

The Internet has largely been constructed on general purpose computers, those devices that may be used for a purpose that is specified by those who buy the device. [RFC1984] presumed that an end device would be most capable of protecting itself. This made sense when the typical device was a workstation or a mainframe, and it continues to make sense for general purpose computing devices today, including laptops, smart phones, and tablets.

[RFC7452] discusses design patterns for, and poses questions about, smart objects. Let us then posit a group of objects that are specifically NOT general purpose computers. These devices have a specific purpose. By definition, therefore, all other uses are NOT intended. The combination of these two statements can be restated as a manufacturer usage description (MUD) that can be applied at various points within a network. Although this memo may seem to stress access requirements, usage intent also consists of quality of service needs a device may have.

We use the notion of "manufacturer" loosely in this context to refer to the entity or organization that will state how a device is intended to be used. In the context of a lightbulb, this might indeed be the lightbulb manufacturer. In the context of a smarter

device that has a built-in Linux stack, it might be an integrator of that device. The key points are that the device itself is expected to serve a limited purpose, and that there may exist an organization in the supply chain of that device that will take responsibility for informing the network about that purpose.

The intent of MUD is to solve for the following problems:

- Substantially reduce the threat surface on a device entering a network to those communications intended by the manufacturer.
- Provide for a means to scale network policies to the ever-increasing number types of devices in the network.
- Provide a means to address at least some vulnerabilities in a way that is faster than it might take to update systems. This will be particularly true for systems that are no longer supported by their manufacturer.
- Keep the cost of implementation of such a system to the bare minimum.
- Provide a means of extensibility for manufacturers to express other device capabilities or requirements.

MUD consists of three architectural building blocks:

- A classifier that a device emits that can be used to locate a description;
- The description itself, including how it is interpreted, and;
- A means for local network management systems to retrieve the description.

In this specification we describe each of these building blocks and how they are intended to be used together. However, they may also be used separately, independent of this specification, by local deployments for their own purposes.

1.1. What MUD doesn’t do

MUD is not intended to address network authorization of general purpose computers, as their manufacturers cannot envision a specific communication pattern to describe. In addition, even those devices that have a single or small number of uses might have very broad communication patterns. MUD on its own is not for them either.
No matter how good a MUD-enabled network is, it will never replace the need for manufacturers to patch vulnerabilities. It may, however, provide network administrators with some additional protection when those vulnerabilities exist.

Finally, no matter what the manufacturer specifies in a MUD file, these are not directives, but suggestions. How they are instantiated locally will depend on many factors and will be ultimately up to the local network administrator, who must decide what is appropriate in a given circumstances.

1.2. A Simple Example

A light bulb is intended to light a room. It may be remotely controlled through the network, and it may make use of a rendezvous service of some form that an app on smart phone accesses. What we can say about that light bulb, then, is that all other network access is unwanted. It will not contact a news service, nor speak to the refrigerator, and it has no need of a printer or other devices. It has no social networking friends. Therefore, an access list applied to it that states that it will only connect to the single rendezvous service will not impede the light bulb in performing its function, while at the same time allowing the network to provide both it and other devices an additional layer of protection.

1.3. Determining Intended Use

The notion of intended use is in itself not new. Network administrators apply access lists every day to allow for only such use. This notion of white listing was well described by Chapman and Zwicky in [FW95]. Profiling systems that make use of heuristics to identify types of systems have existed for years as well.

A Thing could just as easily tell the network what sort of access it requires without going into what sort of system it is. This would, in effect, be the converse of [RFC7488]. In seeking a general purpose solution, however, we assume that a device has so few capabilities that it will implement the least necessary capabilities to function properly. This is a basic economic constraint. Unless the network would refuse access to such a device, its developers would have no reason to provide the network any information. To date, such an assertion has held true.

1.4. Finding A Policy: The MUD URL

Our work begins with the device emitting a Universal Resource Locator (URL) [RFC3986]. This URL serves both to classify the device type and to provide a means to locate a policy file.
In this memo three means are defined to emit the MUD URL. One is a DHCP option[RFC2131],[RFC3315] that the DHCP client uses to inform the DHCP server. The DHCP server may take further actions, such as retrieve the URL or otherwise pass it along to network management system or controller. The second method defined is an X.509 constraint. The IEEE has developed [IEEE8021AR] that provides a certificate-based approach to communicate device characteristics, which itself relies on [RFC5280]. The MUD URL extension is non-critical, as required by IEEE 802.1AR. Various means may be used to communicate that certificate, including Tunnel Extensible Authentication Protocol (TEAP) [RFC7170]. Finally, a Link Layer Discovery Protocol (LLDP) frame is defined [IEEE8021AB].

It is possible that there may be other means for a MUD URL to be learned by a network. For instance, some devices may already be fielded or have very limited ability to communicate a MUD URL, and yet can be identified through some means, such as a serial number or a public key. In these cases, manufacturers may be able to map those identifiers to particular MUD URLs (or even the files themselves). Similarly, there may be alternative resolution mechanisms available for situations where Internet connectivity is limited or does not exist. Such mechanisms are not described in this memo, but are possible. Implementors should allow for this sort of flexibility of how MUD URLs may be learned.

1.5. Types of Policies

When the MUD URL is resolved, the MUD controller retrieves a file that describes what sort of communications a device is designed to have. The manufacturer may specify either specific hosts for cloud based services or certain classes for access within an operational network. An example of a class might be "devices of a specified manufacturer type", where the manufacturer type itself is indicated simply by the authority component (e.g, the domain name) of the MUD URL. Another example might be to allow or disallow local access. Just like other policies, these may be combined. For example:

- Allow access to devices of the same manufacturer
- Allow access to and from controllers via Constrained Application Protocol (COAP)[RFC7252]
- Allow access to local DNS/NTP
- Deny all other access

A printer might have a description that states:
o Allow access for port IPP or port LPD
o Allow local access for port HTTP
o Deny all other access

In this way anyone can print to the printer, but local access would be required for the management interface.

The files that are retrieved are intended to be closely aligned to existing network architectures so that they are easy to deploy. We make use of YANG [RFC7950] because of the time and effort spent to develop accurate and adequate models for use by network devices. JSON is used as a serialization for compactness and readability, relative to XML. Other formats may be chosen with later versions of MUD.

While the policy examples given here focus on access control, this is not intended to be the sole focus. By structuring the model described in this document with clear extension points, other descriptions could be included. One that often comes to mind is quality of service.

The YANG modules specified here are extensions of [I-D.ietf-netmod-acl-model]. The extensions to this model allow for a manufacturer to express classes of systems that a manufacturer would find necessary for the proper function of the device. Two modules are specified. The first module specifies a means for domain names to be used in ACLs so that devices that have their controllers in the cloud may be appropriately authorized with domain names, where the mapping of those names to addresses may rapidly change.

The other module abstracts away IP addresses into certain classes that are instantiated into actual IP addresses through local processing. Through these classes, manufacturers can specify how the device is designed to communicate, so that network elements can be configured by local systems that have local topological knowledge. That is, the deployment populates the classes that the manufacturer specifies. The abstractions below map to zero or more hosts, as follows:

Manufacturer: A device made by a particular manufacturer, as identified by the authority component of its MUD URL

same-manufacturer: Devices that have the same authority component of their MUD URL.
Controller: Devices that the local network administrator admits to the particular class.

my-controller: Devices associated with the MUD URL of a device that the administrator admits.

local: The class of IP addresses that are scoped within some administrative boundary. By default it is suggested that this be the local subnet.

The "manufacturer" classes can be easily specified by the manufacturer, whereas controller classes are initially envisioned to be specified by the administrator.

Because manufacturers do not know who will be using their devices, it is important for functionality referenced in usage descriptions to be relatively ubiquitous and mature. For these reasons only a limited subset YANG-based configuration of is permitted in a MUD file.

1.6. Terminology

MUD: manufacturer usage description.

MUD file: a file containing YANG-based JSON that describes a Thing and associated suggested specific network behavior.

MUD file server: a web server that hosts a MUD file.

MUD controller: the system that requests and receives the MUD file from the MUD server. After it has processed a MUD file, it may direct changes to relevant network elements.

MUD URL: a URL that can be used by the MUD controller to receive the MUD file.

Thing: the device emitting a MUD URL.

Manufacturer: the entity that configures the Thing to emit the MUD URL and the one who asserts a recommendation in a MUD file. The manufacturer might not always be the entity that constructs a Thing. It could, for instance, be a systems integrator, or even a component provider.

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in [RFC2119].
1.7. The Manufacturer Usage Description Architecture

With these components laid out we now have the basis for an architecture. This leads us to ASCII art.

```
.......................................
.......................................
................. MUD Controller ...........
................. -->get URL--> .......... MUD File Server
.......................................
.......................................
            End system network
            ____________________________
........................         .           _____________
........................         |            |  .          |             |
........................         |    MUD     |-->get URL-->|   MUD       |
........................         | Controller |  .(https)   | File Server |
........................  End system network |____________|<-MUD file<-<|_____________|
.......................................
.......................................
........................         .( dhcp et al) | router |  .
........................         Thing ---->MUD URL---> | or    |  .
........................         .               | switch |  .
.......................................
.......................................
.......................................
.......................................
```

Figure 1: MUD Architecture

In the above diagram, the switch or router collects MUD URLs and forwards them to the network management system for processing. This happens in different ways, depending on how the URL is communicated. For instance, in the case of DHCP, the DHCP server might receive the URL and then process it. In the case of IEEE 802.1X, the switch would carry the URL via a certificate to the authentication server via EAP over Radius [RFC3748], which would then process it. One method to do this is TEAP, described in [RFC7170]. The certificate extension is described below.

The information returned by the web site is valid for the duration of the Thing’s connection, or as specified in the description. Thus if the Thing is disconnected, any associated configuration in the switch can be removed. Similarly, from time to time the description may be refreshed, based on new capabilities or communication patterns or vulnerabilities.

The web site is typically run by or on behalf of the manufacturer. Its domain name is that of the authority found in the MUD URL. For legacy cases where Things cannot emit a URL, if the switch is able to determine the appropriate URL, it may proxy it, the trivial cases being a hardcoded MUD-URL on a switch port, or a mapping from some available identifier such as an L2 address or certificate hash to a MUD-URL.
The role of the MUD controller in this environment is to do the following:

- receive MUD URLs,
- retrieve MUD files,
- translate abstractions in the MUD files to specific network element configuration,
- maintain and update any required mappings of the abstractions, and
- update network elements with appropriate configuration.

A MUD controller may be a component of a AAA or network management system. Communication within those systems and from those systems to network elements is beyond the scope of this memo.

1.8. Order of operations

As mentioned above, MUD contains architectural building blocks, and so order of operation may vary. However, here is one clear intended example:

1. Thing emits URL.

2. That URL is forwarded to a MUD controller by the nearest switch (how this happens depends on the way in which the MUD URL is emitted).

3. The MUD controller retrieves the MUD file and signature from the MUD file server, assuming it doesn’t already have copies. After validating the signature, it may test the URL against a web or domain reputation service, and it may test any hosts within the file against those reputation services, as it deems fit.

4. The MUD controller may query the administrator for permission to add the Thing and associated policy. If the Thing is known or the Thing type is known, it may skip this step.

5. The MUD controller instantiates local configuration based on the abstractions defined in this document.

6. The MUD controller configures the switch nearest the Thing. Other systems may be configured as well.

7. When the Thing disconnects, policy is removed.
2. The MUD Model and Semantic Meaning

A MUD file consists of a YANG model that has been serialized in JSON [RFC7951]. For purposes of MUD, the nodes that can be modified are access lists as augmented by this model. The MUD file is limited to the serialization of only the following YANG schema:

- ietf-access-control-list [I-D.ietf-netmod-acl-model]
- ietf-mud (this document)
- ietf-acldns (this document)

Extensions may be used to add additional schema. This is described further on.

To provide the widest possible deployment, publishers of MUD files SHOULD make use of the abstractions in this memo and avoid the use of IP addresses. A MUD controller SHOULD NOT automatically implement any MUD file that contains IP addresses, especially those that might have local significance. The addressing of one side of an access list is implicit, based on whether it is applied as to-device-policy or from-device-policy.

With the exceptions of "acl-name", "acl-type", "rule-name", and TCP and UDP source and destination port information, publishers of MUD files SHOULD limit the use of ACL model leaf nodes expressed to those found in this specification. Absent any extensions, MUD files are assumed to implement only the following ACL model features:

- any-acl, mud-acl, icmp-acl, ipv6-acl, tcp-acl, any-acl, udp-acl, ipv4-acl, and ipv6-acl

Furthermore, only "accept" or "drop" actions SHOULD be included. A MUD controller MAY choose to interpret "reject" as "drop." A MUD controller SHOULD ignore all other actions.

In fact, MUD controllers MAY ignore any particular component of a description or MAY ignore the description in its entirety, and SHOULD carefully inspect all MUD descriptions. Publishers of MUD files MUST NOT include other nodes except as described in Section 3.6. See that section for more information.

2.1. The IETF-MUD YANG Module

This module is structured into three parts:
The first container "mud" holds information that is relevant to retrieval and validity of the MUD file itself, as well as policy intended to and from the Thing.

The second component augments the matching container of the ACL model to add several nodes that are relevant to the MUD URL, or otherwise abstracted for use within a local environment.

The third component augments the tcp-acl container of the ACL model to add the ability to match on the direction of initiation of a TCP connection.

A valid MUD file will contain two root objects, a "mud" container and an "access-lists" container. Extensions may add additional root objects as required. As a reminder, when parsing access-lists, elements within a "match" block are logically ANDed. In general, a single abstraction in a match statement should be used. For instance, it makes little sense to match both "my-controller" and "controller" with an argument, since they are highly unlikely to be the same value.

A simplified graphical representation of the data models is used in this document. The meaning of the symbols in these diagrams is explained in [I-D.ietf-netmod-rfc6087bis].
module: ietf-mud
  +--rw mud
     +--rw mud-url               inet:uri
     +--rw last-update           yang:date-and-time
     +--rw cache-validity?       uint8
     +--rw is-supported          boolean
     +--rw systeminfo?           inet:uri
     +--rw extensions*           string
     +--rw from-device-policy
        +--rw access-lists
        |   +--rw access-list* [acl-name acl-type]
        |      +--rw acl-name -> /acl:access-lists/acl/acl-name
        |      +--rw acl-type    identityref
        +--rw to-device-policy
        +--rw access-lists
        |   +--rw access-list* [acl-name acl-type]
        |      +--rw acl-name -> /acl:access-lists/acl/acl-name
        |      +--rw acl-type    identityref
        augment /acl:access-lists/acl:acl/acl:aces/
        acl:ace/acl:matches:
           +--rw mud-acl
           +--rw manufacturer?        inet:host
           +--rw same-manufacturer?   empty
           +--rw model?               inet:uri
           +--rw local-networks?      empty
           +--rw controller?          inet:uri
           +--rw my-controller?       empty
           augment /acl:access-lists/acl:acl/acl:aces/
           acl:ace/acl:matches/acl:tcp-acl:
           +--rw direction-initiated? direction

3. Data Node Definitions

Note that in this section, when we use the term "match" we are referring to the ACL model "matches" node, and thus returns positive such that an action should be applied.

The following nodes are defined.

3.1. to-device-policy and from-device-policy containers

[I-D.ietf-netmod-acl-model] describes access-lists but does not attempt to indicate where they are applied as that is handled elsewhere in a configuration. However, in this case, a MUD file must be explicit in describing the communication pattern of a Thing, and that includes indicating what is to be permitted or denied in either direction of communication. Hence each of these containers indicate
the appropriate direction of a flow in association with a particular
Thing. They contain references to specific access-lists.

3.2. last-update

This is a date-and-time value of when the MUD file was generated.
This is akin to a version number. Its form is taken from [RFC6991]
which, for those keeping score, in turn was taken from Section 5.6 of
[RFC3339], which was taken from [ISO.8601.1988].

3.3. cache-validity

This uint8 is the period of time in hours that a network management
station MUST wait since its last retrieval before checking for an
update. It is RECOMMENDED that this value be no less than 24 and
MUST NOT be more than 168 for any Thing that is supported. This
period SHOULD be no shorter than any period determined through HTTP
caching directives (e.g., "cache-control" or "Expires"). N.B.,
expiring of this timer does not require the MUD controller to discard
the MUD file, nor terminate access to a Thing. See Section 15 for
more information.

3.4. is-supported

This boolean is an indication from the manufacturer to the network
administrator as to whether or not the Thing is supported. In this
context a Thing is said to NOT be supported if the manufacturer
intends never to issue an update to the Thing or never update the MUD
file. A MUD controller MAY still periodically check for updates.

3.5. systeminfo

This is a URL that points to a description of the Thing to be
connected. The intent is for administrators to be able to see a
localized name associated with the Thing. The referenced URL SHOULD
be a localized display string, and MAY be in either HTML or a raw
UTF-8 text file. It SHOULD NOT exceed 60 characters worth of display
space (that is- what the administrator actually sees), but it MAY
contain links to other documents (presumably product documentation).

3.6. extensions

This optional leaf-list names MUD extensions that are used in the MUD
file. Note that NO MUD extensions may be used in a MUD file prior to
the extensions being declared. Implementations MUST ignore any node
in this file that they do not understand.
Note that extensions can either extend the MUD file as described in the previous paragraph, or they might reference other work. An extension example can be found in Appendix C.

3.7. manufacturer

This node consists of a hostname that would be matched against the authority component of another Thing’s MUD URL. In its simplest form "manufacturer" and "same-manufacturer" may be implemented as access-lists. In more complex forms, additional network capabilities may be used. For example, if one saw the line "manufacturer": "flobbidy.example.com", then all Things that registered with a MUD URL that contained flobbity.example.com in its authority section would match.

3.8. same-manufacturer

This is an equivalent for when the manufacturer element is used to indicate the authority that is found in another Thing’s MUD URL matches that of the authority found in this Thing’s MUD URL. For example, if the Thing’s MUD URL were https://bl.example.com/.well-known/mud/v1/ThingV1, then all devices that had MUD URL with an authority section of bl.example.com would match.

3.9. model

This string matches the entire MUD URL, thus covering the model that is unique within the context of the authority. It may contain not only model information, but versioning information as well, and any other information that the manufacturer wishes to add. The intended use is for devices of this precise class to match, to permit or deny communication between one another.

3.10. local-networks

This null-valued node expands to include local networks. Its default expansion is that packets must not traverse toward a default route that is received from the router. However, administrators may expand the expression as is appropriate in their deployments.

3.11. controller

This URI specifies a value that a controller will register with the mud controller. The node then is expanded to the set of hosts that are so registered. This node may also be a URN. In this case, the URN describes a well known service, such as DNS or NTP.
Great care should be used when invoking the controller class. For one thing, it requires some understanding by the administrator as to when it is appropriate. Classes that are standardized may make it possible to easily name devices that support standard functions. For instance, the MUD controller could have some knowledge of which DNS servers should be used for any particular group of Things. Non-standard classes will likely require some sort of administrator interaction. Pre-registration in such classes by controllers with the MUD server is encouraged. The mechanism to do that is beyond the scope of this work.

Controller URIs MAY take the form of a URL (e.g. "http[s]://"). However, MUD controllers MUST NOT resolve and retrieve such files, and it is RECOMMENDED that there be no such file at this time, as their form and function may be defined at a point in the future. For now, URLs should serve simply as class names and be populated by the local deployment administrator.

3.12. my-controller

This null-valued node signals to the MUD controller to use whatever mapping it has for this MUD URL to a particular group of hosts. This may require prompting the administrator for class members. Future work should seek to automate membership management.

3.13. direction-initiated

When applied this matches packets when the flow was initiated in the corresponding direction. [RFC6092] specifies IPv6 guidance best practices. While that document is scoped specifically to IPv6, its contents are applicable for IPv4 as well. When this flag is set, and the system has no reason to believe a flow has been initiated it MUST drop the packet. This node may be implemented in its simplest form by looking at naked SYN bits, but may also be implemented through more stateful mechanisms.

4. Processing of the MUD file

To keep things relatively simple in addition to whatever definitions exist, we also apply two additional default behaviors:

- Anything not explicitly permitted is denied.
- Local DNS and NTP are, by default, permitted to and from the Thing.

An explicit description of the defaults can be found in Appendix B.
5. What does a MUD URL look like?

To begin with, MUD takes full advantage of both the https: scheme and the use of .well-known. HTTPS is important in this case because a man in the middle attack could otherwise harm the operation of a class of Things. .well-known is used because we wish to add additional structure to the URL, and want to leave open for future versions both the means by which the URL is processed and the format of the MUD file retrieved (there have already been some discussions along these lines). The URL appears as follows:

```
mud-url   = "https://" authority "/.well-known/mud/" mud-rev
   
"/" modelinfo ( "?" extras )

; authority is from RFC3986
mud-rev   = "v1"
modelinfo = segment ; from RFC3986
extras    = query ; from RFC3986
```

mud-rev signifies the version of the manufacturer usage description file. This memo specifies "v1" of that file. Later versions may permit additional schemas or modify the format. In order to provide for the broadest compatibility for the various transmission mechanisms, the length of the URL for v1 MUST NOT exceed 255 octets.

Taken together with the mud-url, "modelinfo" represents a Thing model as the manufacturer wishes to represent it. It could be a brand name or something more specific. It also may provide a means to indicate what version the product is. Specifically if it has been updated in the field, this is the place where evidence of that update would appear. The field should be changed when the intended communication patterns of a Thing change. While from a controller standpoint, only comparison and matching operations are safe, it is envisioned that updates will require some administrative review. Processing of this URL occurs as specified in [RFC2818] and [RFC3986].

"extras" is intended for use by the MUD controller to provide additional information such as posture about the Thing to the MUD file server. This field MUST NOT be configured on the Thing itself by a manufacturer - that is what "modelinfo" is for. It is left as future work to define the full semantics of this field.

6. The MUD YANG Model

```
<CODE BEGINS>file "ietf-mud@2017-10-07.yang"
module ietf-mud {
  yang-version 1.1;
  namespace "urn:ietf:params:xml:ns:yang:ietf-mud";
```
prefix ietf-mud;

import ietf-access-control-list {
    prefix acl;
}
import ietf-yang-types {
    prefix yang;
}
import ietf-inet-types {
    prefix inet;
}

organization "IETF OPSAWG (Ops Area) Working Group";
contact
  *WG Web: http://tools.ietf.org/wg/opsawg/
  WG List: opsawg@ietf.org
  Author: Eliot Lear
    lear@cisco.com
  Author: Ralph Droms
    rdroms@gmail.com
  Author: Dan Romascanu
    dromasca@gmail.com
"

description "This YANG module defines a component that augments the
IETF description of an access list. This specific module
focuses on additional filters that include local, model,
and same-manufacturer.

This module is intended to be serialized via JSON and stored
as a file, as described in RFC XXXX [RFC Editor to fill in with
this document #].

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Redistribution and use in source and binary forms, with or
without modification, is permitted pursuant to, and subject
to the license terms contained in, the Simplified BSD
License set forth in Section 4.c of the IETF Trust’s Legal
Provisions Relating to IETF Documents
(http://trustee.ietf.org/license-info).
This version of this YANG module is part of RFC XXXX; see
the RFC itself for full legal notices.";

revision 2017-10-07 {
    description
typedef direction {
  type enumeration {
    enum "to-device" {
      description
      "packets or flows destined to the target Thing";
    }
    enum "from-device" {
      description
      "packets or flows destined from the target Thing";
    }
  }
  description
  "Which way are we talking about?";
}

container mud {
  presence "Enabled for this particular MUD URL";
  description
  "MUD related information, as specified by RFC-XXXX [RFC Editor to fill in].";
  uses mud-grouping;
}

grouping mud-grouping {
  description
  "Information about when support end(ed), and when to refresh";
  leaf mud-url {
    type inet:uri;
    mandatory true;
    description
    "This is the MUD URL associated with the entry found in a MUD file.";
  }
  leaf last-update {
    type yang:date-and-time;
    mandatory true;
    description
    "This is intended to be when the current MUD file was generated. MUD Controllers SHOULD NOT check
for updates between this time plus cache validity;

leaf cache-validity {
  type uint8 {
    range "1..168";
  }
  units "hours";
  default "48";
  description
    "The information retrieved from the MUD server is
    valid for these many hours, after which it should
    be refreshed. N.B. MUD controller implementations
    need not discard MUD files beyond this period.";
}

leaf is-supported {
  type boolean;
  mandatory true;
  description
    "This boolean indicates whether or not the Thing is
    currently supported by the manufacturer.";
}

leaf systeminfo {
  type inet:uri;
  description
    "A URL to a description of this Thing. This
    should be a brief localized description. The
    reference text should be no more than octets.
    systeminfo may be displayed to the user to
determine whether to allow the Thing on the
    network.";
}

leaf-list extensions {
  type string {
    length "1..40";
  }
  description
    "A list of extension names that are used in this MUD
    file. Each name is registered with the IANA and
described in an RFC.";
}

container from-device-policy {
  description
    "The policies that should be enforced on traffic
    coming from the device. These policies are not
    necessarily intended to be enforced at a single
    point, but may be rendered by the controller to any
    relevant enforcement points in the network or
    elsewhere.";
}
uses access-lists;
}
}

container to-device-policy {

description
"The policies that should be enforced on traffic
going to the device. These policies are not
necessarily intended to be enforced at a single
point, but may be rendered by the controller to any
relevant enforcement points in the network or
elsewhere."
uses access-lists;
}

}

grouping access-lists {

description
"A grouping for access lists in the context of device
policy."
}

container access-lists {

description
"The access lists that should be applied to traffic
to or from the device."

list access-list {

description
"Each entry on this list refers to an ACL that
should be present in the overall access list
data model. Each ACL is identified by name and
type."

leaf acl-name {

type leafref {

path "/acl:access-lists/acl:acl/acl:acl-name";
}

description
"The name of the ACL for this entry."
}

leaf acl-type {

type identityref {

base acl:acl-base;
}

description
"The type of the ACL for this entry. The name is
scoped ONLY to the MUD file, and may not be unique
in any other circumstance.";
}
}
}
augment "/acl:access-lists/acl:acl/acl:aces/acl:ace/acl:matches" {
    description
    "adding abstractions to avoid need of IP addresses";
    container mud-acl {
        description
        "MUD-specific matches.";
        leaf manufacturer {
            type inet:host;
            description
            "A domain that is intended to match the authority
            section of the MUD URL. This node is used to specify
            one or more manufacturers a device should
            be authorized to access.";
        }
        leaf same-manufacturer {
            type empty;
            description
            "This node matches the authority section of the MUD URL
            of a Thing. It is intended to grant access to all
            devices with the same authority section.";
        }
        leaf model {
            type inet:uri;
            description
            "Devices of the specified model type will match if
            they have an identical MUD URL.";
        }
        leaf local-networks {
            type empty;
            description
            "IP addresses will match this node if they are
            considered local addresses. A local address may be
            a list of locally defined prefixes and masks
            that indicate a particular administrative scope.";
        }
        leaf controller {
            type inet:uri;
            description
            "This node names a class that has associated with it
            zero or more IP addresses to match against. These
            may be scoped to a manufacturer or via a standard
            URN.";
        }
        leaf my-controller {
            type empty;
            description
            "This node matches one or more network elements that
            have been configured to be the controller for this
7. The Domain Name Extension to the ACL Model

This module specifies an extension to IETF-ACL model such that domain names may be referenced by augmenting the "matches" node. Different implementations may deploy differing methods to maintain the mapping between IP address and domain name, if indeed any are needed. However, the intent is that resources that are referred to using a name should be authorized (or not) within an access list.

The structure of the change is as follows:

module: ietf-acldns
  augment /acl:access-lists/acl:ac/acl:aces/acl:ace/
    acl:matches/acl:ipv4-acl:
    +--rw src-dnsname? inet:host
    +--rw dst-dnsname? inet:host
  augment /acl:access-lists/acl:ac/acl:aces/acl:ace/
    acl:matches/acl:ipv6-acl:
    +--rw src-dnsname? inet:host
    +--rw dst-dnsname? inet:host

The choice of these particular points in the access-list model is based on the assumption that we are in some way referring to IP-related resources, as that is what the DNS returns. A domain name in our context is defined in [RFC6991]. The augmentations are
replicated across IPv4 and IPv6 to allow MUD file authors the ability to control the IP version that the Thing may utilize.

The following node are defined.

7.1. source-dnsname

The argument corresponds to a domain name of a source as specified by inet:host. A number of means may be used to resolve hosts. What is important is that such resolutions be consistent with ACLs required by Things to properly operate.

7.2. destination-dnsname

The argument corresponds to a domain name of a destination as specified by inet:host See the previous section relating to resolution.

7.3. The ietf-acldns Model

```xml
<CODE BEGINS>file "ietf-acldns@2017-10-07.yang"
module ietf-acldns {
  yang-version 1.1;
  namespace "urn:ietf:params:xml:ns:yang:ietf-acldns";
  prefix "ietf-acldns";

  import ietf-access-control-list {
    prefix "acl";
  }

  import ietf-inet-types {
    prefix "inet";
  }

  organization
    "IETF OPSAWG (Ops Area) Working Group";

  contact
    "WG Web: http://tools.ietf.org/wg/opsawg/
    WG List: opsawg@ietf.org
    Author: Eliot Lear
    lear@cisco.com
    Author: Ralph Droms
    rdroms@gmail.com
    Author: Dan Romascanu
    dromasca@gmail.com
    ";
</CODE ENDS>
This YANG module defines a component that augments the IETF description of an access list to allow dns names as matching criteria.

revision "2017-10-07" {
  description "Base version of dnsname extension of ACL model";
  reference "RFC XXXX: Manufacturer Usage Description Specification";
}

grouping dns-matches {
  description "Domain names for matching.";

  leaf src-dnsname {
    type inet:host;
    description "domain name to be matched against";
  }

  leaf dst-dnsname {
    type inet:host;
    description "domain name to be matched against";
  }
}

augment "/acl:access-lists/acl:acl/acl:aces/acl:ace/" +
  "acl:matches/acl:ipv4-acl" {
  description "Adding domain names to matching";
  uses dns-matches;
}

augment "/acl:access-lists/acl:acl/" +
  "acl:aces/acl:ace/" +
  "acl:matches/acl:ipv6-acl" {
  description "Adding domain names to matching";
  uses dns-matches;
}

<CODE ENDS>

8. MUD File Example

This example contains two access lists that are intended to provide outbound access to a cloud service on TCP port 443.

{
"ietf-mud:mud": {
  "last-update": "2017-10-07T12:16:24+02:00",
  "cache-validity": 48,
  "is-supported": true,
  "systeminfo": "https://bms.example.com/descriptions/lightbulb2000",
  "from-device-policy": {
    "access-lists": {
      "access-list": [
        {
          "acl-name": "mud-14377-v6fr",
          "acl-type": "ietf-access-control-list:ipv6-acl"
        }
      ]
    },
    "to-device-policy": {
      "access-lists": {
        "access-list": [
          {
            "acl-name": "mud-14377-v6to",
            "acl-type": "ietf-access-control-list:ipv6-acl"
          }
        ]
      }
    }
  }
},
"ietf-access-control-list:access-lists": {
  "acl": [
    {
      "acl-name": "mud-14377-v6to",
      "acl-type": "ipv6-acl",
      "access-list-entries": {
        "ace": [
          {
            "rule-name": "cl0-todev",
            "matches": {
              "ipv6-acl": {
                "ietf-acldns:src-dnsname": "service.bms.example.com",
                "protocol": 6,
                "source-port-range": {
                  "lower-port": 443,
                  "upper-port": 443
                }
              }
            }
          }
        ]
      }
    }
  ]
}
In this example, two policies are declared, one from the Thing and the other to the Thing. Each policy names an access list that applies to the Thing, and one that applies from. Within each access list, access is permitted to packets flowing to or from the Thing.
that can be mapped to the domain name of "service.bms.example.com".
For each access list, the enforcement point should expect that the
Thing initiated the connection.

9. The MUD URL DHCP Option

The IPv4 MUD URL client option has the following format:

```
+------+-----+-------------------------------
| code | len | MUD URL                        |
+------+-----+-------------------------------
```

Code OPTION_MUD_URL_V4 (161) is assigned by IANA. len is a single
octet that indicates the length of the URL in octets. MUD URL is a
URL. MUD URLs MUST NOT exceed 255 octets.

The IPv6 MUD URL client option has the following format:

```
0                   1                   2                   3
0 1 2 3 4 5 6 7 8 9 0 1 2 3 4 5 6 7 8 9 0 1
+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+
|         OPTION_MUD_URL_V6     |        option-length          |
+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+
|                            MUD URL                            |
|                              ...                              |
+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+-+
```

OPTION_MUD_URL_V6 (112; assigned by IANA).

option-length contains the length of the URL in octets.

The intent of this option is to provide both a new Thing classifier
to the network as well as some recommended configuration to the
routers that implement policy. However, it is entirely the purview
of the network system as managed by the network administrator to
decide what to do with this information. The key function of this
option is simply to identify the type of Thing to the network in a
structured way such that the policy can be easily found with existing
tools.

9.1. Client Behavior

A DHCPv4 client MAY emit a DHCPv4 option and a DHCPv6 client MAY emit
DHCPv6 option. These options are singletons, as specified in
[RFC7227]. Because clients are intended to have at most one MUD URL
associated with them, they may emit at most one MUD URL option via
DHCPv4 and one MUD URL option via DHCPv6. In the case where both v4 and v6 DHCP options are emitted, the same URL MUST be used.

Clients SHOULD log or otherwise report improper acknowledgments from servers, but they MUST NOT modify their MUD URL configuration based on a server’s response. The server’s response is only an acknowledgment that the server has processed the option, and promises no specific network behavior to the client. In particular, it may not be possible for the server to retrieve the file associated with the MUD URL, or the local network administration may not wish to use the usage description. Neither of these situations should be considered in any way exceptional.

9.2. Server Behavior

A DHCP server may ignore these options or take action based on receipt of these options. If a server successfully parses the option and the URL, it MUST return the option with length field set to zero and a corresponding null URL field as an acknowledgment. Even in this circumstance, no specific network behavior is guaranteed. When a server consumes this option, it will either forward the URL and relevant client information (such as the gateway address or giaddr) to a network management system, or it will retrieve the usage description itself by resolving the URL.

DHCP servers may implement MUD functionality themselves or they may pass along appropriate information to a network management system or MUD controller. A DHCP server that does process the MUD URL MUST adhere to the process specified in [RFC2818] and [RFC5280] to validate the TLS certificate of the web server hosting the MUD file. Those servers will retrieve the file, process it, create and install the necessary configuration on the relevant network element. Servers SHOULD monitor the gateway for state changes on a given interface. A DHCP server that does not provide MUD functionality and has forwarded a MUD URL to a MUD controller MUST notify the MUD controller of any corresponding change to the DHCP state of the client (such as expiration or explicit release of a network address lease).

9.3. Relay Requirements

There are no additional requirements for relays.

10. The Manufacturer Usage Description (MUD) URL X.509 Extension

This section defines an X.509 non-critical certificate extension that contains a single Uniform Resource Locator (URL) that points to an on-line Manufacturer Usage Description concerning the certificate.
subject. URI must be represented as described in Section 7.4 of [RFC5280].

Any Internationalized Resource Identifiers (IRIs) MUST be mapped to URIs as specified in Section 3.1 of [RFC3987] before they are placed in the certificate extension.

The semantics of the URL are defined Section 5 of this document.

The choice of id-pe is based on guidance found in Section 4.2.2 of [RFC5280]:

These extensions may be used to direct applications to on-line information about the issuer or the subject.

The MUD URL is precisely that: online information about the particular subject.

The new extension is identified as follows:
While this extension can appear in either an 802.AR manufacturer certificate (IDevID) or deployment certificate (LDevID), of course it is not guaranteed in either, nor is it guaranteed to be carried over. It is RECOMMENDED that MUD controller implementations maintain a table that maps a Thing to its MUD URL based on IDevIDs.

11. The Manufacturer Usage Description LLDP extension

The IEEE802.1AB Link Layer Discovery Protocol (LLDP) is a one hop vendor-neutral link layer protocol used by end hosts network Things for advertising their identity, capabilities, and neighbors on an IEEE 802 local area network. Its Type-Length-Value (TLV) design allows for ‘vendor-specific’ extensions to be defined. IANA has a registered IEEE 802 organizationally unique identifier (OUI) defined
as documented in [RFC7042]. The MUD LLDP extension uses a subtype defined in this document to carry the MUD URL.

The LLDP vendor specific frame has the following format:

```
+--------+--------+----------+---------+--------------
| TLV Type|  len   |   OUI    | subtype  | MUD URL      |
|  =127   |        |= 00 00 5E|  = 1    |              |
| (7 bits)|(9 bits)|(3 octets)|(1 octet) | (1-255 octets) |
+--------+--------+----------+---------+--------------
```

where:

- TLV Type = 127 indicates a vendor-specific TLV
- len - indicates the TLV string length
- OUI = 00 00 5E is the organizationally unique identifier of IANA
- subtype = 1 (to be assigned by IANA for the MUD URL)
- MUD URL - the length MUST NOT exceed 255 octets

The intent of this extension is to provide both a new Thing classifier to the network as well as some recommended configuration to the routers that implement policy. However, it is entirely the purview of the network system as managed by the network administrator to decide what to do with this information. The key function of this extension is simply to identify the type of Thing to the network in a structured way such that the policy can be easily found with existing toolsets.

Hosts, routers, or other network elements that implement this option are intended to have at most one MUD URL associated with them, so they may transmit at most one MUD URL value.

Hosts, routers, or other network elements that implement this option may ignore these options or take action based on receipt of these options. For example they may fill in information in the respective extensions of the LLDP Management Information Base (LLDP MIB). LLDP operates in a one-way direction. LLDPDU's are not exchanged as information requests by one Thing and response sent by another Thing. The other Things do not acknowledge LLDP information received from a Thing. No specific network behavior is guaranteed. When a Thing consumes this extension, it may either forward the URL and relevant remote Thing information to a MUD controller, or it will retrieve the
usage description by resolving the URL in accordance with normal HTTP semantics.

12. Creating and Processing of Signed MUD Files

Because MUD files contain information that may be used to configure network access lists, they are sensitive. To insure that they have not been tampered with, it is important that they be signed. We make use of DER-encoded Cryptographic Message Syntax (CMS) [RFC5652] for this purpose.

12.1. Creating a MUD file signature

A MUD file MUST be signed using CMS as an opaque binary object. In order to make successful verification more likely, intermediate certificates SHOULD be included. The signature is stored at the same location as the MUD URL but with the suffix of ".p7s". Signatures are transferred using content-type "application/pkcs7-signature".

For example:

% openssl cms -sign -signer mancertfile -inkey mankey \
    -in mudfile -binary -outform DER - \n    -certfile intermediatecert -out mudfile.p7s

Note: A MUD file may need to be re-signed if the signature expires.

12.2. Verifying a MUD file signature

Prior to retrieving a MUD file the MUD controller SHOULD retrieve the MUD signature file using the MUD URL with a suffix of ".p7s". For example, if the MUD URL is "https://example.com/.well-known/v1/modela", the MUD signature URL will be "https://example.com/.well-known/v1/modela.p7s".

Upon retrieving a MUD file, a MUD controller MUST validate the signature of the file before continuing with further processing. A MUD controller MUST cease processing of that file it cannot validate the chain of trust to a known trust anchor until an administrator has given approval.

The purpose of the signature on the file is to assign accountability to an entity, whose reputation can be used to guide administrators on whether or not to accept a given MUD file. It is already common place to check web reputation on the location of a server on which a file resides. While it is likely that the manufacturer will be the signer of the file, this is not strictly necessary, and may not be desirable. For one thing, in some environments, integrators may
install their own certificates. For another, what is more important is the accountability of the recommendation, and not the cryptographic relationship between the device and the file.

An example:

```
% openssl cms -verify -in mudfile.p7s -inform DER -content mudfile
```

Note the additional step of verifying the common trust root.

13. Extensibility

One of our design goals is to see that MUD files are able to be understood by as broad a cross-section of systems as is possible. Coupled with the fact that we have also chosen to leverage existing mechanisms, we are left with no ability to negotiate extensions and a limited desire for those extensions in any event. A such, a two-tier extensibility framework is employed, as follows:

1. At a coarse grain, a protocol version is included in a MUD URL. This memo specifies MUD version 1. Any and all changes are entertained when this version is bumped. Transition approaches between versions would be a matter for discussion in future versions.

2. At a finer grain, only extensions that would not incur additional risk to the Thing are permitted. Specifically, adding nodes to the mud container is permitted with the understanding that such additions will be ignored by unaware implementations. Any such extensions SHALL be standardized through the IETF process, and MUST be named in the "extensions" list. MUD controllers MUST ignore YANG nodes they do not understand and SHOULD create an exception to be resolved by an administrator, so as to avoid any policy inconsistencies.

14. Deployment Considerations

Because MUD consists of a number of architectural building blocks, it is possible to assemble different deployment scenarios. One key aspect is where to place policy enforcement. In order to protect the Thing from other Things within a local deployment, policy can be enforced on the nearest switch or access point. In order to limit unwanted traffic within a network, it may also be advisable to enforce policy as close to the Internet as possible. In some circumstances, policy enforcement may not be available at the closest hop. At that point, the risk of so-called east-west infection is increased to the number of Things that are able to communicate without protection.
A caution about some of the classes: admission of a Thing into the "manufacturer" and "same-manufacturer" class may have impact on access of other Things. Put another way, the admission may grow the access-list on switches connected to other Things, depending on how access is managed. Some care should be given on managing that access-list growth. Alternative methods such as additional network segmentation can be used to keep that growth within reason.

Because as of this writing MUD is a new concept, one can expect a great many devices to not have implemented it. It remains a local deployment decision as to whether a device that is first connected should be allowed broad or limited access. Furthermore, as mentioned in the introduction, a deployment may choose to ignore a MUD policy in its entirety, but simply taken into account the MUD URL as a classifier to be used as part of a local policy decision.

15. Security Considerations

Based on how a MUD URL is emitted, a Thing may be able to lie about what it is, thus gaining additional network access. There are several means to limit risk in this case. The most obvious is to only believe Things that make use of certificate-based authentication such as IEEE 802.1AR certificates. When those certificates are not present, Things claiming to be of a certain manufacturer SHOULD NOT be included in that manufacturer grouping without additional validation of some form. This will occur when it makes use of primitives such as "manufacturer" for the purpose of accessing Things of a particular type. Similarly, network management systems may be able to fingerprint the Thing. In such cases, the MUD URL can act as a classifier that can be proven or disproven. Fingerprinting may have other advantages as well: when 802.1AR certificates are used, because they themselves cannot change, fingerprinting offers the opportunity to add artifacts to the MUD URL. The meaning of such artifacts is left as future work.

Network management systems SHOULD NOT accept a usage description for a Thing with the same MAC address that has indicated a change of authority without some additional validation (such as review by a network administrator). New Things that present some form of unauthenticated MUD URL SHOULD be validated by some external means when they would be otherwise be given increased network access.

It may be possible for a rogue manufacturer to inappropriately exercise the MUD file parser, in order to exploit a vulnerability. There are three recommended approaches to address this threat. The first is to validate the signature of the MUD file. The second is to have a system do a primary scan of the file to ensure that it is both parseable and believable at some level. MUD files will likely be
relatively small, to start with. The number of ACEs used by any
given Thing should be relatively small as well. It may also be
useful to limit retrieval of MUD URLs to only those sites that are
known to have decent web or domain reputations.

Use of a URL necessitates the use of domain names. If a domain name
changes ownership, the new owner of that domain may be able to
provide MUD files that MUD controllers would consider valid. There
are a few approaches that can mitigate this attack. First, MUD
controllers SHOULD cache certificates used by the MUD file server.
When a new certificate is retrieved for whatever reason, the MUD
controller should check to see if ownership of the domain has
changed. A fair programmatic approximation of this is when the name
servers for the domain have changed. If the actual MUD file has
changed, the controller MAY check the WHOIS database to see if
registration ownership of a domain has changed. If a change has
occurred, or if for some reason it is not possible to determine
whether ownership has changed, further review may be warranted.
Note, this remediation does not take into account the case of a Thing
that was produced long ago and only recently fielded, or the case
where a new MUD controller has been installed.

It may not be possible for a MUD controller to retrieve a MUD file at
any given time. Should a MUD controller fail to retrieve a MUD file,
it SHOULD consider the existing one safe to use, at least for a time.
After some period, it SHOULD log that it has been unable to retrieve
the file. There may be very good reasons for such failures,
including the possibility that the MUD controller is in an off-line
environment, the local Internet connection has failed, or the remote
Internet connection has failed. It is also possible that an attacker
is attempting to prevent onboarding of a device. It is a local
deployment decision as to whether or not devices may be onboarded in
the face of such failures.

The release of a MUD URL by a Thing reveals what the Thing is, and
provides an attacker with guidance on what vulnerabilities may be
present.

While the MUD URL itself is not intended to be unique to a specific
Thing, the release of the URL may aid an observer in identifying
individuals when combined with other information. This is a privacy
consideration.

In addressing both of these concerns, implementors should take into
account what other information they are advertising through
mechanisms such as mDNS [RFC6872], how a Thing might otherwise be
identified, perhaps through how it behaves when it is connected to
the network, whether a Thing is intended to be used by individuals or
carry personal identifying information, and then apply appropriate data minimization techniques. One approach is to make use of TEAP [RFC7170] as the means to share information with authorized components in the network. Network elements may also assist in limiting access to the MUD URL through the use of mechanisms such as DHCPv6-Shield [RFC7610].

Please note that the security considerations mentioned in Section 4.7 of [I-D.ietf-netmod-rfc6087bis] are not applicable in this case because the YANG serialization is not intended to be accessed via NETCONF. However, for those who try to instantiate this model in a network element via NETCONF, all objects in each model in this draft exhibit similar security characteristics as [I-D.ietf-netmod-acl-model]. The basic purpose of MUD is to configure access, and so by its very nature can be disruptive if used by unauthorized parties.

16. IANA Considerations

16.1. YANG Module Registrations

The following YANG modules are requested to be registered in the "IANA Module Names" registry:

The ietf-mud module:
- Name: ietf-mud
- Prefix: ietf-mud
- Reference: This memo

The ietf-acldns module:
- Name: ietf-acldns
- Prefix: ietf-acldns
- Reference: This memo
16.2. DHCPv4 and DHCPv6 Options

The IANA has allocated option 161 in the Dynamic Host Configuration Protocol (DHCP) and Bootstrap Protocol (BOOTP) Parameters registry for the MUD DHCPv4 option.

IANA is requested to allocated the DHCPv4 and v6 options as specified in Section 9.

16.3. PKIX Extensions

IANA is kindly requested to make the following assignments for:

- The MUDURLExtnModule-2016 ASN.1 module in the "SMI Security for PKIX Module Identifier" registry (1.3.6.1.5.5.7.0).
- The id-pe-mud-url object identifier from the "SMI Security for PKIX Certificate Extension" registry (1.3.6.1.5.5.7.1).

The use of these values is specified in Section 10.

16.4. Well Known URI Suffix

The IANA has allocated the URL suffix of "mud" as follows:

- URI Suffix: "mud"
- Specification documents: this document
- Related information: n/a

16.5. MIME Media-type Registration for MUD files

The following media-type is defined for transfer of MUD file:
16.6. LLDP IANA TLV Subtype Registry

IANA is requested to create a new registry for IANA Link Layer Discovery Protocol (LLDP) TLV subtype values. The recommended policy for this registry is Expert Review. The maximum number of entries in the registry is 256.

IANA is required to populate the initial registry with the value:

LLDP subtype value = 1 (All the other 255 values should be initially marked as ‘Unassigned’.)

Description = the Manufacturer Usage Description (MUD) Uniform Resource Locator (URL)

Reference = < this document >
16.7. The MUD Well Known Universal Resource Name (URNs)

The following parameter registry is requested to be added in accordance with [RFC3553]

Registry name: "urn:ietf:params:mud" is requested.
Specification: this document
Repository: this document
Index value: Encoded identically to a TCP/UDP port service name, as specified in Section 5.1 of [RFC6335]

The following entries should be added to the "urn:ietf:params:mud" name space:


16.8. Extensions Registry

The IANA is requested to establish a registry of extensions as follows:

Registry name: MUD extensions registry
Registry policy: Standards action
Standard reference: document
Extension name: UTF-8 encoded string, not to exceed 40 characters.

Each extension MUST follow the rules specified in this specification. As is usual, the IANA issues early allocations based in accordance with [RFC7120].

17. Acknowledgments

The authors would like to thank Einar Nilsen-Nygaard, who singlehandedly updated the model to match the updated ACL model, Bernie Volz, Tom Gindin, Brian Weis, Sandeep Kumar, Thorsten Dahm, John Bashinski, Steve Rich, Jim Bieda, Dan Wing, Joe Clarke, Henk Birkholz, Adam Montville, and Robert Sparks for their valuable advice and reviews. Russ Housley entirely rewrote Section 10 to be a complete module. Adrian Farrel provided the basis for privacy considerations text. Kent Watsen provided a thorough review of the architecture and the YANG model. The remaining errors in this work are entirely the responsibility of the authors.
18. References

18.1. Normative References

[I-D.ietf-netmod-acl-model]
Jethanandani, M., Huang, L., Agarwal, S., and D. Blair,
"Network Access Control List (ACL) YANG Data Model",

[IEEE8021AB]
Institute for Electrical and Electronics Engineers, "IEEE Standard for Local and Metropolitan Area Networks--Station and Media Access Control Connectivity Discovery", n.d..


18.2. Informative References


Appendix A. Changes from Earlier Versions

RFC Editor to remove this section prior to publication.

Draft -10 to -12:

These are based on WGLC comments:

- Correct examples based on ACL model changes.
- Change ordering nodes.
- Additional explanatory text around systeminfo.
- Change ordering in examples.

Lear, et al. Expires April 27, 2018
o Make it VERY VERY VERY VERY clear that these are recommendations, not mandates.

o DHCP -> NTP in some of the intro text.

o Remove masa-server

o "Things" to "network elements" in a few key places.

o Reference to JSON YANG RFC added.

Draft -10 to -11:

o Example corrections

o Typo

o Fix two lists.

o Addition of 'any-acl' and 'mud-acl' in the list of allowed features.

o Clarification of what should be in a MUD file.

Draft -09 to -10:

o AD input.

o Correct dates.

o Add compliance sentence as to which ACL module features are implemented.

Draft -08 to -09:

o Resolution of Security Area review, IoT directorate review, GenART review, YANG doctors review.

o change of YANG structure to address mandatory nodes.

o Terminology cleanup.

o specify out extra portion of MUD-URL.

o consistency changes.

o improved YANG descriptions.
o Remove extra revisions.

o Track ACL model changes.

o Additional cautions on use of ACL model; further clarifications on extensions.

Draft -07 to -08:

o a number of editorials corrected.

o definition of MUD file tweaked.

Draft -06 to -07:

o Examples updated.

o Additional clarification for direction-initiated.

o Additional implementation guidance given.

Draft -06 to -07:

o Update models to match new ACL model

o extract directionality from the ACL, introducing a new device container.

Draft -05 to -06:

o Make clear that this is a component architecture (Polk and Watson)

o Add order of operations (Watson)

o Add extensions leaf-list (Pritikin)

o Remove previous-mud-file (Watson)

o Modify text in last-update (Watson)

o Clarify local networks (Weis, Watson)

o Fix contact info (Watson)

o Terminology clarification (Weis)

o Advice on how to handle LDevIDs (Watson)
Appendix B. Default MUD nodes

What follows is the portion of a MUD file that permits DNS traffic to a controller that is registered with the URN "urn:ietf:params:mud:dns" and traffic NTP to a controller that is registered "urn:ietf:params:mud:ntp". This is considered the default behavior and the ACEs are in effect appended to whatever other "ace" entries that a MUD file contains. To block DNS or NTP one repeats the matching statement but replaces the "forwarding" action "accept" with "drop". Because ACEs are processed in the order they are received, the defaults would not be reached. A MUD controller might further decide to optimize to simply not include the defaults when they are overridden.

Four of "acl" list entries that implement default MUD nodes is listed below. Two are for IPv4 and two are for IPv6 (one in each direction for both versions of IP).

```json
"ietf-access-control-list:access-lists": {
  "acl": [
    {
      "acl-name": "mud-v4-default-to-device",
      "acl-type": "ipv4-acl",
      "aces": [
```
"ace": [  
  {  
    "rule-name": "ent0-todev",  
    "matches": {  
      "ietf-mud:mud-acl": {  
        "controller": "urn:ietf:params:mud:dns"  
      },  
      "ipv4-acl": {  
        "protocol": 17,  
        "source-port-range": {  
          "lower-port": 53,  
          "upper-port": 53  
        }  
      }  
    },  
    "actions": {  
      "forwarding": "accept"  
    }  
  },  
  {  
    "rule-name": "ent1-todev",  
    "matches": {  
      "ietf-mud:mud-acl": {  
        "controller": "urn:ietf:params:mud:ntp"  
      },  
      "ipv4-acl": {  
        "protocol": 17,  
        "source-port-range": {  
          "lower-port": 123,  
          "upper-port": 123  
        }  
      }  
    },  
    "actions": {  
      "forwarding": "accept"  
    }  
  }  
],  
"acl-name": "mud-v4-default-from-device",  
"acl-type": "ipv4-acl",  
"aces": [  
  "ace": [  
    {  
      "rule-name": "ent0-frdev",  
      "matches": {  

"ietf-mud:mud-acl": {
    "controller": "urn:ietf:params:mud:dns"
},
"ipv4-acl": {
    "protocol": 17,
    "destination-port-range": {
        "lower-port": 53,
        "upper-port": 53
    }
},
"actions": {
    "forwarding": "accept"
},
{
    "rule-name": "ent1-frdev",
    "matches": {
        "ietf-mud:mud-acl": {
            "controller": "urn:ietf:params:mud:ntp"
        },
        "ipv4-acl": {
            "protocol": 17,
            "destination-port-range": {
                "lower-port": 123,
                "upper-port": 123
            }
        }
    },
    "actions": {
        "forwarding": "accept"
    }
}
}
{
    "acl-name": "mud-v6-default-to-device",
    "acl-type": "ipv6-acl",
    "access-list-entries": {
        "ace": {
            "rule-name": "ent0-todev",
            "matches": {
                "ietf-mud:mud-acl": {
                    "controller": "urn:ietf:params:mud:dns"
                },
                "ipv6-acl": {
"protocol": 17,
"source-port-range": {
   "lower-port": 53,
   "upper-port": 53
}
},
"actions": {
   "forwarding": "accept"
}
},
"rule-name": "ent1-todev",
"matches": {
   "ietf-mud:mud-acl": {
      "controller": "urn:ietf:params:mud:ntp"
   },
   "ipv6-acl": {
      "protocol": 17,
      "source-port-range": {
         "lower-port": 123,
         "upper-port": 123
      }
   }
},
"actions": {
   "forwarding": "accept"
}
]
},
"acl-name": "mud-v6-default-from-device",
"acl-type": "ipv6-acl",
"access-list-entries": {
   "ace": [
      {
         "rule-name": "ent0-frdev",
         "matches": {
            "ietf-mud:mud-acl": {
               "controller": "urn:ietf:params:mud:dns"
            },
            "ipv6-acl": {
               "protocol": 17,
               "destination-port-range": {
                  "lower-port": 53,
                  "upper-port": 53
               }
            }
         }
      }
   ]
}
"actions": {  
  "forwarding": "accept"
}
],
{
  "rule-name": "ent1-frdev",
  "matches": {
    "ietf-mud:mud-acl": {
      "controller": "urn:ietf:params:mud:ntp"
    },
    "ipv6-acl": {
      "protocol": 17,
      "destination-port-range": {
        "lower-port": 123,
        "upper-port": 123
      }
    }
  },
  "actions": {
    "forwarding": "accept"
  }
}
]
}
]

Appendix C. A Sample Extension: DETNET-indicator

In this sample extension we augment the core MUD model to indicate whether the device implements DETNET. If a device later attempts to make use of DETNET, an notification or exception might be generated. Note that this example is intended only for illustrative purposes.

Extension Name: "Example-Extension" (to be used in the extensions list)
Standard: this document (but do not register the example)

This extension augments the MUD model to include a single node, using the following sample module that has the following tree structure:
The model is defined as follows:

```
<CODE BEGINS>file "ietf-mud-detext-example@2016-09-07.yang"
module ietf-mud-detext-example {
  yang-version 1.1;
  namespace "urn:ietf:params:xml:ns:yang:ietf-mud-detext-example";
  prefix ietf-mud-detext-example;

  import ietf-mud {
    prefix ietf-mud;
  }

  organization
    "IETF OPSAWG (Ops Area) Working Group";

  contact
    "WG Web: http://tools.ietf.org/wg/opsawg/
     WG List: opsawg@ietf.org
     Author: Eliot Lear
        lear@cisco.com
     Author: Ralph Droms
        rdroms@gmail.com
     Author: Dan Romascanu
        dromasca@gmail.com
    
    description
      "Sample extension to a MUD module to indicate a need
      for DETNET support.";

  revision 2017-09-05 {
    description
      "Initial revision.";
    reference
      "RFC XXXX: Manufacturer Usage Description
         Specification";
  }

  augment "/ietf-mud:mud" {
    description
      "This adds a simple extension for a manufacturer
      to indicate whether DETNET is required by a
device.";
    leaf is-detnet-required {
```

```
Using the previous example, we now show how the extension would be expressed:

```json
{
    "ietf-mud:mud": {
        "mud-url": "https://bms.example.com/.well-known/mud/v1/lightbulb",
        "last-update": "2017-09-20T15:49:18+02:00",
        "cache-validity": 48,
        "is-supported": true,
        "systeminfo": "https://bms.example.com/descriptions/lightbulb",
        "extensions": [
            "ietf-mud-detext-example",
            "ietf-mud-detext-example:is-detnet-required": "false",
            "from-device-policy": {
                "access-lists": {
                    "access-list": [
                        {
                            "acl-name": "mud-54684-v6fr",
                            "acl-type": "ietf-access-control-list:ipv6-acl"
                        }
                    ]
                }
            },
            "to-device-policy": {
                "access-lists": {
                    "access-list": [
                        {
                            "acl-name": "mud-54684-v6to",
                            "acl-type": "ietf-access-control-list:ipv6-acl"
                        }
                    ]
                }
            }
        ],
        "ietf-access-control-list:access-lists": {
            "acl": {
                "acl-name": "mud-54684-v6to",
                "acl-type": "ietf-access-control-list:ipv6-acl"
            }
        }
    }
}
```
"acl-type": "ipv6-acl",
"access-list-entries": {
  "ace": [
    {
      "rule-name": "cl0-todev",
      "matches": {
        "ipv6-acl": {
          "ietf-acldns:src-dnsname": "service.bms.example.com",
          "protocol": 6,
          "source-port-range": {
            "lower-port": 443,
            "upper-port": 443
          }
        },
        "tcp-acl": {
          "ietf-mud:direction-initiated": "from-device"
        }
      },
      "actions": {
        "forwarding": "accept"
      }
    }
  ]
},
"acl-name": "mud-54684-v6fr",
"acl-type": "ipv6-acl",
"access-list-entries": {
  "ace": [
    {
      "rule-name": "cl0-frdev",
      "matches": {
        "ipv6-acl": {
          "ietf-acldns:dst-dnsname": "service.bms.example.com",
          "protocol": 6,
          "destination-port-range": {
            "lower-port": 443,
            "upper-port": 443
          }
        },
        "tcp-acl": {
          "ietf-mud:direction-initiated": "from-device"
        }
      },
      "actions": {
        "forwarding": "accept"
      }
    }
  ]
}
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Abstract

This memo specifies a component-based architecture for manufacturer usage descriptions (MUD). The goal of MUD is to provide a means for end devices to signal to the network what sort of access and network functionality they require to properly function. The initial focus is on access control. Later work can delve into other aspects.

This memo specifies two YANG modules, IPv4 and IPv6 DHCP options, an LLDP TLV, a URL, an X.509 certificate extension and a means to sign and verify the descriptions.

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1. Introduction

The Internet has largely been constructed for general purpose computers, those devices that may be used for a purpose that is specified by those who own the device. [RFC1984] presumed that an end device would be most capable of protecting itself. This made sense when the typical device was a workstation or a mainframe, and it continues to make sense for general purpose computing devices today, including laptops, smart phones, and tablets.

[RFC7452] discusses design patterns for, and poses questions about, smart objects. Let us then posit a group of objects that are specifically not intended to be used for general purpose computing tasks. These devices, which this memo refers to as Things, have a specific purpose. By definition, therefore, all other uses are not intended. If a small number of communication patterns follows from those small number of uses, the combination of these two statements can be restated as a manufacturer usage description (MUD) that can be applied at various points within a network. MUD primarily addresses
threats to the device rather than the device as a threat. In some circumstances, however, MUD may offer some protection in the latter case, depending on the MUD-URL is communicated, and how devices and their communications are authenticated.

We use the notion of "manufacturer" loosely in this context to refer to the entity or organization that will state how a device is intended to be used. For example, in the context of a lightbulb, this might indeed be the lightbulb manufacturer. In the context of a smarter device that has a built in Linux stack, it might be an integrator of that device. The key points are that the device itself is assumed to serve a limited purpose, and that there exists an organization in the supply chain of that device that will take responsibility for informing the network about that purpose.

The intent of MUD is to provide the following:

- Substantially reduce the threat surface on a device to those communications intended by the manufacturer.
- Provide a means to scale network policies to the ever-increasing number of types of devices in the network.
- Provide a means to address at least some vulnerabilities in a way that is faster than the time it might take to update systems. This will be particularly true for systems that are no longer supported.
- Keep the cost of implementation of such a system to the bare minimum.
- Provide a means of extensibility for manufacturers to express other device capabilities or requirements.

MUD consists of three architectural building blocks:

- A URL that can be used to locate a description;
- The description itself, including how it is interpreted, and;
- A means for local network management systems to retrieve the description.

MUD is most effective when the network is able to identify in some way the remote endpoints that Things will talk to.

In this specification we describe each of these building blocks and how they are intended to be used together. However, they may also be
used separately, independent of this specification, by local deployments for their own purposes.

1.1. What MUD Doesn’t Do

MUD is not intended to address network authorization of general purpose computers, as their manufacturers cannot envision a specific communication pattern to describe. In addition, even those devices that have a single or small number of uses might have very broad communication patterns. MUD on its own is not for them either.

Although MUD can provide network administrators with some additional protection when device vulnerabilities exist, it will never replace the need for manufacturers to patch vulnerabilities.

Finally, no matter what the manufacturer specifies in a MUD file, these are not directives, but suggestions. How they are instantiated locally will depend on many factors and will be ultimately up to the local network administrator, who must decide what is appropriate in a given circumstances.

1.2. A Simple Example

A light bulb is intended to light a room. It may be remotely controlled through the network, and it may make use of a rendezvous service of some form that an application on a smart phone. What we can say about that light bulb, then, is that all other network access is unwanted. It will not contact a news service, nor speak to the refrigerator, and it has no need of a printer or other devices. It has no social networking friends. Therefore, an access list applied to it that states that it will only connect to the single rendezvous service will not impede the light bulb in performing its function, while at the same time allowing the network to provide both it and other devices an additional layer of protection.

1.3. Terminology

MUD: manufacturer usage description.

MUD file: a file containing YANG-based JSON that describes a Thing and associated suggested specific network behavior.

MUD file server: a web server that hosts a MUD file.

MUD manager: the system that requests and receives the MUD file from the MUD server. After it has processed a MUD file, it may direct changes to relevant network elements.
MUD controller: a synonym that has been used in the past for MUD manager.

MUD URL: a URL that can be used by the MUD manager to receive the MUD file.

Thing: the device emitting a MUD URL.

Manufacturer: the entity that configures the Thing to emit the MUD URL and the one who asserts a recommendation in a MUD file. The manufacturer might not always be the entity that constructs a Thing. It could, for instance, be a systems integrator, or even a component provider.

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "NOT RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in BCP 14 [RFC2119] [RFC8174] when, and only when, they appear in all capitals, as shown here.

1.4. Determining Intended Use

The notion of intended use is in itself not new. Network administrators apply access lists every day to allow for only such use. This notion of white listing was well described by Chapman and Zwicky in [FW95]. Profiling systems that make use of heuristics to identify types of systems have existed for years as well.

A Thing could just as easily tell the network what sort of access it requires without going into what sort of system it is. This would, in effect, be the converse of [RFC7488]. In seeking a general solution, however, we assume that a device will implement functionality necessary to fulfill its limited purpose. This is basic economic constraint. Unless the network would refuse access to such a device, its developers would have no reason to provide the network any information. To date, such an assertion has held true.

1.5. Finding A Policy: The MUD URL

Our work begins with the device emitting a Universal Resource Locator (URL) [RFC3986]. This URL serves both to classify the device type and to provide a means to locate a policy file.

MUD URLs MUST use the HTTPS scheme [RFC7230].

In this memo three means are defined to emit the MUD URL, as follows:
o A DHCP option[RFC2131],[RFC3315] that the DHCP client uses to inform the DHCP server. The DHCP server may take further actions, such as act as the MUD manager or otherwise pass the MUD URL along to the MUD manager.

o An X.509 constraint. The IEEE has developed [IEEE8021AR] that provides a certificate-based approach to communicate device characteristics, which itself relies on [RFC5280]. The MUD URL extension is non-critical, as required by IEEE 802.1AR. Various means may be used to communicate that certificate, including Tunnel Extensible Authentication Protocol (TEAP) [RFC7170].

o Finally, a Link Layer Discovery Protocol (LLDP) frame is defined [IEEE8021AB].

It is possible that there may be other means for a MUD URL to be learned by a network. For instance, some devices may already be fielded or have very limited ability to communicate a MUD URL, and yet can be identified through some means, such as a serial number or a public key. In these cases, manufacturers may be able to map those identifiers to particular MUD URLs (or even the files themselves). Similarly, there may be alternative resolution mechanisms available for situations where Internet connectivity is limited or does not exist. Such mechanisms are not described in this memo, but are possible. Implementors are encouraged to allow for this sort of flexibility of how MUD URLs may be learned.

1.6. Processing of the MUD URL

MUD managers that are able to do so SHOULD retrieve MUD URLs and signature files as per [RFC7230], using the GET method [RFC7231]. They MUST validate the certificate using the rules in [RFC2818], Section 3.1.

Requests for MUD URLs SHOULD include an "Accept" header ([RFC7231], Section 5.3.2) containing "application/mud+json", an "Accept-Language" header field ([RFC7231], Section 5.3.5), and a "User-Agent" header ([RFC7231], Section 5.5.3).

MUD managers SHOULD automatically process 3xx response status codes.

If a MUD manager is not able to fetch a MUD URL, other means MAY be used to import MUD files and associated signature files. So long as the signature of the file can be validated, the file can be used. In such environments, controllers SHOULD warn administrators when cache-validity expiry is approaching so that they may check for new files.
It may not be possible for a MUD manager to retrieve a MUD file at any given time. Should a MUD manager fail to retrieve a MUD file, it SHOULD consider the existing one safe to use, at least for a time. After some period, it SHOULD log that it has been unable to retrieve the file. There may be very good reasons for such failures, including the possibility that the MUD manager is in an off-line environment, the local Internet connection has failed, or the remote Internet connection has failed. It is also possible that an attacker is attempting to interfere with the deployment of a device. It is a local decision as to how to handle such circumstances.

1.7. Types of Policies

When the MUD URL is resolved, the MUD manager retrieves a file that describes what sort of communications a device is designed to have. The manufacturer may specify either specific hosts for cloud based services or certain classes for access within an operational network. An example of a class might be "devices of a specified manufacturer type", where the manufacturer type itself is indicated simply by the authority component (e.g., the domain name) of the MUD URL. Another example might be to allow or disallow local access. Just like other policies, these may be combined. For example:

- Allow access to devices of the same manufacturer
- Allow access to and from controllers via Constrained Application Protocol (CoAP) [RFC7252]
- Allow access to local DNS/NTP
- Deny all other access

A printer might have a description that states:

- Allow access for port IPP or port LPD
- Allow local access for port HTTP
- Deny all other access

In this way anyone can print to the printer, but local access would be required for the management interface.

The files that are retrieved are intended to be closely aligned to existing network architectures so that they are easy to deploy. We make use of YANG [RFC7950] because it provides accurate and adequate models for use by network devices. JSON [RFC8259] is used as a
Serialization format for compactness and readability, relative to XML. Other formats may be chosen with later versions of MUD.

While the policy examples given here focus on access control, this is not intended to be the sole focus. By structuring the model described in this document with clear extension points, other descriptions could be included. One that often comes to mind is quality of service.

The YANG modules specified here are extensions of [I-D.ietf-netmod-acl-model]. The extensions to this model allow for a manufacturer to express classes of systems that a manufacturer would find necessary for the proper function of the device. Two modules are specified. The first module specifies a means for domain names to be used in ACLs so that devices that have their controllers in the cloud may be appropriately authorized with domain names, where the mapping of those names to addresses may rapidly change.

The other module abstracts away IP addresses into certain classes that are instantiated into actual IP addresses through local processing. Through these classes, manufacturers can specify how the device is designed to communicate, so that network elements can be configured by local systems that have local topological knowledge. That is, the deployment populates the classes that the manufacturer specifies. The abstractions below map to zero or more hosts, as follows:

Manufacturer: A device made by a particular manufacturer, as identified by the authority component of its MUD URL

same-manufacturer: Devices that have the same authority component of their MUD URL.

creator: Devices that the local network administrator admits to the particular class.

my-controller: Devices intended to serve as controllers for the MUD-URL that the Thing emitted.

local: The class of IP addresses that are scoped within some administrative boundary. By default it is suggested that this be the local subnet.

The "manufacturer" classes can be easily specified by the manufacturer, whereas controller classes are initially envisioned to be specified by the administrator.
Because manufacturers do not know who will be using their devices, it is important for functionality referenced in usage descriptions to be relatively ubiquitous and mature. For these reasons the YANG-based configuration in a MUD file is limited to either the modules specified or referenced in this document, or those specified in documented extensions.

1.8. The Manufacturer Usage Description Architecture

With these components laid out we now have the basis for an architecture. This leads us to ASCII art.

```
.................................
.                             .
.                             .
.                             .
.   End system network       |
.                             .
.                             .
.                             .
.   Thing                  --->MUD URL-->
.                             .
.                             .
```

Figure 1: MUD Architecture

In the above diagram, the switch or router collects MUD URLs and forwards them to the MUD manager (a network management system) for processing. This happens in different ways, depending on how the URL is communicated. For instance, in the case of DHCP, the DHCP server might receive the URL and then process it. In the case of IEEE 802.1X [IEEE8021X], the switch would carry the URL via a certificate to the authentication server via EAP over Radius [RFC3748], which would then process it. One method to do this is TEAP, described in [RFC7170]. The certificate extension is described below.

The information returned by the MUD file server is valid for as long as the Thing is connected. There is no expiry. However, if the MUD manager has detected that the MUD file for a Thing has changed, it SHOULD update the policy expeditiously, taking into account whatever approval flow is required in a deployment. In this way, new recommendations from the manufacturer can be processed in a timely fashion.
The information returned by the MUD file server (a web server) is valid for the duration of the Thing’s connection, or as specified in the description. Thus if the Thing is disconnected, any associated configuration in the switch can be removed. Similarly, from time to time the description may be refreshed, based on new capabilities or communication patterns or vulnerabilities.

The web server is typically run by or on behalf of the manufacturer. Its domain name is that of the authority found in the MUD URL. For legacy cases where Things cannot emit a URL, if the switch is able to determine the appropriate URL, it may proxy it. In the trivial case it may hardcode MUD-URL on a switch port or a map from some available identifier such as an L2 address or certificate hash to a MUD-URL.

The role of the MUD manager in this environment is to do the following:

- receive MUD URLs,
- fetch MUD files,
- translate abstractions in the MUD files to specific network element configuration,
- maintain and update any required mappings of the abstractions, and
- update network elements with appropriate configuration.

A MUD manager may be a component of a AAA or network management system. Communication within those systems and from those systems to network elements is beyond the scope of this memo.

1.9. Order of operations

As mentioned above, MUD contains architectural building blocks, and so order of operation may vary. However, here is one clear intended example:

1. Thing emits URL.

2. That URL is forwarded to a MUD manager by the nearest switch (how this happens depends on the way in which the MUD URL is emitted).

3. The MUD manager retrieves the MUD file and signature from the MUD file server, assuming it doesn’t already have copies. After validating the signature, it may test the URL against a web or domain reputation service, and it may test any hosts within the file against those reputation services, as it deems fit.
4. The MUD manager may query the administrator for permission to add the Thing and associated policy. If the Thing is known or the Thing type is known, it may skip this step.

5. The MUD manager instantiates local configuration based on the abstractions defined in this document.

6. The MUD manager configures the switch nearest the Thing. Other systems may be configured as well.

7. When the Thing disconnects, policy is removed.

2. The MUD Model and Semantic Meaning

A MUD file consists of a YANG model instance that has been serialized in JSON [RFC7951]. For purposes of MUD, the nodes that can be modified are access lists as augmented by this model. The MUD file is limited to the serialization of only the following YANG schema:

- ietf-access-control-list [I-D.ietf-netmod-acl-model]
- ietf-mud (this document)
- ietf-acldns (this document)

Extensions may be used to add additional schema. This is described further on.

To provide the widest possible deployment, publishers of MUD files SHOULD make use of the abstractions in this memo and avoid the use of IP addresses. A MUD manager SHOULD NOT automatically implement any MUD file that contains IP addresses, especially those that might have local significance. The addressing of one side of an access list is implicit, based on whether it is applied as to-device-policy or from-device-policy.

With the exceptions of "name" of the ACL, "type", "name" of the ACE, and TCP and UDP source and destination port information, publishers of MUD files SHOULD limit the use of ACL model leaf nodes expressed to those found in this specification. Absent any extensions, MUD files are assumed to implement only the following ACL model features:

- match-on-ipv4, match-on-ipv6, match-on-tcp, match-on-udp, match-on-icmp

Furthermore, only "accept" or "drop" actions SHOULD be included. A MUD manager MAY choose to interpret "reject" as "drop". A MUD manager SHOULD ignore all other actions. This is because
manufacturers do not have sufficient context within a local deployment to know whether reject is appropriate. That is a decision that should be left to a network administrator.

Given that MUD does not deal with interfaces, the support of the "ietf-interfaces" module [RFC8343] is not required. Specifically, the support of interface-related features and branches (e.g., interface-attachment and interface-stats) of the ACL YANG module is not required.

In fact, MUD managers MAY ignore any particular component of a description or MAY ignore the description in its entirety, and SHOULD carefully inspect all MUD descriptions. Publishers of MUD files MUST NOT include other nodes except as described in Section 3.9. See that section for more information.

2.1. The IETF-MUD YANG Module

This module is structured into three parts:

- The first component, the "mud" container, holds information that is relevant to retrieval and validity of the MUD file itself, as well as policy intended to and from the Thing.

- The second component augments the matching container of the ACL model to add several nodes that are relevant to the MUD URL, or otherwise abstracted for use within a local environment.

- The third component augments the tcp-acl container of the ACL model to add the ability to match on the direction of initiation of a TCP connection.

A valid MUD file will contain two root objects, a "mud" container and an "acls" container. Extensions may add additional root objects as required. As a reminder, when parsing acls, elements within a "match" block are logically ANDed. In general, a single abstraction in a match statement should be used. For instance, it makes little sense to match both "my-controller" and "controller" with an argument, since they are highly unlikely to be the same value.

A simplified graphical representation of the data models is used in this document. The meaning of the symbols in these diagrams is explained in [RFC8340].
module: ietf-mud

++-rw mud!
  +--rw mud-version    uint8
  +--rw mud-url        inet:uri
  +--rw last-update    yang:date-and-time
  +--rw mud-signature? inet:uri
  +--rw cache-validity? uint8
  +--rw is-supported   boolean
  +--rw systeminfo?    string
  +--rw mfg-name?      string
  +--rw model-name?    string
  +--rw firmware-rev?  string
  +--rw software-rev?  string
  +--rw documentation? inet:uri
  +--rw extensions*    string
  +--rw from-device-policy
    +--rw acls
      +--rw access-list* [name]
        +--rw name -> /acl:acls/acl/name
  +--rw to-device-policy
    +--rw acls
      +--rw access-list* [name]
        +--rw name -> /acl:acls/acl/name

  +--rw mud
    +--rw manufacturer? inet:host
    +--rw same-manufacturer? empty
    +--rw model? inet:uri
    +--rw local-networks? empty
    +--rw controller? inet:uri
    +--rw my-controller? empty
  /acl:14/acl:tcp/acl:tcp:
  +--rw direction-initiated? direction

3. MUD model definitions for the root mud container

3.1. mud-version

This node specifies the integer version of the MUD specification.
This memo specifies version 1.
3.2. mud-url

This URL identifies the MUD file. This is useful when the file and associated signature are manually uploaded, say, in an offline mode.

3.3. to-device-policy and from-device-policy containers

[I-D.ietf-netmod-acl-model] describes access-lists. In the case of MUD, a MUD file must be explicit in describing the communication pattern of a Thing, and that includes indicating what is to be permitted or denied in either direction of communication. Hence each of these containers indicates the appropriate direction of a flow in association with a particular Thing. They contain references to specific access-lists.

3.4. last-update

This is a date-and-time value of when the MUD file was generated. This is akin to a version number. Its form is taken from [RFC6991] which, for those keeping score, in turn was taken from Section 5.6 of [RFC3339], which was taken from [ISO.8601.1988].

3.5. cache-validity

This uint8 is the period of time in hours that a network management station MUST wait since its last retrieval before checking for an update. It is RECOMMENDED that this value be no less than 24 and MUST NOT be more than 168 for any Thing that is supported. This period SHOULD be no shorter than any period determined through HTTP caching directives (e.g., "cache-control" or "Expires"). N.B., expiring of this timer does not require the MUD manager to discard the MUD file, nor terminate access to a Thing. See Section 16 for more information.

3.6. is-supported

This boolean is an indication from the manufacturer to the network administrator as to whether or not the Thing is supported. In this context a Thing is said to not be supported if the manufacturer intends never to issue a firmware or software update to the Thing or never update the MUD file. A MUD manager MAY still periodically check for updates.

3.7. systeminfo

This is a textual UTF-8 description of the Thing to be connected. The intent is for administrators to be able to see a brief
displayable description of the Thing. It SHOULD NOT exceed 60 characters worth of display space.

3.8. mfg-name, software-rev, model-name firmware-rev

These optional fields are filled in as specified by [RFC8348]. Note that firmware-rev and software-rev MUST NOT be populated in a MUD file if the device can be upgraded but the MUD-URL cannot be. This would be the case, for instance, with MUD-URLs that are contained in 802.1AR certificates.

3.9. extensions

This optional leaf-list names MUD extensions that are used in the MUD file. Note that MUD extensions MUST NOT be used in a MUD file without the extensions being declared. Implementations MUST ignore any node in this file that they do not understand.

Note that extensions can either extend the MUD file as described in the previous paragraph, or they might reference other work. An extension example can be found in Appendix C.

4. Augmentation to the ACL Model

Note that in this section, when we use the term "match" we are referring to the ACL model "matches" node.

4.1. manufacturer

This node consists of a hostname that would be matched against the authority component of another Thing’s MUD URL. In its simplest form "manufacturer" and "same-manufacturer" may be implemented as access-lists. In more complex forms, additional network capabilities may be used. For example, if one saw the line "manufacturer" : "flobbidy.example.com", then all Things that registered with a MUD URL that contained flobbity.example.com in its authority section would match.

4.2. same-manufacturer

This null-valued node is an equivalent for when the manufacturer element is used to indicate the authority that is found in another Thing’s MUD URL matches that of the authority found in this Thing’s MUD URL. For example, if the Thing’s MUD URL were https://b1.example.com/ThingV1, then all devices that had MUD URL with an authority section of b1.example.com would match.
4.3. documentation

This URI consists of a URL that points to documentation relating to the device and the MUD file. This can prove particularly useful when the "controller" class is used, so that its use can be explained.

4.4. model

This string matches the entire MUD URL, thus covering the model that is unique within the context of the authority. It may contain not only model information, but versioning information as well, and any other information that the manufacturer wishes to add. The intended use is for devices of this precise class to match, to permit or deny communication between one another.

4.5. local-networks

This null-valued node expands to include local networks. Its default expansion is that packets must not traverse toward a default route that is received from the router. However, administrators may expand the expression as is appropriate in their deployments.

4.6. controller

This URI specifies a value that a controller will register with the MUD manager. The node then is expanded to the set of hosts that are so registered. This node may also be a URN. In this case, the URN describes a well known service, such as DNS or NTP, that has been standardized. Both of those URNs may be found in Section 17.6.

When "my-controller" is used, it is possible that the administrator will be prompted to populate that class for each and every model. Use of "controller" with a named class allows the user to populate that class only once for many different models that a manufacturer may produce.

Controller URIs MAY take the form of a URL (e.g. "http[s]://"). However, MUD managers MUST NOT resolve and retrieve such files, and it is RECOMMENDED that there be no such file at this time, as their form and function may be defined at a point in the future. For now, URLs should serve simply as class names and may be populated by the local deployment administrator.

Great care should be taken by MUD managers when invoking the controller class in the form of URLs. For one thing, it requires some understanding by the administrator as to when it is appropriate. Pre-registration in such classes by controllers with the MUD server
is encouraged. The mechanism to do that is beyond the scope of this work.

4.7. my-controller

This null-valued node signals to the MUD manager to use whatever mapping it has for this MUD URL to a particular group of hosts. This may require prompting the administrator for class members. Future work should seek to automate membership management.

4.8. direction-initiated

This MUST only be applied to TCP. This matches the direction in which a TCP connection is initiated. When direction initiated is "from-device", packets that are transmitted in the direction of a thing MUST be dropped unless the thing has first initiated a TCP connection. By way of example, this node may be implemented in its simplest form by looking at naked SYN bits, but may also be implemented through more stateful mechanisms.

When applied this matches packets when the flow was initiated in the corresponding direction. [RFC6092] specifies IPv6 guidance best practices. While that document is scoped specifically to IPv6, its contents are applicable for IPv4 as well.

5. Processing of the MUD file

To keep things relatively simple in addition to whatever definitions exist, we also apply two additional default behaviors:

- Anything not explicitly permitted is denied.
- Local DNS and NTP are, by default, permitted to and from the Thing.

An explicit description of the defaults can be found in Appendix B. These are applied AFTER all other explicit rules. Thus, a default behavior can be changed with a "drop" action.

6. What does a MUD URL look like?

MUD URLs are required to use the HTTPS scheme, in order to establish the MUD file server’s identity and assure integrity of the MUD file.

Any "https://" URL can be a MUD URL. For example:
A manufacturer may construct a MUD URL in any way, so long as it makes use of the "https" schema.

7. The MUD YANG Model

```yml
<CODE BEGINS>file "ietf-mud@2018-06-15.yang"
module ietf-mud {
  yang-version 1.1;
  namespace "urn:ietf:params:xml:ns:yang:ietf-mud";
  prefix ietf-mud;

  import ietf-access-control-list {
    prefix acl;
  }
  import ietf-yang-types {
    prefix yang;
  }
  import ietf-inet-types {
    prefix inet;
  }

  organization  
    "IETF OPSAWG (Ops Area) Working Group";
  contact  
    "WG Web: http://tools.ietf.org/wg/opsawg/
    WG List: opsawg@ietf.org
    Author: Eliot Lear
    lear@cisco.com
    Author: Ralph Droms
    rdroms@gmail.com
    Author: Dan Romascanu
    dromasca@gmail.com"

  description  
    "This YANG module defines a component that augments the IETF description of an access list. This specific module focuses on additional filters that include local, model, and same-manufacturer.

    This module is intended to be serialized via JSON and stored as a file, as described in RFC XXXX [RFC Editor to fill in with this document #]."

Lear, et al. Expiring December 17, 2018 [Page 19]"
typedef direction {
  type enumeration {
    enum to-device {
      description
      "packets or flows destined to the target Thing";
    }
    enum from-device {
      description
      "packets or flows destined from the target Thing";
    }
  }
  description
  "Which way are we talking about?";
}

container mud {
  presence "Enabled for this particular MUD URL";
  description
  "MUD related information, as specified by RFC-XXXX [RFC Editor to fill in].";
  uses mud-grouping;
}

grouping mud-grouping {
  description
  "Information about when support end(ed), and when to refresh";
}
leaf mud-version {
    type uint8;
    mandatory true;
    description
        "This is the version of the MUD specification. This memo specifies version 1.";
}
leaf mud-url {
    type inet:uri;
    mandatory true;
    description
        "This is the MUD URL associated with the entry found in a MUD file.";
}
leaf last-update {
    type yang:date-and-time;
    mandatory true;
    description
        "This is intended to be when the current MUD file was generated. MUD Managers SHOULD NOT check for updates between this time plus cache validity";
}
leaf mud-signature {
    type inet:uri;
    description
        "A URI that resolves to a signature as described in this specification.";
}
leaf cache-validity {
    type uint8 {
        range "1..168";
    }
    units "hours";
    default "48";
    description
        "The information retrieved from the MUD server is valid for these many hours, after which it should be refreshed. N.B. MUD manager implementations need not discard MUD files beyond this period.";
}
leaf is-supported {
    type boolean;
    mandatory true;
    description
        "This boolean indicates whether or not the Thing is currently supported by the manufacturer.";
}
leaf systeminfo {
type string;
description  
"A UTF-8 description of this Thing. This should be a brief description that may be displayed to the user to determine whether to allow the Thing on the network.";
}
leaf mfg-name {
type string;
description  
"Manufacturer name, as described in the ietf-hardware YANG module.";
}
leaf model-name {
type string;
description  
"Model name, as described in the ietf-hardware YANG module.";
}
leaf firmware-rev {
type string;
description  
"firmware-rev, as described in the ietf-hardware YANG module. Note this field MUST NOT be included when the device can be updated but the MUD-URL cannot.";
}
leaf software-rev {
type string;
description  
"software-rev, as described in the ietf-hardware YANG module. Note this field MUST NOT be included when the device can be updated but the MUD-URL cannot.";
}
leaf documentation {
type inet:uri;
description  
"This URL points to documentation that relates to this device and any classes that it uses in its MUD file. A caution: MUD managers need not resolve this URL on their own, but rather simply provide it to the administrator. Parsing HTML is not an intended function of a MUD manager.";
}
leaf-list extensions {
type string {

Lear, et al. Expires December 17, 2018 [Page 22]
length "1..40";
}
}description
"A list of extension names that are used in this MUD file. Each name is registered with the IANA and described in an RFC."
}
}container from-device-policy {
  description
  "The policies that should be enforced on traffic coming from the device. These policies are not necessarily intended to be enforced at a single point, but may be rendered by the controller to any relevant enforcement points in the network or elsewhere.";
  uses access-lists;
}
}container to-device-policy {
  description
  "The policies that should be enforced on traffic going to the device. These policies are not necessarily intended to be enforced at a single point, but may be rendered by the controller to any relevant enforcement points in the network or elsewhere.";
  uses access-lists;
}
}
}
}
grouping access-lists {
  description
  "A grouping for access lists in the context of device policy.";
}container access-lists {
  description
  "The access lists that should be applied to traffic to or from the device.";
}list access-list {
  key "name";
  description
  "Each entry on this list refers to an ACL that should be present in the overall access list data model. Each ACL is identified by name and type.";
}leaf name {
  type leafref {
    path "/acl:acls/acl:acl/acl:name";
  }
}
description
"The name of the ACL for this entry."
}
}
}
}
}
}


description
"adding abstractions to avoid need of IP addresses";

container mud {

description
"MUD-specific matches.";

leaf manufacturer {

type inet:host;

description
"A domain that is intended to match the authority section of the MUD URL. This node is used to specify one or more manufacturers a device should be authorized to access.";

}

leaf same-manufacturer {

type empty;

description
"This node matches the authority section of the MUD URL of a Thing. It is intended to grant access to all devices with the same authority section.";

}

leaf model {

type inet:uri;

description
"Devices of the specified model type will match if they have an identical MUD URL.";

}

leaf local-networks {

type empty;

description
"IP addresses will match this node if they are considered local addresses. A local address may be a list of locally defined prefixes and masks that indicate a particular administrative scope.";

}

leaf controller {

type inet:uri;

description
"This node names a class that has associated with it zero or more IP addresses to match against. These may be scoped to a manufacturer or via a standard
8. The Domain Name Extension to the ACL Model

This module specifies an extension to IETF-ACL model such that domain names may be referenced by augmenting the "matches" node. Different implementations may deploy differing methods to maintain the mapping between IP address and domain name, if indeed any are needed. However, the intent is that resources that are referred to using a name should be authorized (or not) within an access list.

The structure of the change is as follows:

```xml
module: ietf-acldns
    +---rw src-dnsname?    inet:host
    +---rw dst-dnsname?    inet:host
    +---rw src-dnsname?    inet:host
    +---rw dst-dnsname?    inet:host
```
The choice of these particular points in the access-list model is based on the assumption that we are in some way referring to IP-related resources, as that is what the DNS returns. A domain name in our context is defined in [RFC6991]. The augmentations are replicated across IPv4 and IPv6 to allow MUD file authors the ability to control the IP version that the Thing may utilize.

The following node are defined.

8.1. src-dnsname

The argument corresponds to a domain name of a source as specified by inet:host. A number of means may be used to resolve hosts. What is important is that such resolutions be consistent with ACLs required by Things to properly operate.

8.2. dst-dnsname

The argument corresponds to a domain name of a destination as specified by inet:host See the previous section relating to resolution.

Note when using either of these with a MUD file, because access is associated with a particular Thing, MUD files MUST NOT contain either a src-dnsname in an ACL associated with from-device-policy or a dst-dnsname associated with to-device-policy.

8.3. The ietf-acldns Model

<CODE BEGINS>file "ietf-acldns@2018-06-15.yang"
module ietf-acldns {
  yang-version 1.1;
  namespace "urn:ietf:params:xml:ns:yang:ietf-acldns";
  prefix ietf-acldns;

  import ietf-access-control-list {
    prefix acl;
  }
  import ietf-inet-types {
    prefix inet;
  }

  organization
    "IETF OPSAWG (Ops Area) Working Group";
  contact
    *WG Web: http://tools.ietf.org/wg/opsawg/
      WG List: opsawg@ietf.org
      Author: Eliot Lear
This YANG module defines a component that augments the IETF description of an access list to allow DNS names as matching criteria.

revision 2018-06-15 {
    description
    "Base version of dnsname extension of ACL model";
    reference
    "RFC XXXX: Manufacturer Usage Description Specification";
}
grouping dns-matches {
    description
    "Domain names for matching."
    leaf src-dnsname {
        type inet:host;
        description
        "domain name to be matched against";
    }
    leaf dst-dnsname {
        type inet:host;
        description
        "domain name to be matched against";
    }
}
    description
    "Adding domain names to matching";
    uses dns-matches;
}
"/acl:l3/acl:ipv6/acl:ipv6" {
    description
    "Adding domain names to matching";
    uses dns-matches;
}
}
9. MUD File Example

This example contains two access lists that are intended to provide outbound access to a cloud service on TCP port 443.

```json
{
    "ietf-mud:mud": {
        "mud-version": 1,
        "mud-url": "https://lighting.example.com/lightbulb2000",
        "last-update": "2018-03-02T11:20:51+01:00",
        "cache-validity": 48,
        "is-supported": true,
        "systeminfo": "The BMS Example Lightbulb",
        "from-device-policy": {
            "access-lists": {
                "access-list": {
                    "name": "mud-76100-v6fr"
                }
            }
        },
        "to-device-policy": {
            "access-lists": {
                "access-list": {
                    "name": "mud-76100-v6to"
                }
            }
        }
    },
    "ietf-access-control-list:acls": {
        "acl": {
            "name": "mud-76100-v6to",
            "type": "ipv6-acl-type",
            "aces": {
                "ace": {
                    "name": "cl0-to-dev",
                    "matches": {
                        "ipv6": {
                            "ietf-acldns:src-dnsname": "test.example.com",
                            "protocol": 6
                        }
                    },
                    "tcp": {
                        "ietf-mud:direction-initiated": "from-device",
```
"source-port": {
  "operator": "eq",
  "port": 443
},
"actions": {
  "forwarding": "accept"
}
},
"name": "mud-76100-v6fr",
"type": "ipv6-acl-type",
"aces": {
  "ace": {
    "name": "cl0-frdev",
    "matches": {
      "ipv6": {
        "ietf-acldns:dst-dnsname": "test.example.com",
        "protocol": 6
      },
      "tcp": {
        "ietf-mud:direction-initiated": "from-device",
        "destination-port": {
          "operator": "eq",
          "port": 443
        }
      }
    },
    "actions": {
      "forwarding": "accept"
    }
  }
}
}
}

In this example, two policies are declared, one from the Thing and the other to the Thing. Each policy names an access list that applies to the Thing, and one that applies from. Within each access
list, access is permitted to packets flowing to or from the Thing that can be mapped to the domain name of "service.bms.example.com". For each access list, the enforcement point should expect that the Thing initiated the connection.

10. The MUD URL DHCP Option

The IPv4 MUD URL client option has the following format:


Code OPTION_MUD_URL_V4 (161) is assigned by IANA. len is a single octet that indicates the length of MUD string in octets. The MUD string is defined as follows:

MUDstring = mudurl [ " " reserved ]
mudurl = URI; a URL [RFC3986] that uses the "https" schema [RFC7230] reserved = 1*( OCTET ) ; from [RFC5234]

The entire option MUST NOT exceed 255 octets. If a space follows the MUD URL, a reserved string that will be defined in future specifications follows. MUD managers that do not understand this field MUST ignore it.

The IPv6 MUD URL client option has the following format:


OPTION_MUD_URL_V6 (112; assigned by IANA).

option-length contains the length of the MUDstring, as defined above, in octets.

The intent of this option is to provide both a new Thing classifier to the network as well as some recommended configuration to the routers that implement policy. However, it is entirely the purview
of the network system as managed by the network administrator to
decide what to do with this information. The key function of this
option is simply to identify the type of Thing to the network in a
structured way such that the policy can be easily found with existing
toolsets.

10.1. Client Behavior

A DHCPv4 client MAY emit a DHCPv4 option and a DHCPv6 client MAY emit
DHCPv6 option. These options are singletons, as specified in
[RFC7227]. Because clients are intended to have at most one MUD URL
associated with them, they may emit at most one MUD URL option via
DHCPv4 and one MUD URL option via DHCPv6. In the case where both v4
and v6 DHCP options are emitted, the same URL MUST be used.

10.2. Server Behavior

A DHCP server may ignore these options or take action based on
receipt of these options. When a server consumes this option, it
will either forward the URL and relevant client information (such as
the gateway address or giaddr and requested IP address, and lease
length) to a network management system, or it will retrieve the usage
description itself by resolving the URL.

DHCP servers may implement MUD functionality themselves or they may
pass along appropriate information to a network management system or
MUD manager. A DHCP server that does process the MUD URL MUST adhere
to the process specified in [RFC2818] and [RFC5280] to validate the
TLS certificate of the web server hosting the MUD file. Those
servers will retrieve the file, process it, create and install the
necessary configuration on the relevant network element. Servers
SHOULD monitor the gateway for state changes on a given interface. A
DHCP server that does not provide MUD functionality and has forwarded
a MUD URL to a MUD manager MUST notify the MUD manager of any
corresponding change to the DHCP state of the client (such as
expiration or explicit release of a network address lease).

Should the DHCP server fail, in the case when it implements the MUD
manager functionality, any backup mechanisms SHOULD include the MUD
state, and the server SHOULD resolve the status of clients upon its
restart, similar to what it would do, absent MUD manager
functionality. In the case where the DHCP server forwards
information to the MUD manager, the MUD manager will either make use
of redundant DHCP servers for information, or otherwise clear state
based on other network information, such as monitoring port status on
a switch via SNMP, Radius accounting, or similar mechanisms.
10.3. Relay Requirements

There are no additional requirements for relays.

11. The Manufacturer Usage Description (MUD) URL X.509 Extension

This section defines an X.509 non-critical certificate extension that contains a single Uniform Resource Locator (URL) that points to an on-line Manufacturer Usage Description concerning the certificate subject. URI must be represented as described in Section 7.4 of [RFC5280].

Any Internationalized Resource Identifiers (IRIs) MUST be mapped to URIs as specified in Section 3.1 of [RFC3987] before they are placed in the certificate extension.

The semantics of the URL are defined Section 6 of this document.

The choice of id-pe is based on guidance found in Section 4.2.2 of [RFC5280]:

These extensions may be used to direct applications to on-line information about the issuer or the subject.

The MUD URL is precisely that: online information about the particular subject.

In addition, a separate new extension is defined as id-pe-mudsigner. This contains the subject field of the signing certificate of the MUD file. Processing of this field is specified in Section 13.2.

The purpose of this signature is to make a claim that the MUD file found on the server is valid for a given device, independent of any other factors. There are several security considerations below in Section 16.

A new content-type id-ct-mud is also defined. While signatures are detached today, should a MUD file be transmitted as part of a CMS message, this content-type SHOULD be used.

The new extension is identified as follows:

\texttt{<CODE BEGINS>}

\begin{verbatim}
MUDURLExtnModule-2016 { iso(1) identified-organization(3) dod(6)
internet(1) security(5) mechanisms(5) pkix(7)
id-mod(0) id-mod-mudURLExtn2016(88) }
DEFINITIONS IMPLICIT TAGS ::= BEGIN
\end{verbatim}

\texttt{<CODE ENDS>}
-- EXPORTS ALL --

IMPORTS

-- RFC 5912
EXTENSION
FROM PKIX-CommonTypes-2009
{ iso(1) identified-organization(3) dod(6) internet(1)
  security(5) mechanisms(5) pkix(7) id-mod(0)
  id-mod-pkixCommon-02(57) }

-- RFC 5912
id-ct
FROM PKIXCRMF-2009
{ iso(1) identified-organization(3) dod(6) internet(1)
  security(5) mechanisms(5) pkix(7) id-mod(0)
  id-mod-crmf2005-02(55) }

-- RFC 6268
CONTENT-TYPE
FROM CryptographicMessageSyntax-2010
{ iso(1) member-body(2) us(840) rsadsi(113549)
  pkcs(1) pkcs-9(9) smime(16) modules(0) id-mod-cms-2009(58) }

-- RFC 5912
id-pe, Name
FROM PKIX1Explicit-2009
{ iso(1) identified-organization(3) dod(6) internet(1)
  security(5) mechanisms(5) pkix(7) id-mod(0)
  id-mod-pkix1-explicit-02(51) };

-- Certificate Extensions
--

MUDCertExtensions EXTENSION ::= { ext-MUDURL | ext-MUDsigner, ... }

ext-MUDURL EXTENSION ::= { SYNTAX MUDURLSyntax IDENTIFIED BY id-pe-mud-url }

id-pe-mud-url OBJECT IDENTIFIER ::= { id-pe 25 }

MUDURLSyntax ::= IA5String

ext-MUDsigner EXTENSION ::= { SYNTAX MUDsignerSyntax IDENTIFIED BY id-pe-mudsigner }
id-pe-mudsinger OBJECT IDENTIFIER ::= { id-pe TBD1 }

MUDsignerSyntax ::= Name

--
-- CMS Content Types
--

MUDContentTypes CONTENT-TYPE ::= 
  { ct-mud, ... }

ct-mud CONTENT-TYPE ::= 
  { -- directly include the content
    IDENTIFIED BY id-ct-mudtype }
-- The binary data that is in the form
-- 'application/mud+json" is directly encoded as the
-- signed data. No additional ASN.1 encoding is added.

id-ct-mudtype OBJECT IDENTIFIER ::= { id-ct TBD2 }

END

<CODE ENDS>

While this extension can appear in either an 802.AR manufacturer certificate (IDevID) or deployment certificate (LDevID), of course it is not guaranteed in either, nor is it guaranteed to be carried over. It is RECOMMENDED that MUD manager implementations maintain a table that maps a Thing to its MUD URL based on IDevIDs.

12.  The Manufacturer Usage Description LLDP extension

The IEEE802.1AB Link Layer Discovery Protocol (LLDP) is a one hop vendor-neutral link layer protocol used by end hosts network Things for advertising their identity, capabilities, and neighbors on an IEEE 802 local area network. Its Type-Length-Value (TLV) design allows for 'vendor-specific’ extensions to be defined. IANA has a registered IEEE 802 organizationally unique identifier (OUI) defined as documented in [RFC7042]. The MUD LLDP extension uses a subtype defined in this document to carry the MUD URL.

The LLDP vendor specific frame has the following format:

+--------+--------+----------+---------+--------------
|TLV Type|  len   |   OUI    |subtype  | MUDString    |
|  =127  |        |= 00 00 5E|  = 1    |              |
| (7 bits)|(9 bits)|(3 octets)|(1 octet) (1-255 octets)
+--------+--------+----------+---------+--------------
where:

- TLV Type = 127 indicates a vendor-specific TLV
- len - indicates the TLV string length
- OUI = 00 00 5E is the organizationally unique identifier of IANA
- subtype = 1 (to be assigned by IANA for the MUD URL)
- MUD URL - the length MUST NOT exceed 255 octets

The intent of this extension is to provide both a new Thing classifier to the network as well as some recommended configuration to the routers that implement policy. However, it is entirely the purview of the network system as managed by the network administrator to decide what to do with this information. The key function of this extension is simply to identify the type of Thing to the network in a structured way such that the policy can be easily found with existing toolsets.

Hosts, routers, or other network elements that implement this option are intended to have at most one MUD URL associated with them, so they may transmit at most one MUD URL value.

Hosts, routers, or other network elements that implement this option may ignore these options or take action based on receipt of these options. For example they may fill in information in the respective extensions of the LLDP Management Information Base (LLDP MIB). LLDP operates in a one-way direction. LLDPDUs are not exchanged as information requests by one Thing and response sent by another Thing. The other Things do not acknowledge LLDP information received from a Thing. No specific network behavior is guaranteed. When a Thing consumes this extension, it may either forward the URL and relevant remote Thing information to a MUD manager, or it will retrieve the usage description by resolving the URL in accordance with normal HTTP semantics.

13. Creating and Processing of Signed MUD Files

Because MUD files contain information that may be used to configure network access lists, they are sensitive. To ensure that they have not been tampered with, it is important that they be signed. We make use of DER-encoded Cryptographic Message Syntax (CMS) [RFC5652] for this purpose.
13.1. Creating a MUD file signature

A MUD file MUST be signed using CMS as an opaque binary object. In order to make successful verification more likely, intermediate certificates SHOULD be included. The signature is stored at the location specified in the MUD file. Signatures are transferred using content-type "application/pkcs7-signature".

For example:

```
% openssl cms -sign -signer mancertfile -inkey mankey \
  -in mudfile -binary -outform DER -binary \
  -certfile intermediatecert -out mudfile.p7s
```

Note: A MUD file may need to be re-signed if the signature expires.

13.2. Verifying a MUD file signature

Prior to processing the rest of a MUD file, the MUD manager MUST retrieve the MUD signature file by retrieving the value of "mud-signature" and validating the signature across the MUD file. The Key Usage Extension in the signing certificate MUST be present and have the bit digitalSignature(0) set. When the id-pe-mudsigner extension is present in a device’s X.509 certificate, the MUD signature file MUST have been generated by a certificate whose subject matches the contents of that id-pe-mudsigner extension. If these conditions are not met, or if it cannot validate the chain of trust to a known trust anchor, the MUD manager MUST cease processing the MUD file until an administrator has given approval.

The purpose of the signature on the file is to assign accountability to an entity, whose reputation can be used to guide administrators on whether or not to accept a given MUD file. It is already common place to check web reputation on the location of a server on which a file resides. While it is likely that the manufacturer will be the signer of the file, this is not strictly necessary, and may not be desirable. For one thing, in some environments, integrators may install their own certificates. For another, what is more important is the accountability of the recommendation, and not just the relationship between the Thing and the file.

An example:

```
% openssl cms -verify -in mudfile.p7s -inform DER -content mudfile
```

Note the additional step of verifying the common trust root.
14. Extensibility

One of our design goals is to see that MUD files are able to be understood by as broad a cross-section of systems as is possible. Coupled with the fact that we have also chosen to leverage existing mechanisms, we are left with no ability to negotiate extensions and a limited desire for those extensions in any event. As such, a two-tier extensibility framework is employed, as follows:

1. At a coarse grain, a protocol version is included in a MUD URL. This memo specifies MUD version 1. Any and all changes are entertained when this version is bumped. Transition approaches between versions would be a matter for discussion in future versions.

2. At a finer grain, only extensions that would not incur additional risk to the Thing are permitted. Specifically, adding nodes to the mud container is permitted with the understanding that such additions will be ignored by unaware implementations. Any such extensions SHALL be standardized through the IETF process, and MUST be named in the "extensions" list. MUD managers MUST ignore YANG nodes they do not understand and SHOULD create an exception to be resolved by an administrator, so as to avoid any policy inconsistencies.

15. Deployment Considerations

Because MUD consists of a number of architectural building blocks, it is possible to assemble different deployment scenarios. One key aspect is where to place policy enforcement. In order to protect the Thing from other Things within a local deployment, policy can be enforced on the nearest switch or access point. In order to limit unwanted traffic within a network, it may also be advisable to enforce policy as close to the Internet as possible. In some circumstances, policy enforcement may not be available at the closest hop. At that point, the risk of lateral infection (infection of devices that reside near one another) is increased to the number of Things that are able to communicate without protection.

A caution about some of the classes: admission of a Thing into the "manufacturer" and "same-manufacturer" class may have impact on access of other Things. Put another way, the admission may grow the access-list on switches connected to other Things, depending on how access is managed. Some care should be given on managing that access-list growth. Alternative methods such as additional network segmentation can be used to keep that growth within reason.
Because as of this writing MUD is a new concept, one can expect a great many devices to not have implemented it. It remains a local deployment decision as to whether a device that is first connected should be allowed broad or limited access. Furthermore, as mentioned in the introduction, a deployment may choose to ignore a MUD policy in its entirety, but simply taken into account the MUD URL as a classifier to be used as part of a local policy decision.

Finally, please see directly below regarding device lifetimes and use of domain names.

16. Security Considerations

Based on how a MUD URL is emitted, a Thing may be able to lie about what it is, thus gaining additional network access. This can happen in a number of ways when a device emits a MUD URL using DHCP or LLDP, such as being inappropriately admitted to a class such as "same-manufacturer", given access to a device such as "my-controller", or being permitted access to an Internet resource, where such access would otherwise be disallowed. Whether that is the case will depend on the deployment. Implementations SHOULD be configurable to disallow additive access for devices using MUD-URLs that are not emitted in a secure fashion such as in a certificate. Similarly, implementations SHOULD NOT grant elevated permissions (beyond those of devices presenting no MUD policy) to devices which do not strongly bind their identity to their L2/L3 transmissions. When insecure methods are used by the MUD Manager, the classes SHOULD NOT contain devices that use both insecure and secure methods, in order to prevent privilege escalation attacks, and MUST NOT contain devices with the same MUD-URL that are derived from both strong and weak authentication methods.

Devices may forge source (L2/L3) information. Deployments should apply appropriate protections to bind communications to the authentication that has taken place. For 802.1X authentication, IEEE 802.1AE (MACsec) [IEEE8021AE] is one means by which this may happen. A similar approach can be used with 802.11i (WPA2) [IEEE80211i]. Other means are available with other lower layer technologies. Implementations using session-oriented access that is not cryptographically bound should take care to remove state when any form of break in the session is detected.

A rogue CA may sign a certificate that contains the same subject name as is listed in the MUDsigner field in the manufacturer certificate, thus seemingly permitting a substitute MUD file for a device. There are two mitigations available: first, if the signer changes, this may be flagged as an exception by the MUD manager. If the MUD file also changes, the MUD manager SHOULD seek administrator approval (it
should do this in any case). In all circumstances, the MUD manager MUST maintain a cache of trusted CAs for this purpose. When such a rogue is discovered, it SHOULD be removed.

Additional mitigations are described below.

When certificates are not present, Things claiming to be of a certain manufacturer SHOULD NOT be included in that manufacturer grouping without additional validation of some form. This will be relevant when the MUD manager makes use of primitives such as "manufacturer" for the purpose of accessing Things of a particular type. Similarly, network management systems may be able to fingerprint the Thing. In such cases, the MUD URL can act as a classifier that can be proven or disproven. Fingerprinting may have other advantages as well: when 802.1AR certificates are used, because they themselves cannot change, fingerprinting offers the opportunity to add artifacts to the MUD string in the form of the reserved field discussed in Section 10. The meaning of such artifacts is left as future work.

MUD managers SHOULD NOT accept a usage description for a Thing with the same MAC address that has indicated a change of the URL authority without some additional validation (such as review by a network administrator). New Things that present some form of unauthenticated MUD URL SHOULD be validated by some external means when they would be given increased network access.

It may be possible for a rogue manufacturer to inappropriately exercise the MUD file parser, in order to exploit a vulnerability. There are three recommended approaches to address this threat. The first is to validate that the signer of the MUD file is known to and trusted by the MUD manager. The second is to have a system do a primary scan of the file to ensure that it is both parseable and believable at some level. MUD files will likely be relatively small, to start with. The number of ACEs used by any given Thing should be relatively small as well. It may also be useful to limit retrieval of MUD URLs to only those sites that are known to have decent web or domain reputations.

Use of a URL necessitates the use of domain names. If a domain name changes ownership, the new owner of that domain may be able to provide MUD files that MUD managers would consider valid. There are a few approaches that can mitigate this attack. First, MUD managers SHOULD cache certificates used by the MUD file server. When a new certificate is retrieved for whatever reason, the MUD manager should check to see if ownership of the domain has changed. A fair programmatic approximation of this is when the name servers for the domain have changed. If the actual MUD file has changed, the MUD manager MAY check the WHOIS database to see if registration ownership...
of a domain has changed. If a change has occurred, or if for some reason it is not possible to determine whether ownership has changed, further review may be warranted. Note, this remediation does not take into account the case of a Thing that was produced long ago and only recently fielded, or the case where a new MUD manager has been installed.

The release of a MUD URL by a Thing reveals what the Thing is, and provides an attacker with guidance on what vulnerabilities may be present.

While the MUD URL itself is not intended to be unique to a specific Thing, the release of the URL may aid an observer in identifying individuals when combined with other information. This is a privacy consideration.

In addressing both of these concerns, implementors should take into account what other information they are advertising through mechanisms such as mDNS [RFC6872], how a Thing might otherwise be identified, perhaps through how it behaves when it is connected to the network, whether a Thing is intended to be used by individuals or carry personal identifying information, and then apply appropriate data minimization techniques. One approach is to make use of TEAP [RFC7170] as the means to share information with authorized components in the network. Network elements may also assist in limiting access to the MUD URL through the use of mechanisms such as DHCPv6-Shield [RFC7610].

There is the risk of the MUD manager itself being spied on to determine what things are connected to the network. To address this risk, MUD managers may choose to make use of TLS proxies that they trust that would aggregate other information.

Please note that the security considerations mentioned in Section 4.7 of [I-D.ietf-netmod-rfc6087bis] are not applicable in this case because the YANG serialization is not intended to be accessed via NETCONF. However, for those who try to instantiate this model in a network element via NETCONF, all objects in each model in this draft exhibit similar security characteristics as [I-D.ietf-netmod-acl-model]. The basic purpose of MUD is to configure access, and so by its very nature can be disruptive if used by unauthorized parties.

17. IANA Considerations

[ There was originally a registry entry for .well-known suffixes. This has been removed from the draft and may be marked as deprecated in the registry. RFC Editor: please remove this comment. ]
17.1. YANG Module Registrations

The following YANG modules are requested to be registered in the "IANA Module Names" registry:

The ietf-mud module:
- Name: ietf-mud
- Prefix: ietf-mud
- Registrant contact: The IESG
- Reference: [RFCXXXX]

The ietf-acldns module:
- Name: ietf-acldns
- Prefix: ietf-acldns
- Registrant: the IESG
- Reference: [RFCXXXX]

17.2. DHCPv4 and DHCPv6 Options

The IANA has allocated option 161 in the Dynamic Host Configuration Protocol (DHCP) and Bootstrap Protocol (BOOTP) Parameters registry for the MUD DHCPv4 option, and option 112 for DHCPv6, as described in Section 10.

17.3. PKIX Extensions

IANA is kindly requested to make the following assignments for:

- The MUDURLExtnModule-2016 ASN.1 module in the "SMI Security for PKIX Module Identifier" registry (1.3.6.1.5.5.7.0).
- id-pe-mud-url object identifier from the "SMI Security for PKIX Certificate Extension" registry (1.3.6.1.5.5.7.1).
- id-pe-mudsigner object identifier from the "SMI Security for PKIX Certificate Extension" registry (TBD1).
o id-ct-mudtype object identifier from the "SMI Security for S/MIME CMS Content Type" registry (TBD2).

The use of these values is specified in Section 11.

17.4. MIME Media-type Registration for MUD files

The following media-type is defined for transfer of MUD file:

- Type name: application
- Subtype name: mud+json
- Required parameters: n/a
- Optional parameters: n/a
- Encoding considerations: 8bit; application/mud+json values are represented as a JSON object; UTF-8 encoding MUST be employed. [RFC3629]
- Interoperability considerations: n/a
- Published specification: [RFCXXXX]
- Applications that use this media type: MUD managers as specified by [RFCXXXX].
- Fragment identifier considerations: n/a
- Additional information:

  Magic number(s): n/a
  File extension(s): n/a
  Macintosh file type code(s): n/a

  Person & email address to contact for further information:
  Eliot Lear <lear@cisco.com>, Ralph Droms <rdroms@gmail.com>
- Intended usage: COMMON
- Restrictions on usage: none
- Author:
  Eliot Lear <lear@cisco.com>
  Ralph Droms <rdroms@gmail.com>
- Change controller: IESG
- Provisional registration? (standards tree only): No.

17.5. LLDP IANA TLV Subtype Registry

IANA is requested to create a new registry for IANA Link Layer Discovery Protocol (LLDP) TLV subtype values. The recommended policy for this registry is Expert Review. The maximum number of entries in the registry is 256.

IANA is required to populate the initial registry with the value:
LLDP subtype value = 1 (All the other 255 values should be initially marked as ‘Unassigned’.)

Description = the Manufacturer Usage Description (MUD) Uniform Resource Locator (URL)

Reference = < this document >

17.6. The MUD Well Known Universal Resource Name (URNs)

The following parameter registry is requested to be added in accordance with [RFC3553]

Registry name: "urn:ietf:params:mud" is requested.
Specification: this document
Repository: this document
Index value: Encoded identically to a TCP/UDP port service name, as specified in Section 5.1 of [RFC6335]

The following entries should be added to the "urn:ietf:params:mud" name space:


17.7. Extensions Registry

The IANA is requested to establish a registry of extensions as follows:

Registry name: MUD extensions registry
Registry policy: Standards action
Standard reference: document
Extension name: UTF-8 encoded string, not to exceed 40 characters.

Each extension MUST follow the rules specified in this specification. As is usual, the IANA issues early allocations based in accordance with [RFC7120].

18. Acknowledgments

The authors would like to thank Einar Nilsen-Nygaard, who singlehandedly updated the model to match the updated ACL model, Bernie Volz, Tom Gindin, Brian Weis, Sandeep Kumar, Thorsten Dahm, John Bashinski, Steve Rich, Jim Bieda, Dan Wing, Joe Clarke, Henk Birkholz, Adam Montville, Jim Schaad, and Robert Sparks for their valuable advice and reviews. Russ Housley entirely rewrote
Section 11 to be a complete module. Adrian Farrel provided the basis for privacy considerations text. Kent Watsen provided a thorough review of the architecture and the YANG model. The remaining errors in this work are entirely the responsibility of the authors.

19. References

19.1. Normative References

[I-D.ietf-netmod-acl-model]

[IEEE8021AB]
Institute for Electrical and Electronics Engineers, "IEEE Standard for Local and Metropolitan Area Networks--Station and Media Access Control Connectivity Discovery", n.d..


19.2. Informative References


[ISO.8601.1988]  

[RFC1984]  

[RFC3339]  

[RFC3553]  

[RFC6092]  

[RFC6872]  

[RFC7042]  

[RFC7170]  

[RFC7252]  


Appendix A. Changes from Earlier Versions

RFC Editor to remove this section prior to publication.

Draft -19: * Edits after discussion with apps area to address reserved field for the future. * Correct systeminfo to be utf8. * Remove "hardware-rev" from list.

Draft -18: * Correct an error in the augment statement * Changes to the ACL model re ports.

Draft -17:
  o One editorial.

Draft -16
  o add mud-signature element based on review comments
  o redo mud-url
  o make clear that systeminfo uses UTF8

Draft -13 to -14:
  o Final WGLC comments and review comments
  o Move version from MUD-URL to Model
  o Have MUD-URL in model
  o Update based on update to draft-ietf-netmod-acl-model
  o Point to tree diagram draft instead of 6087bis.
Draft -12 to -13:
  o Additional WGLC comments

Draft -10 to -12:
These are based on WGLC comments:
  o Correct examples based on ACL model changes.
  o Change ordering nodes.
  o Additional explanatory text around systeminfo.
  o Change ordering in examples.
  o Make it VERY VERY VERY VERY clear that these are recommendations, not mandates.
  o DHCP -> NTP in some of the intro text.
  o Remove masa-server
  o "Things" to "network elements" in a few key places.
  o Reference to JSON YANG RFC added.

Draft -10 to -11:
  o Example corrections
  o Typo
  o Fix two lists.
  o Addition of 'any-acl' and 'mud-acl' in the list of allowed features.
  o Clarification of what should be in a MUD file.

Draft -09 to -10:
  o AD input.
  o Correct dates.
  o Add compliance sentence as to which ACL module features are implemented.
Draft -08 to -09:

- Resolution of Security Area review, IoT directorate review, GenART review, YANG doctors review.
- Change of YANG structure to address mandatory nodes.
- Terminology cleanup.
- Specify out extra portion of MUD-URL.
- Consistency changes.
- Improved YANG descriptions.
- Remove extra revisions.
- Track ACL model changes.
- Additional cautions on use of ACL model; further clarifications on extensions.

Draft -07 to -08:

- A number of editorials corrected.
- Definition of MUD file tweaked.

Draft -06 to -07:

- Examples updated.
- Additional clarification for direction-initiated.
- Additional implementation guidance given.

Draft -06 to -07:

- Update models to match new ACL model
- Extract directionality from the ACL, introducing a new device container.

Draft -05 to -06:

- Make clear that this is a component architecture (Polk and Watson)
- Add order of operations (Watson)
- Add extensions leaf-list (Pritikin)
- Remove previous-mud-file (Watson)
- Modify text in last-update (Watson)
- Clarify local networks (Weis, Watson)
- Fix contact info (Watson)
- Terminology clarification (Weis)
- Advice on how to handle LDevIDs (Watson)
- Add deployment considerations (Watson)
- Add some additional text about fingerprinting (Watson)
- Appropriate references to 6087bis (Watson)
- Change systeminfo to a URL to be referenced (Lear)

Draft -04 to -05: * syntax error correction
Draft -03 to -04: * Re-add my-controller
Draft -02 to -03: * Additional IANA updates * Format correction in YANG. * Add reference to TEAP.
Draft -01 to -02: * Update IANA considerations * Accept Russ Housley rewrite of X.509 text * Include privacy considerations text * Redo the URL limit. Still 255 bytes, but now stated in the URL definition. * Change URI registration to be under urn:ietf:params
Draft -00 to -01: * Fix cert trust text. * change supportInformation to meta-info * Add an informational element in. * add urn registry and create first entry * add default elements

Appendix B. Default MUD nodes

What follows is the portion of a MUD file that permits DNS traffic to a controller that is registered with the URN "urn:ietf:params:mud:dn" and traffic NTP to a controller that is registered "urn:ietf:params:mud:ntp". This is considered the default behavior and the ACEs are in effect appended to whatever other "ace" entries that a MUD file contains. To block DNS or NTP one repeats the matching statement but replaces the "forwarding" action "accept" with "drop". Because ACEs are processed in the order they are
received, the defaults would not be reached. A MUD manager might further decide to optimize to simply not include the defaults when they are overridden.

Four "acl" list entries that implement default MUD nodes are listed below. Two are for IPv4 and two are for IPv6 (one in each direction for both versions of IP). Note that neither access-list name nor ace name need be retained or used in any way by local implementations, but are simply there for completeness’ sake.

```json
"ietf-access-control-list:acls": {
  "acl": [
    {
      "name": "mud-59776-v4to",
      "type": "ipv4-acl-type",
      "aces": {
        "ace": [
          {
            "name": "ent0-todev",
            "matches": {
              "ietf-mud:mud": {
                "controller": "urn:ietf:params:mud:dns"
              },
              "ipv4": {
                "protocol": 17
              },
              "udp": {
                "source-port": {
                  "operator": "eq",
                  "port": 53
                }
              }
            },
            "actions": {
              "forwarding": "accept"
            }
          }
        ]
      }
    },
    {
      "name": "ent1-todev",
      "matches": {
        "ietf-mud:mud": {
          "controller": "urn:ietf:params:mud:ntp"
        },
        "ipv4": {
          "protocol": 17
        },
        "udp": {
          "source-port": {
            "operator": "eq",
            "port": 53
          }
        }
      }
    }
  ]
}
```
"operator": "eq",
"port": 123
}
}

"actions": {
  "forwarding": "accept"
}
}

"name": "mud-59776-v4fr",
"type": "ipv4-acl-type",
"aces": {
  "ace": {
    "name": "ent0-frdev",
    "matches": {
      "ietf-mud:mud": {
        "controller": "urn:ietf:params:mud:dns"
      },
      "ipv4": {
        "protocol": 17
      },
      "udp": {
        "destination-port": {
          "operator": "eq",
          "port": 53
        }
      }
    },
    "actions": {
      "forwarding": "accept"
    }
  },
  "name": "ent1-frdev",
  "matches": {
    "ietf-mud:mud": {
      "controller": "urn:ietf:params:mud:ntp"
    },
    "ipv4": {
      "protocol": 17
    },
    "udp": {
      "destination-port": {
        "operator": "eq",
        "port": 53
      }
    }
  }
}
"operator": "eq",
"port": 123
}
}
"actions": {
"forwarding": "accept"
}
]
},

"name": "mud-59776-v6to",
"type": "ipv6-acl-type",
"aces": {
"ace": [
{
"name": "ent0-todev",
"matches": {
"ietf-mud:mud": {
"controller": "urn:ietf:params:mud:dns"
},
"ipv6": {
"protocol": 17
},
"udp": {
"source-port": {
"operator": "eq",
"port": 53
}
}
"actions": {
"forwarding": "accept"
}
}
},

"name": "ent1-todev",
"matches": {
"ietf-mud:mud": {
"controller": "urn:ietf:params:mud:ntp"
},
"ipv6": {
"protocol": 17
},
"udp": {
"source-port": {

"operator": "eq",
"port": 123
}
}
"actions": {
"forwarding": "accept"
}
]
}

"name": "mud-59776-v6fr",
"type": "ipv6-acl-type",
"aces": {
"ace": [
{
"name": "ent0-frdev",
"matches": {
"ietf-mud:mud": {
"controller": "urn:ietf:params:mud:dns"
},
"ipv6": {
"protocol": 17
},
"udp": {
"destination-port": {
"operator": "eq",
"port": 53
}
}
"actions": {
"forwarding": "accept"
}
}
{
"name": "ent1-frdev",
"matches": {
"ietf-mud:mud": {
"controller": "urn:ietf:params:mud:ntp"
},
"ipv6": {
"protocol": 17
},
"udp": {
"destination-port": {
}
Appendix C. A Sample Extension: DETNET-indicator

In this sample extension we augment the core MUD model to indicate whether the device implements DETNET. If a device claims not to use DETNET, but then later attempts to do so, a notification or exception might be generated. Note that this example is intended only for illustrative purposes.

Extension Name: "Example-Extension" (to be used in the extensions list)
Standard: this document (but do not register the example)

This extension augments the MUD model to include a single node, using the following sample module that has the following tree structure:

module: ietf-mud-detext-example
    augment /ietf-mud:mud:
        ---rw is-detnet-required? boolean

The model is defined as follows:

<CODE BEGINS>file "ietf-mud-detext-example@2018-06-15.yang"
module ietf-mud-detext-example {
    yang-version 1.1;
    namespace "urn:ietf:params:xml:ns:yang:ietf-mud-detext-example";
    prefix ietf-mud-detext-example;

    import ietf-mud {
        prefix ietf-mud;
    }

Lear, et al.            Expires December 17, 2018              [Page 57]
organization
"IETF OPSAWG (Ops Area) Working Group";

contact
"WG Web: http://tools.ietf.org/wg/opsawg/
WG List: opsawg@ietf.org
Author: Eliot Lear
lear@cisco.com
Author: Ralph Droms
rdroms@gmail.com
Author: Dan Romascanu
dromasca@gmail.com
"

description
"Sample extension to a MUD module to indicate a need for DETNET support.";

revision 2018-06-15 {

description
"Initial revision.";

reference
"RFC XXXX: Manufacturer Usage Description Specification";
}

augment "/ietf-mud:mud" {

description
"This adds a simple extension for a manufacturer to indicate whether DETNET is required by a device.";

leaf is-detnet-required {

type boolean;

description
"This value will equal true if a device requires detnet to properly function";
}
}

Using the previous example, we now show how the extension would be expressed:

{
"ietf-mud:mud": {
"mud-version": 1,
"mud-url": "https://lighting.example.com/lightbulb2000",
"last-update": "2018-03-02T11:20:51+01:00",
}
"cache-validity": 48,
"extensions": [
   "ietf-mud-detext-example"
],
"ietf-mud-detext-example:is-detnet-required": "false",
"is-supported": true,
"systeminfo": "The BMS Example Lightbulb",
"from-device-policy": {
   "access-lists": {
      "access-list": [
         {  
            "name": "mud-76100-v6fr"
         }
      ]
   }
}]
"
"to-device-policy": {
   "access-lists": {
      "access-list": [
         {  
            "name": "mud-76100-v6to"
         }
      ]
   }
}]
"ietf-access-control-list:acls": {
   "acl": [
      
      {  
         "name": "mud-76100-v6to",
         "type": "ipv6-acl-type",
         "aces": {
            "ace": [
            {  
               "name": "c10-todev",
               "matches": {
                  "ipv6": {
                     
                     "ietf-acldns:src-dnsname": "test.example.com",
                     "protocol": 6
                  },
                  "tcp": {
                     "ietf-mud:direction-initiated": "from-device",
                     "source-port": {
                        "operator": "eq",
                        "port": 443
                     }
                  }
               }
            }
         }
      }
   ]
}
"actions": {
   "forwarding": "accept"
}
}
}
"name": "mud-76100-v6fr",
"type": "ipv6-acl-type",
"aces": [
   "ace": {
      "name": "cl0-frdev",
      "matches": {
         "ipv6": {
            "ietf-acldns:dst-dnsname": "test.example.com",
            "protocol": 6
         },
         "tcp": {
            "ietf-mud:direction-initiated": "from-device",
            "destination-port": {
               "operator": "eq",
               "port": 443
            }
         }
      },
      "actions": {
         "forwarding": "accept"
      }
   }
}
Authors' Addresses
Recommendations on the Filtering of IPv6 Packets Containing IPv6 Extension Headers
draft-ietf-opsec-ipv6-eh-filtering-03

Abstract

It is common operator practice to mitigate security risks by enforcing appropriate packet filtering. This document analyzes both the general security implications of IPv6 Extension Headers and the specific security implications of each Extension Header and Option type. Additionally, it discusses the operational and interoperability implications of discarding packets based on the IPv6 Extension Headers and IPv6 options they contain. Finally, it provides advice on the filtering of such IPv6 packets at transit routers, for those cases in which such filtering is deemed as necessary.

Status of This Memo

This Internet-Draft is submitted in full conformance with the provisions of BCP 78 and BCP 79.

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1. Introduction

Recent studies (see e.g. [RFC7872]) suggest that there is widespread dropping of IPv6 packets that contain IPv6 Extension Headers (EHs). In some cases, such packet drops occur at transit routers. While some operators "officially" drop packets that contain IPv6 EHs, it is possible that some of the measured packet drops be the result of...
improper configuration defaults, or inappropriate advice in this area.

This document analyzes both the general security implications of IPv6 EHs and the specific security implications of each EH and Option type, and provides advice on the filtering of IPv6 packets based on the IPv6 EHs and the IPv6 options they contain. Since various protocols may use IPv6 EHs (possibly with IPv6 options), discarding packets based on the IPv6 EHs or IPv6 options they contain may have implications on the proper functioning of such protocols. Thus, this document also attempts to discuss the operational and interoperability implications of such filtering policies.

The filtering policy typically depends on where in the network such policy is enforced: when the policy is enforced in a transit network, the policy typically follows a "black-list" approach, where only packets with clear negative implications are dropped. On the other hand, when the policy is enforced closer to the destination systems, the policy typically follows a "white-list" approach, where only traffic that is expected to be received is allowed. The advice in this document is aimed only at transit routers that may need to enforce a filtering policy based on the EHs and IPv6 options a packet may contain, following a "black-list" approach, and hence is likely to be much more permissive that a filtering policy to be employed e.g. at the edge of an enterprise network. The advice in this document is meant to improve the current situation of the dropping of packets with IPv6 EHs in the Internet [RFC7872].

This document is similar in nature to [RFC7126], which addresses the same problem for the IPv4 case. However, in IPv6, the problem space is compounded by the fact that IPv6 specifies a number of IPv6 EHs, and a number of IPv6 options which may be valid only when included in specific EH types.

This document completes and complements the considerations for protecting the control plane from packets containing IP options that can be found in [RFC6192].

Section 2 of this document specifies the terminology and conventions employed throughout this document. Section 3 of this document discusses IPv6 EHs and provides advice in the area of filtering IPv6 packets that contain such IPv6 EHs. Section 4 of this document discusses IPv6 options and provides advice in the area of filtering IPv6 packets that contain such options.
2. Terminology and Conventions Used in This Document

2.1. Terminology

The terms "fast path", "slow path", and associated relative terms ("faster path" and "slower path") are loosely defined as in Section 2 of [RFC6398].

The terms "permit" (allow the traffic), "drop" (drop with no notification to sender), and "reject" (drop with appropriate notification to sender) are employed as defined in [RFC3871]. Throughout this document we also employ the term "discard" as a generic term to indicate the act of discarding a packet, irrespective of whether the sender is notified of such drops, and irrespective of whether the specific filtering action is logged.

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in [RFC2119].

2.2. Conventions

This document assumes that nodes comply with the requirements in [RFC7045]. Namely (from [RFC7045]),

- If a forwarding node discards a packet containing a standard IPv6 EH, it MUST be the result of a configurable policy and not just the result of a failure to recognise such a header.

- The discard policy for each standard type of EH MUST be individually configurable.

- The default configuration SHOULD allow all standard IPv6 EHs.

The advice provided in this document is only meant to guide an operator in configuring forwarding devices, and is *not* to be interpreted as advice regarding default configuration settings for network devices. That is, this document provides advice with respect to operational configurations, but does not change the implementation defaults required by [RFC7045].

We recommend that configuration options are made available to govern the processing of each IPv6 EH type and each IPv6 option type. Such configuration options may include the following possible settings:

- Permit this IPv6 EH or IPv6 Option type

- Discard (and log) packets containing this IPv6 EH or option type
Reject (and log) packets containing this IPv6 EH or option type (where the packet drop is signaled with an ICMPv6 error message)

Rate-limit traffic containing this IPv6 EH or option type

Ignore this IPv6 EH or option type (as if it was not present) and forward the packet. We note that if a packet carries forwarding information (e.g., in an IPv6 Routing Header) this might be an inappropriate or undesirable action.

We note that special care needs to be taken when devices log packet drops/rejects. Devices should count the number of packets dropped/rejected, but the logging of drop/reject events should be limited so as to not overburden device resources.

Finally, we note that when discarding packets, it is generally desirable that the sender be signaled of the packet drop, since this is of use for trouble-shooting purposes. However, throughout this document (when recommending that packets be discarded) we generically refer to the action as "discard" without specifying whether the sender is signaled of the packet drop.

3. IPv6 Extension Headers

3.1. General Discussion

IPv6 [RFC2460] EHs allow for the extension of the IPv6 protocol. Since both IPv6 EHs and upper-layer protocols share the same namespace ("Next Header" registry/namespace), [RFC7045] identifies which of the currently assigned Internet Protocol numbers identify IPv6 EHs vs. upper-layer protocols. This document discusses the filtering of packets based on the IPv6 EHs (as specified by [RFC7045]) they contain.

NOTE: [RFC7112] specifies that non-fragmented IPv6 datagrams and IPv6 First-Fragments MUST contain the entire IPv6 header chain [RFC7112]. Therefore, intermediate systems can enforce the filtering policies discussed in this document, or resort to simply discarding the offending packets when they fail to comply with the requirements in [RFC7112]. We note that, in order to implement filtering rules on the fast path, it may be necessary for the filtering device to limit the depth into the packet that can be inspected before giving up. In circumstances where there is such a limitation, it is recommended that implementations discard packets if, when trying to determine whether to discard or permit a packet, the aforementioned limit is encountered.
3.2. General Security Implications

In some specific device architectures, IPv6 packets that contain IPv6 EHs may cause the corresponding packets to be processed on the slow path, and hence may be leveraged for the purpose of Denial of Service (DoS) attacks [I-D.gont-v6ops-ipv6-ehs-packet-drops] [Cisco-EH] [FW-Benchmark].

Operators are urged to consider IPv6 EH filtering and IPv6 options handling capabilities of different devices as they make deployment decisions in future.

3.3. Summary of Advice on the Handling of IPv6 Packets with Specific IPv6 Extension Headers

This section summarizes the advice provided in Section 3.4, providing references to the specific sections in which a detailed analysis can be found.
### Table 1: Summary of Advice on the Handling of IPv6 Packets with Specific IPv6 Extension Headers

<table>
<thead>
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<th>EH type</th>
<th>Filtering policy</th>
<th>Reference</th>
</tr>
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<td>Section 3.4.1</td>
</tr>
<tr>
<td>Routing Header for IPv6 (Proto=43)</td>
<td>Drop only RTH0, Permit other RH Types</td>
<td>Section 3.4.2</td>
</tr>
<tr>
<td>Fragment Header for IPv6 (Proto=44)</td>
<td>Permit</td>
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<tr>
<td>Encapsulating Security Payload (Proto=50)</td>
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<td>Use for experimentation and testing (Proto=253 and 254)</td>
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</tr>
</tbody>
</table>

3.4. Advice on the Handling of IPv6 Packets with Specific IPv6 Extension Headers

3.4.1. IPv6 Hop-by-Hop Options (Protocol Number=0)

3.4.1.1. Uses

The Hop-by-Hop Options header is used to carry optional information that should be examined by every node along a packet’s delivery path.
3.4.1.2. Specification

This EH is specified in [RFC2460], and its processing rules have been updated by [RFC7045]. At the time of this writing, the following options have been specified for the Hop-by-Hop Options EH:

- Type 0x00: Pad1 [RFC2460]
- Type 0x01: PadN [RFC2460]
- Type 0x05: Router Alert [RFC2711]
- Type 0x07: CALIPSO [RFC5570]
- Type 0x08: SMF_DPD [RFC6621]
- Type 0x26: Quick-Start [RFC4782]
- Type 0x4D: (Deprecated)
- Type 0x63: RPL Option [RFC6553]
- Type 0x6D: MPL Option [RFC7731]
- Type 0x8A: Endpoint Identification (Deprecated) [draft-ietf-nimrod-eid]
- Type 0xC2: Jumbo Payload [RFC2675]
- Type 0xEE: IPv6 DFF Header [RFC6971]
- Type 0x1E: RFC3692-style Experiment [RFC4727]
- Type 0x3E: RFC3692-style Experiment [RFC4727]
- Type 0x5E: RFC3692-style Experiment [RFC4727]
- Type 0x7E: RFC3692-style Experiment [RFC4727]
- Type 0x9E: RFC3692-style Experiment [RFC4727]
- Type 0xBE: RFC3692-style Experiment [RFC4727]
- Type 0xDE: RFC3692-style Experiment [RFC4727]
- Type 0xFE: RFC3692-style Experiment [RFC4727]
3.4.1.3. Specific Security Implications

Since this EH is required to be processed by all intermediate-systems en route, it can be leveraged to perform Denial of Service attacks against the network infrastructure.

NOTE: Ongoing work essentially aims at requiring the Hop-by-Hop Option EH to be processed only in cases where the intermediate node is making use of any functionality provided by such header (see [I-D.ietf-6man-hbh-header-handling]). However, the deployed base is likely to reflect the traditional behavior for a while, and hence the potential security problems of this EH are still of concern.

3.4.1.4. Operational and Interoperability Impact if Blocked

Discarding packets containing a Hop-by-Hop Options EH would break any of the protocols that rely on it for proper functioning. For example, it would break RSVP [RFC2205] and multicast deployments, and would cause IPv6 jumbograms to be discarded.

3.4.1.5. Advice

The recommended configuration for the processing of these packets depends on the features and capabilities of the underlying platform. On platforms that allow forwarding of packets with HBH Options on the fast path, we recommend that packets with a HBH Options EH be forwarded as normal (for instance, [RFC7045] allows for implementations to ignore the HBH Options EH when forwarding packets). Otherwise, on platforms in which processing of packets with a IPv6 HBH Options EH is carried out in the slow path, and an option is provided to rate-limit these packets, we recommend that this option be selected. Finally, when packets containing a HBH Options EH are processed in the slow-path, and the underlying platform does not have any mitigation options available for attacks based on these packets, we recommend that such platforms discard packets containing IPv6 HBH Options EHs.

Finally, we note that, for obvious reasons, RPL (Routing Protocol for Low-Power and Lossy Networks) [RFC6550] routers must not discard packets based on the presence of an IPv6 Hop-by-Hop Options EH.

3.4.2. Routing Header for IPv6 (Protocol Number=43)

3.4.2.1. Uses

The Routing header is used by an IPv6 source to list one or more intermediate nodes to be "visited" on the way to a packet’s destination.
3.4.2.2. Specification

This EH is specified in [RFC2460]. [RFC2460] originally specified the Routing Header Type 0, which has been later obsoleted by [RFC5095].

At the time of this writing, the following Routing Types have been specified:

- Type 0: Source Route (DEPRECATED) [RFC2460] [RFC5095]
- Type 1: Nimrod (DEPRECATED)
- Type 2: Type 2 Routing Header [RFC6275]
- Type 3: RPL Source Route Header [RFC6554]
- Types 4-252: Unassigned
- Type 253: RFC3692-style Experiment 1 [RFC4727]
- Type 254: RFC3692-style Experiment 2 [RFC4727]
- Type 255: Reserved

3.4.2.3. Specific Security Implications

The security implications of RHT0 have been discussed in detail in [Biondi2007] and [RFC5095].

3.4.2.4. Operational and Interoperability Impact if Blocked

Blocking packets containing a RHT0 or RTH1 has no operational implications. However, blocking packets employing other routing header types will break the protocols that rely on them.

3.4.2.5. Advice

Intermediate systems should discard packets containing a RHT0 or RHT1. RHT2 and RHT3 should be permitted, as required by [RFC7045]. Other routing header types should be discarded.

3.4.3. Fragment Header for IPv6 (Protocol Number=44)
3.4.3.1. Uses

This EH provides the fragmentation functionality for IPv6.

3.4.3.2. Specification

This EH is specified in [RFC2460].

3.4.3.3. Specific Security Implications

The security implications of the Fragment Header range from Denial of Service attacks (e.g. based on flooding a target with IPv6 fragments) to information leakage attacks [RFC7739].

3.4.3.4. Operational and Interoperability Impact if Blocked

Blocking packets that contain a Fragment Header will break any protocol that may rely on fragmentation (e.g., the DNS [RFC1034]).

3.4.3.5. Advice

Intermediate systems should permit packets that contain a Fragment Header.

3.4.4. Encapsulating Security Payload (Protocol Number=50)

3.4.4.1. Uses

This EH is employed for the IPsec suite [RFC4303].

3.4.4.2. Specification

This EH is specified in [RFC4303].

3.4.4.3. Specific Security Implications

Besides the general implications of IPv6 EHs, this EH could be employed to potentially perform a DoS attack at the destination system by wasting CPU resources in validating the contents of the packet.

3.4.4.4. Operational and Interoperability Impact if Blocked

Discarding packets that employ this EH would break IPsec deployments.
3.4.4.5. Advice

Intermediate systems should permit packets containing the Encapsulating Security Payload EH.

3.4.5. Authentication Header (Protocol Number=51)

3.4.5.1. Uses

The Authentication Header can be employed for provide authentication services in IPv4 and IPv6.

3.4.5.2. Specification

This EH is specified in [RFC4302].

3.4.5.3. Specific Security Implications

Besides the general implications of IPv6 EHs, this EH could be employed to potentially perform a DoS attack at the destination system by wasting CPU resources in validating the contents of the packet.

3.4.5.4. Operational and Interoperability Impact if Blocked

Discarding packets that employ this EH would break IPsec deployments.

3.4.5.5. Advice

Intermediate systems should permit packets containing an Authentication Header.

3.4.6. Destination Options for IPv6 (Protocol Number=60)

3.4.6.1. Uses

The Destination Options header is used to carry optional information that needs be examined only by a packet's destination node(s).

3.4.6.2. Specification

This EH is specified in [RFC2460]. At the time of this writing, the following options have been specified for this EH:

- Type 0x00: Pad1 [RFC2460]
- Type 0x01: PadN [RFC2460]
3.4.6.3. Specific Security Implications

No security implications are known, other than the general implications of IPv6 EHs. For a discussion of possible security implications of specific options specified for the DO header, please see the Section 4.3.

3.4.6.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain a Destination Options header would break protocols that rely on this EH type for conveying information, including protocols such as ILNP [RFC6740] and Mobile IPv6 [RFC6275], and IPv6 tunnels that employ the Tunnel Encapsulation Limit option.

3.4.6.5. Advice

Intermediate systems should permit packets that contain a Destination Options Header.
3.4.7. Mobility Header (Protocol Number=135)

3.4.7.1. Uses

The Mobility Header is an EH used by mobile nodes, correspondent nodes, and home agents in all messaging related to the creation and management of bindings in Mobile IPv6.

3.4.7.2. Specification

This EH is specified in [RFC6275].

3.4.7.3. Specific Security Implications

A thorough security assessment of the security implications of the Mobility Header and related mechanisms can be found in Section 15 of [RFC6275].

3.4.7.4. Operational and Interoperability Impact if Blocked

Discarding packets containing this EH would break Mobile IPv6.

3.4.7.5. Advice

Intermediate systems should permit packets containing this EH.

3.4.8. Host Identity Protocol (Protocol Number=139)

3.4.8.1. Uses

This EH is employed with the Host Identity Protocol (HIP), an experimental protocol that allows consenting hosts to securely establish and maintain shared IP-layer state, allowing separation of the identifier and locator roles of IP addresses, thereby enabling continuity of communications across IP address changes.

3.4.8.2. Specification

This EH is specified in [RFC5201].

3.4.8.3. Specific Security Implications

The security implications of the HIP header are discussed in detail in Section 8 of [RFC6275].
3.4.8.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain the Host Identity Protocol would break HIP deployments.

3.4.8.5. Advice

Intermediate systems should permit packets that contain a Host Identity Protocol EH.

3.4.9. Shim6 Protocol (Protocol Number=140)

3.4.9.1. Uses

This EH is employed by the Shim6 [RFC5533] Protocol.

3.4.9.2. Specification

This EH is specified in [RFC5533].

3.4.9.3. Specific Security Implications

The specific security implications are discussed in detail in Section 16 of [RFC5533].

3.4.9.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain this EH will break Shim6.

3.4.9.5. Advice

Intermediate systems should permit packets containing this EH.

3.4.10. Use for experimentation and testing (Protocol Numbers=253 and 254)

3.4.10.1. Uses

These IPv6 EHs are employed for performing RFC3692-Style experiments (see [RFC3692] for details).

3.4.10.2. Specification

These EHs are specified in [RFC3692] and [RFC4727].
3.4.10.3. Specific Security Implications

The security implications of these EHs will depend on their specific use.

3.4.10.4. Operational and Interoperability Impact if Blocked

For obvious reasons, discarding packets that contain these EHs limits the ability to perform legitimate experiments across IPv6 routers.

3.4.10.5. Advice

Intermediate systems should discard packets containing these EHs. Only in specific scenarios in which RFC3692-Style experiments are to be performed should these EHs be permitted.

3.5. Advice on the Handling of Packets with Unknown IPv6 Extension Headers

We refer to IPv6 EHs that have not been assigned an Internet Protocol Number by IANA (and marked as such) in [IANA-PROTOCOLS] as "unknown IPv6 extension headers" ("unknown IPv6 EHs").

3.5.1. Uses

New IPv6 EHs may be specified as part of future extensions to the IPv6 protocol.

Since IPv6 EHs and Upper-layer protocols employ the same namespace, it is impossible to tell whether an unknown "Internet Protocol Number" is being employed for an IPv6 EH or an Upper-Layer protocol.

3.5.2. Specification

The processing of unknown IPv6 EHs is specified in [RFC2460] and [RFC7045].

3.5.3. Specific Security Implications

For obvious reasons, it is impossible to determine specific security implications of unknown IPv6 EHs. However, from security standpoint, a device should discard IPv6 extension headers for which the security implications cannot be determined. We note that this policy is allowed by [RFC7045].
3.5.4. Operational and Interoperability Impact if Blocked

As noted in [RFC7045], discarding unknown IPv6 EHs may slow down the deployment of new IPv6 EHs and transport protocols. The corresponding IANA registry ([IANA-PROTOCOLS]) should be monitored such that filtering rules are updated as new IPv6 EHs are standardized.

We note that since IPv6 EHs and upper-layer protocols share the same numbering space, discarding unknown IPv6 EHs may result in packets encapsulating unknown upper-layer protocols being discarded.

3.5.5. Advice

Intermediate systems should discard packets containing unknown IPv6 EHs.

4. IPv6 Options

4.1. General Discussion

The following subsections describe specific security implications of different IPv6 options, and provide advice regarding filtering packets that contain such options.

4.2. General Security Implications of IPv6 Options

The general security implications of IPv6 options are closely related to those discussed in Section 3.2 for IPv6 EHs. Essentially, packets that contain IPv6 options might need to be processed by an IPv6 router’s general-purpose CPU, and hence could present a DDoS risk to that router’s general-purpose CPU (and thus to the router itself). For some architectures, a possible mitigation would be to rate-limit the packets that are to be processed by the general-purpose CPU (see e.g. [Cisco-EH]).

4.3. Advice on the Handling of Packets with Specific IPv6 Options

The following subsections contain a description of each of the IPv6 options that have so far been specified, a summary of the security implications of each of such options, a discussion of possible interoperability implications if packets containing such options are discarded, and specific advice regarding whether packets containing these options should be permitted.
4.3.1. Pad1 (Type=0x00)

4.3.1.1. Uses

This option is used when necessary to align subsequent options and to pad out the containing header to a multiple of 8 octets in length.

4.3.1.2. Specification

This option is specified in [RFC2460].

4.3.1.3. Specific Security Implications

None.

4.3.1.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain this option would potentially break any protocol that relies on IPv6 EHs.

4.3.1.5. Advice

Intermediate systems should not discard packets based on the presence of this option.

4.3.2. PadN (Type=0x01)

4.3.2.1. Uses

This option is used when necessary to align subsequent options and to pad out the containing header to a multiple of 8 octets in length.

4.3.2.2. Specification

This option is specified in [RFC2460].

4.3.2.3. Specific Security Implications

Because of the possible size of this option, it could be leveraged as a large-bandwidth covert channel.

4.3.2.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain this option would potentially break any protocol that relies on IPv6 EHs.
4.3.2.5. Advice

Intermediate systems should not discard IPv6 packets based on the presence of this option.

4.3.3. Jumbo Payload (Type=0XC2)

4.3.3.1. Uses

The Jumbo payload option provides the means of specifying payloads larger than 65535 bytes.

4.3.3.2. Specification

This option is specified in [RFC2675].

4.3.3.3. Specific Security Implications

There are no specific issues arising from this option, except for improper validity checks of the option and associated packet lengths.

4.3.3.4. Operational and Interoperability Impact if Blocked

Discarding packets based on the presence of this option will cause IPv6 jumbograms to be discarded.

4.3.3.5. Advice

Intermediate systems should discard packets that contain this option. An operator should permit this option only in specific scenarios in which support for IPv6 jumbograms is desired.

4.3.4. RPL Option (Type=0x63)

4.3.4.1. Uses

The RPL Option provides a mechanism to include routing information with each datagram that an RPL router forwards.

4.3.4.2. Specification

This option is specified in [RFC6553].

4.3.4.3. Specific Security Implications

Those described in [RFC6553].
4.3.4.4. Operational and Interoperability Impact if Blocked

This option is meant to be employed within an RPL instance. As a result, discarding packets based on the presence of this option (e.g. at an ISP) will not result in interoperability implications.

4.3.4.5. Advice

Non-RPL routers should discard packets that contain an RPL option.

4.3.5. Tunnel Encapsulation Limit (Type=0x04)

4.3.5.1. Uses

The Tunnel Encapsulation Limit option can be employed to specify how many further levels of nesting the packet is permitted to undergo.

4.3.5.2. Specification

This option is specified in [RFC2473].

4.3.5.3. Specific Security Implications

Those described in [RFC2473].

4.3.5.4. Operational and Interoperability Impact if Blocked

Discarding packets based on the presence of this option could result in tunnel traffic being discarded.

4.3.5.5. Advice

Intermediate systems should not discard packets based on the presence of this option.

4.3.6. Router Alert (Type=0x05)

4.3.6.1. Uses

The Router Alert option [RFC2711] is typically employed for the RSVP protocol [RFC2205] and the MLD protocol [RFC2710].

4.3.6.2. Specification

This option is specified in [RFC2711].
4.3.6.3. Specific Security Implications

Since this option causes the contents of the packet to be inspected by the handling device, this option could be leveraged for performing DoS attacks.

4.3.6.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain this option would break RSVP and multicast deployments.

4.3.6.5. Advice

Intermediate systems should discard packets that contain this option. Only in specific environments where support for RSVP, multicast routing, or similar protocols is desired, should this option be permitted.

4.3.7. Quick-Start (Type=0x26)

4.3.7.1. Uses

This IP Option is used in the specification of Quick-Start for TCP and IP, which is an experimental mechanism that allows transport protocols, in cooperation with routers, to determine an allowed sending rate at the start and, at times, in the middle of a data transfer (e.g., after an idle period) [RFC4782].

4.3.7.2. Specification

This option is specified in [RFC4782], on the "Experimental" track.

4.3.7.3. Specific Security Implications

Section 9.6 of [RFC4782] notes that Quick-Start is vulnerable to two kinds of attacks:

- attacks to increase the routers’ processing and state load, and,

- attacks with bogus Quick-Start Requests to temporarily tie up available Quick-Start bandwidth, preventing routers from approving Quick-Start Requests from other connections.

We note that if routers in a given environment do not implement and enable the Quick-Start mechanism, only the general security implications of IP options (discussed in Section 4.2) would apply.
4.3.7.4. Operational and Interoperability Impact if Blocked

The Quick-Start functionality would be disabled, and additional delays in TCP’s connection establishment (for example) could be introduced. (Please see Section 4.7.2 of [RFC4782].) We note, however, that Quick-Start has been proposed as a mechanism that could be of use in controlled environments, and not as a mechanism that would be intended or appropriate for ubiquitous deployment in the global Internet [RFC4782].

4.3.7.5. Advice

Intermediate systems should not discard IPv6 packets based on the presence of this option.

4.3.8. CALIPSO (Type=0x07)

4.3.8.1. Uses

This option is used for encoding explicit packet Sensitivity Labels on IPv6 packets. It is intended for use only within Multi-Level Secure (MLS) networking environments that are both trusted and trustworthy.

4.3.8.2. Specification

This option is specified in [RFC5570].

4.3.8.3. Specific Security Implications

Presence of this option in a packet does not by itself create any specific new threat. Packets with this option ought not normally be seen on the global public Internet.

4.3.8.4. Operational and Interoperability Impact if Blocked

If packets with this option are discarded or if the option is stripped from the packet during transmission from source to destination, then the packet itself is likely to be discarded by the receiver because it is not properly labeled. In some cases, the receiver might receive the packet but associate an incorrect sensitivity label with the received data from the packet whose CALIPSO was stripped by an intermediate router or firewall. Associating an incorrect sensitivity label can cause the received information either to be handled as more sensitive than it really is ("upgrading") or as less sensitive than it really is ("downgrading"), either of which is problematic.
4.3.8.5. Advice

Intermediate systems that do not operate in Multi-Level Secure (MLS) networking environments should discard packets that contain this option.

4.3.9. SMF_DPD (Type=0x08)

4.3.9.1. Uses

This option is employed in the (experimental) Simplified Multicast Forwarding (SMF) for unique packet identification for IPv6 I-DPD, and as a mechanism to guarantee non-collision of hash values for different packets when H-DPD is used.

4.3.9.2. Specification

This option is specified in [RFC6621].

4.3.9.3. Specific Security Implications

None. The use of identifiers is subject to the security and privacy considerations discussed in [I-D.gont-predictable-numeric-ids].

4.3.9.4. Operational and Interoperability Impact if Blocked

Dropping packets containing this option within a MANET domain would break SMF. However, dropping such packets at the border of such domain would have no negative impact.

4.3.9.5. Advice

Intermediate system should discard packets that contain this option.

4.3.10. Home Address (Type=0xC9)

4.3.10.1. Uses

The Home Address option is used by a Mobile IPv6 node while away from home, to inform the recipient of the mobile node’s home address.

4.3.10.2. Specification

This option is specified in [RFC6275].
4.3.10.3. Specific Security Implications

No (known) additional security implications than those described in [RFC6275].

4.3.10.4. Operational and Interoperability Impact if Blocked

Discarding IPv6 packets based on the presence of this option will break Mobile IPv6.

4.3.10.5. Advice

Intermediate systems should not discard IPv6 packets based on the presence of this option.

4.3.11. Endpoint Identification (Type=0x8A)

4.3.11.1. Uses

The Endpoint Identification option was meant to be used with the Nimrod routing architecture [NIMROD-DOC], but has never seen widespread deployment.

4.3.11.2. Specification

This option is specified in [NIMROD-DOC].

4.3.11.3. Specific Security Implications

Undetermined.

4.3.11.4. Operational and Interoperability Impact if Blocked

None.

4.3.11.5. Advice

Intermediate systems should discard packets that contain this option.

4.3.12. ILNP Nonce (Type=0x8B)

4.3.12.1. Uses

This option is employed by Identifier-Locator Network Protocol for IPv6 (ILNPv6) for providing protection against off-path attacks for packets when ILNPv6 is in use, and as a signal during initial network-layer session creation that ILNPv6 is proposed for use with this network-layer session, rather than classic IPv6.
4.3.12.2. Specification

This option is specified in [RFC6744].

4.3.12.3. Specific Security Implications

Those described in [RFC6744].

4.3.12.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain this option will break INLPv6 deployments.

4.3.12.5. Advice

Intermediate systems should not discard packets based on the presence of this option.

4.3.13. Line-Identification Option (Type=0x8C)

4.3.13.1. Uses

This option is used by an Edge Router to identify the subscriber premises in scenarios where several subscriber premises may be logically connected to the same interface of an Edge Router.

4.3.13.2. Specification

This option is specified in [RFC6788].

4.3.13.3. Specific Security Implications

Those described in [RFC6788].

4.3.13.4. Operational and Interoperability Impact if Blocked

Since this option is meant to be employed in Router Solicitation messages, discarding packets based on the presence of this option at intermediate systems will result in no interoperability implications.

4.3.13.5. Advice

Intermediate devices should discard packets that contain this option.
4.3.14. Deprecated (Type=0x4D)

4.3.14.1. Uses

No information has been found about this option type.

4.3.14.2. Specification

No information has been found about this option type.

4.3.14.3. Specific Security Implications

No information has been found about this option type, and hence it has been impossible to perform the corresponding security assessment.

4.3.14.4. Operational and Interoperability Impact if Blocked

Unknown.

4.3.14.5. Advice

Intermediate systems should discard packets that contain this option.

4.3.15. MPL Option (Type=0x6D)

4.3.15.1. Uses

This option is used with the Multicast Protocol for Low power and Lossy Networks (MPL), that provides IPv6 multicast forwarding in constrained networks.

4.3.15.2. Specification

This option is specified in [RFC7731], and is meant to be included only in Hop-by-Hop Option headers.

4.3.15.3. Specific Security Implications

Those described in [RFC7731].

4.3.15.4. Operational and Interoperability Impact if Blocked

Dropping packets that contain an MPL option within an MPL network would break the Multicast Protocol for Low power and Lossy Networks (MPL). However, dropping such packets at the border of such networks will have no negative impact.
4.3.15.5. Advice

Intermediate systems should not discard packets based on the presence of this option. However, since this option has been specified for the Hop-by-Hop Options, such systems should consider the discussion in Section 3.4.1.

4.3.16. IP_DFF (Type=0xEE)

4.3.16.1. Uses

This option is employed with the (Experimental) Depth-First Forwarding (DFF) in Unreliable Networks.

4.3.16.2. Specification

This option is specified in [RFC6971].

4.3.16.3. Specific Security Implications

Those specified in [RFC6971].

4.3.16.4. Operational and Interoperability Impact if Blocked

Dropping packets containing this option within a routing domain that is running DFF would break DFF. However, dropping such packets at the border of such domains will have no security implications.

4.3.16.5. Advice

Intermediate systems that do not operate within a routing domain that is running DFF should discard packets containing this option.

4.3.17. RFC3692-style Experiment (Types = 0x1E, 0x3E, 0x5E, 0x7E, 0x9E, 0xB6, 0xDE, 0xFE)

4.3.17.1. Uses

These options can be employed for performing RFC3692-style experiments. It is only appropriate to use these values in explicitly configured experiments; they must not be shipped as defaults in implementations.

4.3.17.2. Specification

Specified in RFC 4727 [RFC4727] in the context of RFC3692-style experiments.
4.3.17.3. Specific Security Implications

The specific security implications will depend on the specific use of these options.

4.3.17.4. Operational and Interoperability Impact if Blocked

For obvious reasons, discarding packets that contain these options limits the ability to perform legitimate experiments across IPv6 routers.

4.3.17.5. Advice

Intermediate systems should discard packets that contain these options. Only in specific environments where RFC3692-style experiments are meant to be performed should these options be permitted.

4.4. Advice on the handling of Packets with Unknown IPv6 Options

We refer to IPv6 options that have not been assigned an IPv6 option type in the corresponding registry ([IANA-IPV6-PARAM]) as "unknown IPv6 options".

4.4.1. Uses

New IPv6 options may be specified as part of future protocol work.

4.4.2. Specification

The processing of unknown IPv6 options is specified in [RFC2460].

4.4.3. Specific Security Implications

For obvious reasons, it is impossible to determine specific security implications of unknown IPv6 options.

4.4.4. Operational and Interoperability Impact if Blocked

Discarding unknown IPv6 options may slow down the deployment of new IPv6 options. As noted in [draft-gont-6man-ipv6-opt-transmit], the corresponding IANA registry ([IANA-IPV6-PARAM]) should be monitored such that IPv6 option filtering rules are updated as new IPv6 options are standardized.
4.4.5. Advice

Enterprise intermediate systems that process the contents of IPv6 EHs should discard packets that contain unknown options. Other intermediate systems that process the contents of IPv6 EHs should permit packets that contain unknown options.

5. IANA Considerations

This document has no actions for IANA.

6. Security Considerations

This document provides advice on the filtering of IPv6 packets that contain IPv6 EHs (and possibly IPv6 options) at IPv6 transit routers. It is meant to improve the current situation of widespread dropping of such IPv6 packets in those cases where the drops result from improper configuration defaults, or inappropriate advice in this area.

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This document borrows some text an analysis from [RFC7126], authored by Fernando Gont, Randall Atkinson, and Carlos Pignataro.

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Recommendations on the Filtering of IPv6 Packets Containing IPv6 Extension Headers
draft-ietf-opsec-ipv6-eh-filtering-06

Abstract

It is common operator practice to mitigate security risks by enforcing appropriate packet filtering. This document analyzes both the general security implications of IPv6 Extension Headers and the specific security implications of each Extension Header and Option type. Additionally, it discusses the operational and interoperability implications of discarding packets based on the IPv6 Extension Headers and IPv6 options they contain. Finally, it provides advice on the filtering of such IPv6 packets at transit routers for traffic *not* directed to them, for those cases in which such filtering is deemed as necessary.

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Recent studies (see e.g. [RFC7872]) suggest that there is widespread dropping of IPv6 packets that contain IPv6 Extension Headers (EHs). In some cases, such packet drops occur at transit routers. While some operators "officially" drop packets that contain IPv6 EHs, it is possible that some of the measured packet drops be the result of improper configuration defaults, or inappropriate advice in this area.
This document analyzes both the general security implications of IPv6 EHs and the specific security implications of each EH and Option type, and provides advice on the filtering of IPv6 packets based on the IPv6 EHs and the IPv6 options they contain. Since various protocols may use IPv6 EHs (possibly with IPv6 options), discarding packets based on the IPv6 EHs or IPv6 options they contain may have implications on the proper functioning of such protocols. Thus, this document also attempts to discuss the operational and interoperability implications of such filtering policies.

The filtering policy typically depends on where in the network such policy is enforced: when the policy is enforced in a transit network, the policy typically follows a "black-list" approach, where only packets with clear negative implications are dropped. On the other hand, when the policy is enforced closer to the destination systems, the policy typically follows a "white-list" approach, where only traffic that is expected to be received is allowed. The advice in this document is aimed only at transit routers that may need to enforce a filtering policy based on the EHs and IPv6 options a packet may contain, following a "black-list" approach, and hence is likely to be much more permissive that a filtering policy to be employed e.g. at the edge of an enterprise network. The advice in this document is meant to improve the current situation of the dropping of packets with IPv6 EHs in the Internet [RFC7872].

This document is similar in nature to [RFC7126], which addresses the same problem for the IPv4 case. However, in IPv6, the problem space is compounded by the fact that IPv6 specifies a number of IPv6 EHs, and a number of IPv6 options which may be valid only when included in specific EH types.

This document completes and complements the considerations for protecting the control plane from packets containing IP options that can be found in [RFC6192].

Section 2 of this document specifies the terminology and conventions employed throughout this document. Section 3 of this document discusses IPv6 EHs and provides advice in the area of filtering IPv6 packets that contain such IPv6 EHs. Section 4 of this document discusses IPv6 options and provides advice in the area of filtering IPv6 packets that contain such options.

2. Terminology and Conventions Used in This Document
2.1. Terminology

The terms "fast path", "slow path", and associated relative terms ("faster path" and "slower path") are loosely defined as in Section 2 of [RFC6398].

The terms "permit" (allow the traffic), "drop" (drop with no notification to sender), and "reject" (drop with appropriate notification to sender) are employed as defined in [RFC3871]. Throughout this document we also employ the term "discard" as a generic term to indicate the act of discarding a packet, irrespective of whether the sender is notified of such drops, and irrespective of whether the specific filtering action is logged.

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in [RFC2119].

2.2. Applicability Statement

This document provides advice on the filtering of IPv6 packets with EHs at transit routers for traffic *not* explicitly destined to such transit routers, for those cases in which such filtering is deemed as necessary.

2.3. Conventions

This document assumes that nodes comply with the requirements in [RFC7045]. Namely (from [RFC7045]),

- If a forwarding node discards a packet containing a standard IPv6 EH, it MUST be the result of a configurable policy and not just the result of a failure to recognise such a header.
- The discard policy for each standard type of EH MUST be individually configurable.
- The default configuration SHOULD allow all standard IPv6 EHs.

The advice provided in this document is only meant to guide an operator in configuring forwarding devices, and is *not* to be interpreted as advice regarding default configuration settings for network devices. That is, this document provides advice with respect to operational configurations, but does not change the implementation defaults required by [RFC7045].
We recommend that configuration options are made available to govern the processing of each IPv6 EH type and each IPv6 option type. Such configuration options may include the following possible settings:

- Permit this IPv6 EH or IPv6 Option type
- Discard (and log) packets containing this IPv6 EH or option type
- Reject (and log) packets containing this IPv6 EH or option type (where the packet drop is signaled with an ICMPv6 error message)
- Rate-limit traffic containing this IPv6 EH or option type
- Ignore this IPv6 EH or option type (as if it was not present) and forward the packet. We note that if a packet carries forwarding information (e.g., in an IPv6 Routing Header) this might be an inappropriate or undesirable action.

We note that special care needs to be taken when devices log packet drops/rejects. Devices should count the number of packets dropped/rejected, but the logging of drop/reject events should be limited so as to not overburden device resources.

Finally, we note that when discarding packets, it is generally desirable that the sender be signaled of the packet drop, since this is of use for trouble-shooting purposes. However, throughout this document (when recommending that packets be discarded) we generically refer to the action as "discard" without specifying whether the sender is signaled of the packet drop.

3. IPv6 Extension Headers

3.1. General Discussion

IPv6 [RFC8200] EHs allow for the extension of the IPv6 protocol. Since both IPv6 EHs and upper-layer protocols share the same namespace ("Next Header" registry/namespace), [RFC7045] identifies which of the currently assigned Internet Protocol numbers identify IPv6 EHs vs. upper-layer protocols. This document discusses the filtering of packets based on the IPv6 EHs (as specified by [RFC7045]) they contain.

NOTE: [RFC7112] specifies that non-fragmented IPv6 datagrams and IPv6 First-Fragments MUST contain the entire IPv6 header chain [RFC7112]. Therefore, intermediate systems can enforce the filtering policies discussed in this document, or resort to simply discarding the offending packets when they fail to comply with the requirements in [RFC7112]. We note that, in order to implement
filtering rules on the fast path, it may be necessary for the filtering device to limit the depth into the packet that can be inspected before giving up. In circumstances where there is such a limitation, it is recommended that implementations discard packets if, when trying to determine whether to discard or permit a packet, the aforementioned limit is encountered.

3.2. General Security Implications

In some specific device architectures, IPv6 packets that contain IPv6 EHs may cause the corresponding packets to be processed on the slow path, and hence may be leveraged for the purpose of Denial of Service (DoS) attacks [I-D.gont-v6ops-ipv6-ehs-packet-drops] [Cisco-EH] [FW-Benchmark].

Operators are urged to consider IPv6 EH filtering and IPv6 options handling capabilities of different devices as they make deployment decisions in future.

3.3. Summary of Advice on the Handling of IPv6 Packets with Specific IPv6 Extension Headers

This section summarizes the advice provided in Section 3.4, providing references to the specific sections in which a detailed analysis can be found.
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Table 1: Summary of Advice on the Handling of IPv6 Packets with Specific IPv6 Extension Headers

3.4. Advice on the Handling of IPv6 Packets with Specific IPv6 Extension Headers

3.4.1. IPv6 Hop-by-Hop Options (Protocol Number=0)
3.4.1.1. Uses

The Hop-by-Hop Options header is used to carry optional information that may be examined by every node along a packet’s delivery path. It is expected that nodes will examine the Hop-by-Hop Options header if explicitly configured to do so.

NOTE: [RFC2460] required that all nodes examined and processed the Hop-by-Hop Options header. However, even before the publication of [RFC8200], a number of implementations already provided the option of ignoring this header unless explicitly configured to examine it.

3.4.1.2. Specification

This EH is specified in [RFC8200]. At the time of this writing, the following options have been specified for the Hop-by-Hop Options EH:

- Type 0x00: Pad1 [RFC8200]
- Type 0x01: PadN [RFC8200]
- Type 0x05: Router Alert [RFC2711]
- Type 0x07: CALIPSO [RFC5570]
- Type 0x08: SMF_DPD [RFC6621]
- Type 0x23: RPL Option [I-D.ietf-roll-useofrplinfo]
- Type 0x26: Quick-Start [RFC4782]
- Type 0x4D: (Deprecated)
- Type 0x63: RPL Option [RFC6553]
- Type 0x6D: MPL Option [RFC7731]
- Type 0x8A: Endpoint Identification (Deprecated) [draft-ietf-nimrod-eid]
- Type 0xC2: Jumbo Payload [RFC2675]
- Type 0xEE: IPv6 DFF Header [RFC6971]
- Type 0x1E: RFC3692-style Experiment [RFC4727]
- Type 0x3E: RFC3692-style Experiment [RFC4727]
3.4.1.3. Specific Security Implications

Legacy nodes that may process this extension header could be subject to Denial of Service attacks.

NOTE: While [RFC8200] has removed this requirement, the deployed base may still reflect the traditional behavior for a while, and hence the potential security problems of this EH are still of concern.

3.4.1.4. Operational and Interoperability Impact if Blocked

Discarding packets containing a Hop-by-Hop Options EH would break any of the protocols that rely on it for proper functioning. For example, it would break RSVP [RFC2205] and multicast deployments, and would cause IPv6 jumbograms to be discarded.

3.4.1.5. Advice

Nodes implementing [RFC8200] would already ignore this extension header unless explicitly required to process it. For legacy ([RFC2460] nodes, the recommended configuration for the processing of these packets depends on the features and capabilities of the underlying platform. On platforms that allow forwarding of packets with HBH Options on the fast path, we recommend that packets with a HBH Options EH be forwarded as normal. Otherwise, on platforms in which processing of packets with an IPv6 HBH Options EH is carried out in the slow path, and an option is provided to rate-limit these packets, we recommend that this option be selected. Finally, when packets containing a HBH Options EH are processed in the slow-path, and the underlying platform does not have any mitigation options available for attacks based on these packets, we recommend that such platforms discard packets containing IPv6 HBH Options EHS.

Finally, we note that, for obvious reasons, RPL (Routing Protocol for Low-Power and Lossy Networks) [RFC6550] routers must not discard packets based on the presence of an IPv6 Hop-by-Hop Options EH.
3.4.2. Routing Header for IPv6 (Protocol Number=43)

3.4.2.1. Uses

The Routing header is used by an IPv6 source to list one or more intermediate nodes to be "visited" on the way to a packet’s destination.

3.4.2.2. Specification

This EH is specified in [RFC8200]. [RFC2460] had originally specified the Routing Header Type 0, which was later obsoleted by [RFC5095], and thus removed from [RFC8200].

At the time of this writing, the following Routing Types have been specified:

- Type 0: Source Route (DEPRECATED) [RFC2460] [RFC5095]
- Type 1: Nimrod (DEPRECATED)
- Type 2: Type 2 Routing Header [RFC6275]
- Type 3: RPL Source Route Header [RFC6554]
- Types 4-252: Unassigned
- Type 253: RFC3692-style Experiment 1 [RFC4727]
- Type 254: RFC3692-style Experiment 2 [RFC4727]
- Type 255: Reserved

3.4.2.3. Specific Security Implications

The security implications of RHT0 have been discussed in detail in [Biondi2007] and [RFC5095].

3.4.2.4. Operational and Interoperability Impact if Blocked

Blocking packets containing a RHT0 or RTH1 has no operational implications, since both have been deprecated. However, blocking packets employing other routing header types will break the protocols that rely on them.
3.4.2.5. Advice

Intermediate systems should discard packets containing a RHT0 or RHT1. Other routing header types should be permitted, as required by [RFC7045].

3.4.3. Fragment Header for IPv6 (Protocol Number=44)

3.4.3.1. Uses

This EH provides the fragmentation functionality for IPv6.

3.4.3.2. Specification

This EH is specified in [RFC8200].

3.4.3.3. Specific Security Implications

The security implications of the Fragment Header range from Denial of Service attacks (e.g., based on flooding a target with IPv6 fragments) to information leakage attacks [RFC7739].

3.4.3.4. Operational and Interoperability Impact if Blocked

Blocking packets that contain a Fragment Header will break any protocol that may rely on fragmentation (e.g., the DNS [RFC1034]).

3.4.3.5. Advice

Intermediate systems should permit packets that contain a Fragment Header.

3.4.4. Encapsulating Security Payload (Protocol Number=50)

3.4.4.1. Uses

This EH is employed for the IPsec suite [RFC4303].

3.4.4.2. Specification

This EH is specified in [RFC4303].

3.4.4.3. Specific Security Implications

Besides the general implications of IPv6 EHS, this EH could be employed to potentially perform a DoS attack at the destination system by wasting CPU resources in validating the contents of the packet.
3.4.4.4. Operational and Interoperability Impact if Blocked

Discarding packets that employ this EH would break IPsec deployments.

3.4.4.5. Advice

Intermediate systems should permit packets containing the Encapsulating Security Payload EH.

3.4.5. Authentication Header (Protocol Number=51)

3.4.5.1. Uses

The Authentication Header can be employed for provide authentication services in IPv4 and IPv6.

3.4.5.2. Specification

This EH is specified in [RFC4302].

3.4.5.3. Specific Security Implications

Besides the general implications of IPv6 EHs, this EH could be employed to potentially perform a DoS attack at the destination system by wasting CPU resources in validating the contents of the packet.

3.4.5.4. Operational and Interoperability Impact if Blocked

Discarding packets that employ this EH would break IPsec deployments.

3.4.5.5. Advice

Intermediate systems should permit packets containing an Authentication Header.

3.4.6. Destination Options for IPv6 (Protocol Number=60)

3.4.6.1. Uses

The Destination Options header is used to carry optional information that needs be examined only by a packet’s destination node(s).

3.4.6.2. Specification

This EH is specified in [RFC8200]. At the time of this writing, the following options have been specified for this EH:
o Type 0x00: Pad1 [RFC8200]
o Type 0x01: PadN [RFC8200]
o Type 0x04: Tunnel Encapsulation Limit [RFC2473]
o Type 0x4D: (Deprecated)
o Type 0xC9: Home Address [RFC6275]
o Type 0x8A: Endpoint Identification (Deprecated) [draft-ietf-nimrod-eid]
o Type 0x8B: ILNP Nonce [RFC6744]
o Type 0x8C: Line-Identification Option [RFC6788]
o Type 0x1E: RFC3692-style Experiment [RFC4727]
o Type 0x3E: RFC3692-style Experiment [RFC4727]
o Type 0x5E: RFC3692-style Experiment [RFC4727]
o Type 0x7E: RFC3692-style Experiment [RFC4727]
o Type 0x9E: RFC3692-style Experiment [RFC4727]
o Type 0xBE: RFC3692-style Experiment [RFC4727]
o Type 0xDE: RFC3692-style Experiment [RFC4727]
o Type 0xFE: RFC3692-style Experiment [RFC4727]

3.4.6.3. Specific Security Implications

No security implications are known, other than the general implications of IPv6 EHs. For a discussion of possible security implications of specific options specified for the DO header, please see the Section 4.3.

3.4.6.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain a Destination Options header would break protocols that rely on this EH type for conveying information, including protocols such as ILNP [RFC6740] and Mobile IPv6 [RFC6275], and IPv6 tunnels that employ the Tunnel Encapsulation Limit option.
3.4.6.5. Advice

Intermediate systems should permit packets that contain a Destination Options Header.

3.4.7. Mobility Header (Protocol Number=135)

3.4.7.1. Uses

The Mobility Header is an EH used by mobile nodes, correspondent nodes, and home agents in all messaging related to the creation and management of bindings in Mobile IPv6.

3.4.7.2. Specification

This EH is specified in [RFC6275].

3.4.7.3. Specific Security Implications

A thorough security assessment of the security implications of the Mobility Header and related mechanisms can be found in Section 15 of [RFC6275].

3.4.7.4. Operational and Interoperability Impact if Blocked

Discarding packets containing this EH would break Mobile IPv6.

3.4.7.5. Advice

Intermediate systems should permit packets containing this EH.

3.4.8. Host Identity Protocol (Protocol Number=139)

3.4.8.1. Uses

This EH is employed with the Host Identity Protocol (HIP), an experimental protocol that allows consenting hosts to securely establish and maintain shared IP-layer state, allowing separation of the identifier and locator roles of IP addresses, thereby enabling continuity of communications across IP address changes.

3.4.8.2. Specification

This EH is specified in [RFC5201].
3.4.8.3. Specific Security Implications

The security implications of the HIP header are discussed in detail in Section 8 of [RFC6275].

3.4.8.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain the Host Identity Protocol would break HIP deployments.

3.4.8.5. Advice

Intermediate systems should permit packets that contain a Host Identity Protocol EH.

3.4.9. Shim6 Protocol (Protocol Number=140)

3.4.9.1. Uses

This EH is employed by the Shim6 [RFC5533] Protocol.

3.4.9.2. Specification

This EH is specified in [RFC5533].

3.4.9.3. Specific Security Implications

The specific security implications are discussed in detail in Section 16 of [RFC5533].

3.4.9.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain this EH will break Shim6.

3.4.9.5. Advice

Intermediate systems should permit packets containing this EH.

3.4.10. Use for experimentation and testing (Protocol Numbers=253 and 254)

3.4.10.1. Uses

These IPv6 EHs are employed for performing RFC3692-Style experiments (see [RFC3692] for details).
3.4.10.2. Specification

These EHs are specified in [RFC3692] and [RFC4727].

3.4.10.3. Specific Security Implications

The security implications of these EHs will depend on their specific use.

3.4.10.4. Operational and Interoperability Impact if Blocked

For obvious reasons, discarding packets that contain these EHs limits the ability to perform legitimate experiments across IPv6 routers.

3.4.10.5. Advice

Intermediate systems should discard packets containing these EHs. Only in specific scenarios in which RFC3692-Style experiments are to be performed should these EHs be permitted.

3.5. Advice on the Handling of Packets with Unknown IPv6 Extension Headers

We refer to IPv6 EHs that have not been assigned an Internet Protocol Number by IANA (and marked as such) in [IANA-PROTOCOLS] as "unknown IPv6 extension headers" ("unknown IPv6 EHs").

3.5.1. Uses

New IPv6 EHs may be specified as part of future extensions to the IPv6 protocol.

Since IPv6 EHs and Upper-layer protocols employ the same namespace, it is impossible to tell whether an unknown "Internet Protocol Number" is being employed for an IPv6 EH or an Upper-Layer protocol.

3.5.2. Specification

The processing of unknown IPv6 EHs is specified in [RFC8200] and [RFC7045].

3.5.3. Specific Security Implications

For obvious reasons, it is impossible to determine specific security implications of unknown IPv6 EHs. However, from security standpoint, a device should discard IPv6 extension headers for which the security implications cannot be determined. We note that this policy is allowed by [RFC7045].
3.5.4. Operational and Interoperability Impact if Blocked

As noted in [RFC7045], discarding unknown IPv6 EHs may slow down the deployment of new IPv6 EHs and transport protocols. The corresponding IANA registry ([IANA-PROTOCOLS]) should be monitored such that filtering rules are updated as new IPv6 EHs are standardized.

We note that since IPv6 EHs and upper-layer protocols share the same numbering space, discarding unknown IPv6 EHs may result in packets encapsulating unknown upper-layer protocols being discarded.

3.5.5. Advice

Intermediate systems should discard packets containing unknown IPv6 EHs.

4. IPv6 Options

4.1. General Discussion

The following subsections describe specific security implications of different IPv6 options, and provide advice regarding filtering packets that contain such options.

4.2. General Security Implications of IPv6 Options

The general security implications of IPv6 options are closely related to those discussed in Section 3.2 for IPv6 EHs. Essentially, packets that contain IPv6 options might need to be processed by an IPv6 router’s general-purpose CPU, and hence could present a DDoS risk to that router’s general-purpose CPU (and thus to the router itself). For some architectures, a possible mitigation would be to rate-limit the packets that are to be processed by the general-purpose CPU (see e.g. [Cisco-EH]).

4.3. Advice on the Handling of Packets with Specific IPv6 Options

The following subsections contain a description of each of the IPv6 options that have so far been specified, a summary of the security implications of each of such options, a discussion of possible interoperability implications if packets containing such options are discarded, and specific advice regarding whether packets containing these options should be permitted.
4.3.1. Pad1 (Type=0x00)

4.3.1.1. Uses

This option is used when necessary to align subsequent options and to pad out the containing header to a multiple of 8 octets in length.

4.3.1.2. Specification

This option is specified in [RFC8200].

4.3.1.3. Specific Security Implications

None.

4.3.1.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain this option would potentially break any protocol that relies on IPv6 EHs.

4.3.1.5. Advice

Intermediate systems should not discard packets based on the presence of this option.

4.3.2. PadN (Type=0x01)

4.3.2.1. Uses

This option is used when necessary to align subsequent options and to pad out the containing header to a multiple of 8 octets in length.

4.3.2.2. Specification

This option is specified in [RFC8200].

4.3.2.3. Specific Security Implications

Because of the possible size of this option, it could be leveraged as a large-bandwidth covert channel.

4.3.2.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain this option would potentially break any protocol that relies on IPv6 EHs.
4.3.2.5. Advice

Intermediate systems should not discard IPv6 packets based on the presence of this option.

4.3.3. Jumbo Payload (Type=0XC2)

4.3.3.1. Uses

The Jumbo payload option provides the means of specifying payloads larger than 65535 bytes.

4.3.3.2. Specification

This option is specified in [RFC2675].

4.3.3.3. Specific Security Implications

There are no specific issues arising from this option, except for improper validity checks of the option and associated packet lengths.

4.3.3.4. Operational and Interoperability Impact if Blocked

Discarding packets based on the presence of this option will cause IPv6 jumbograms to be discarded.

4.3.3.5. Advice

Intermediate systems should discard packets that contain this option. An operator should permit this option only in specific scenarios in which support for IPv6 jumbograms is desired.

4.3.4. RPL Option (Type=0x63)

4.3.4.1. Uses

The RPL Option provides a mechanism to include routing information with each datagram that an RPL router forwards.

4.3.4.2. Specification

This option was originally specified in [RFC6553]. It has been deprecated by [I-D.ietf-roll-useofrplinfo].
4.3.4.3. Specific Security Implications

Those described in [RFC6553].

4.3.4.4. Operational and Interoperability Impact if Blocked

This option is meant to be employed within an RPL instance. As a result, discarding packets based on the presence of this option (e.g. at an ISP) will not result in interoperability implications.

4.3.4.5. Advice

Non-RPL routers should discard packets that contain an RPL option.

4.3.5. RPL Option (Type=0x23)

4.3.5.1. Uses

The RPL Option provides a mechanism to include routing information with each datagram that an RPL router forwards.

4.3.5.2. Specification

This option is specified in [I-D.ietf-roll-useofrplinfo].

4.3.5.3. Specific Security Implications

Those described in [I-D.ietf-roll-useofrplinfo].

4.3.5.4. Operational and Interoperability Impact if Blocked

This option is meant to survive outside of an RPL instance. As a result, discarding packets based on the presence of this option would break some use cases for RPL (see [I-D.ietf-roll-useofrplinfo]).

4.3.5.5. Advice

Intermediate systems should not discard IPv6 packets based on the presence of this option.

4.3.6. Tunnel Encapsulation Limit (Type=0x04)

4.3.6.1. Uses

The Tunnel Encapsulation Limit option can be employed to specify how many further levels of nesting the packet is permitted to undergo.
4.3.6.2. Specification

This option is specified in [RFC2473].

4.3.6.3. Specific Security Implications

Those described in [RFC2473].

4.3.6.4. Operational and Interoperability Impact if Blocked

Discarding packets based on the presence of this option could result in tunnel traffic being discarded.

4.3.6.5. Advice

Intermediate systems should not discard packets based on the presence of this option.

4.3.7. Router Alert (Type=0x05)

4.3.7.1. Uses

The Router Alert option [RFC2711] is typically employed for the RSVP protocol [RFC2205] and the MLD protocol [RFC2710].

4.3.7.2. Specification

This option is specified in [RFC2711].

4.3.7.3. Specific Security Implications

Since this option causes the contents of the packet to be inspected by the handling device, this option could be leveraged for performing DoS attacks.

4.3.7.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain this option would break RSVP and multicast deployments.

4.3.7.5. Advice

Intermediate systems should discard packets that contain this option. Only in specific environments where support for RSVP, multicast routing, or similar protocols is desired, should this option be permitted.
4.3.8. Quick-Start (Type=0x26)

4.3.8.1. Uses

This IP Option is used in the specification of Quick-Start for TCP and IP, which is an experimental mechanism that allows transport protocols, in cooperation with routers, to determine an allowed sending rate at the start and, at times, in the middle of a data transfer (e.g., after an idle period) [RFC4782].

4.3.8.2. Specification

This option is specified in [RFC4782], on the "Experimental" track.

4.3.8.3. Specific Security Implications

Section 9.6 of [RFC4782] notes that Quick-Start is vulnerable to two kinds of attacks:

- attacks to increase the routers’ processing and state load, and,
- attacks with bogus Quick-Start Requests to temporarily tie up available Quick-Start bandwidth, preventing routers from approving Quick-Start Requests from other connections.

We note that if routers in a given environment do not implement and enable the Quick-Start mechanism, only the general security implications of IP options (discussed in Section 4.2) would apply.

4.3.8.4. Operational and Interoperability Impact if Blocked

The Quick-Start functionality would be disabled, and additional delays in TCP’s connection establishment (for example) could be introduced. (Please see Section 4.7.2 of [RFC4782].) We note, however, that Quick-Start has been proposed as a mechanism that could be of use in controlled environments, and not as a mechanism that would be intended or appropriate for ubiquitous deployment in the global Internet [RFC4782].

4.3.8.5. Advice

Intermediate systems should not discard IPv6 packets based on the presence of this option.
4.3.9. CALIPSO (Type=0x07)

4.3.9.1. Uses

This option is used for encoding explicit packet Sensitivity Labels on IPv6 packets. It is intended for use only within Multi-Level Secure (MLS) networking environments that are both trusted and trustworthy.

4.3.9.2. Specification

This option is specified in [RFC5570].

4.3.9.3. Specific Security Implications

Presence of this option in a packet does not by itself create any specific new threat. Packets with this option ought not normally be seen on the global public Internet.

4.3.9.4. Operational and Interoperability Impact if Blocked

If packets with this option are discarded or if the option is stripped from the packet during transmission from source to destination, then the packet itself is likely to be discarded by the receiver because it is not properly labeled. In some cases, the receiver might receive the packet but associate an incorrect sensitivity label with the received data from the packet whose CALIPSO was stripped by an intermediate router or firewall. Associating an incorrect sensitivity label can cause the received information either to be handled as more sensitive than it really is ("upgrading") or as less sensitive than it really is ("downgrading"), either of which is problematic.

4.3.9.5. Advice

Intermediate systems that do not operate in Multi-Level Secure (MLS) networking environments should discard packets that contain this option.

4.3.10. SMF_DPD (Type=0x08)

4.3.10.1. Uses

This option is employed in the (experimental) Simplified Multicast Forwarding (SMF) for unique packet identification for IPv6 I-DPD, and as a mechanism to guarantee non-collision of hash values for different packets when H-DPD is used.
4.3.10.2. Specification
This option is specified in [RFC6621].

4.3.10.3. Specific Security Implications
None. The use of identifiers is subject to the security and privacy considerations discussed in [I-D.gont-predictable-numeric-ids].

4.3.10.4. Operational and Interoperability Impact if Blocked
Dropping packets containing this option within a MANET domain would break SMF. However, dropping such packets at the border of such domain would have no negative impact.

4.3.10.5. Advice
Intermediate system should discard packets that contain this option.

4.3.11. Home Address (Type=0xC9)

4.3.11.1. Uses
The Home Address option is used by a Mobile IPv6 node while away from home, to inform the recipient of the mobile node’s home address.

4.3.11.2. Specification
This option is specified in [RFC6275].

4.3.11.3. Specific Security Implications
No (known) additional security implications than those described in [RFC6275].

4.3.11.4. Operational and Interoperability Impact if Blocked
Discarding IPv6 packets based on the presence of this option will break Mobile IPv6.

4.3.11.5. Advice
Intermediate systems should not discard IPv6 packets based on the presence of this option.
4.3.12. Endpoint Identification (Type=0x8A)

4.3.12.1. Uses

The Endpoint Identification option was meant to be used with the Nimrod routing architecture [NIMROD-DOC], but has never seen widespread deployment.

4.3.12.2. Specification

This option is specified in [NIMROD-DOC].

4.3.12.3. Specific Security Implications

Undetermined.

4.3.12.4. Operational and Interoperability Impact if Blocked

None.

4.3.12.5. Advice

Intermediate systems should discard packets that contain this option.

4.3.13. ILNP Nonce (Type=0x8B)

4.3.13.1. Uses

This option is employed by Identifier-Locator Network Protocol for IPv6 (ILNPv6) for providing protection against off-path attacks for packets when ILNPv6 is in use, and as a signal during initial network-layer session creation that ILNPv6 is proposed for use with this network-layer session, rather than classic IPv6.

4.3.13.2. Specification

This option is specified in [RFC6744].

4.3.13.3. Specific Security Implications

Those described in [RFC6744].

4.3.13.4. Operational and Interoperability Impact if Blocked

Discarding packets that contain this option will break INLPv6 deployments.
4.3.13.5. Advice

Intermediate systems should not discard packets based on the presence of this option.

4.3.14. Line-Identification Option (Type=0x8C)

4.3.14.1. Uses

This option is used by an Edge Router to identify the subscriber premises in scenarios where several subscriber premises may be logically connected to the same interface of an Edge Router.

4.3.14.2. Specification

This option is specified in [RFC6788].

4.3.14.3. Specific Security Implications

Those described in [RFC6788].

4.3.14.4. Operational and Interoperability Impact if Blocked

Since this option is meant to be employed in Router Solicitation messages, discarding packets based on the presence of this option at intermediate systems will result in no interoperability implications.

4.3.14.5. Advice

Intermediate devices should discard packets that contain this option.

4.3.15. Deprecated (Type=0x4D)

4.3.15.1. Uses

No information has been found about this option type.

4.3.15.2. Specification

No information has been found about this option type.

4.3.15.3. Specific Security Implications

No information has been found about this option type, and hence it has been impossible to perform the corresponding security assessment.
4.3.15.4. Operational and Interoperability Impact if Blocked

Unknown.

4.3.15.5. Advice

Intermediate systems should discard packets that contain this option.

4.3.16. MPL Option (Type=0x6D)

4.3.16.1. Uses

This option is used with the Multicast Protocol for Low power and Lossy Networks (MPL), that provides IPv6 multicast forwarding in constrained networks.

4.3.16.2. Specification

This option is specified in [RFC7731], and is meant to be included only in Hop-by-Hop Option headers.

4.3.16.3. Specific Security Implications

Those described in [RFC7731].

4.3.16.4. Operational and Interoperability Impact if Blocked

Dropping packets that contain an MPL option within an MPL network would break the Multicast Protocol for Low power and Lossy Networks (MPL). However, dropping such packets at the border of such networks will have no negative impact.

4.3.16.5. Advice

Intermediate systems should not discard packets based on the presence of this option. However, since this option has been specified for the Hop-by-Hop Options, such systems should consider the discussion in Section 3.4.1.

4.3.17. IP_DFF (Type=0xEE)

4.3.17.1. Uses

This option is employed with the (Experimental) Depth-First Forwarding (DFF) in Unreliable Networks.
4.3.17.2. Specification

This option is specified in [RFC6971].

4.3.17.3. Specific Security Implications

Those specified in [RFC6971].

4.3.17.4. Operational and Interoperability Impact if Blocked

Dropping packets containing this option within a routing domain that is running DFF would break DFF. However, dropping such packets at the border of such domains will have no security implications.

4.3.17.5. Advice

Intermediate systems that do not operate within a routing domain that is running DFF should discard packets containing this option.

4.3.18. RFC3692-style Experiment (Types = 0x1E, 0x3E, 0x5E, 0x7E, 0x9E, 0xBE, 0xDE, 0xFE)

4.3.18.1. Uses

These options can be employed for performing RFC3692-style experiments. It is only appropriate to use these values in explicitly configured experiments; they must not be shipped as defaults in implementations.

4.3.18.2. Specification

Specified in RFC 4727 [RFC4727] in the context of RFC3692-style experiments.

4.3.18.3. Specific Security Implications

The specific security implications will depend on the specific use of these options.

4.3.18.4. Operational and Interoperability Impact if Blocked

For obvious reasons, discarding packets that contain these options limits the ability to perform legitimate experiments across IPv6 routers.
4.3.18.5. Advice

Intermediate systems should discard packets that contain these options. Only in specific environments where RFC3692-style experiments are meant to be performed should these options be permitted.

4.4. Advice on the handling of Packets with Unknown IPv6 Options

We refer to IPv6 options that have not been assigned an IPv6 option type in the corresponding registry ([IANA-IPV6-PARAM]) as "unknown IPv6 options".

4.4.1. Uses

New IPv6 options may be specified as part of future protocol work.

4.4.2. Specification

The processing of unknown IPv6 options is specified in [RFC8200].

4.4.3. Specific Security Implications

For obvious reasons, it is impossible to determine specific security implications of unknown IPv6 options.

4.4.4. Operational and Interoperability Impact if Blocked

Discarding unknown IPv6 options may slow down the deployment of new IPv6 options. As noted in [draft-gont-6man-ipv6-opt-transmit], the corresponding IANA registry ([IANA-IPV6-PARAM]) should be monitored such that IPv6 option filtering rules are updated as new IPv6 options are standardized.

4.4.5. Advice

Enterprise intermediate systems that process the contents of IPv6 EHs should discard packets that contain unknown options. Other intermediate systems that process the contents of IPv6 EHs should permit packets that contain unknown options.

5. IANA Considerations

This document has no actions for IANA.
6. Security Considerations

This document provides advice on the filtering of IPv6 packets that contain IPv6 EHs (and possibly IPv6 options) at IPv6 transit routers. It is meant to improve the current situation of widespread dropping of such IPv6 packets in those cases where the drops result from improper configuration defaults, or inappropriate advice in this area.

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Abstract

Knowledge and experience on how to operate IPv4 securely is available: whether it is the Internet or an enterprise internal network. However, IPv6 presents some new security challenges. RFC 4942 describes the security issues in the protocol but network managers also need a more practical, operations-minded document to enumerate advantages and/or disadvantages of certain choices.

This document analyzes the operational security issues in all places of a network (enterprises, service providers and residential users) and proposes technical and procedural mitigations techniques.

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1. Introduction

Running an IPv6 network is new for most operators not only because they are not yet used to large scale IPv6 networks but also because there are subtle differences between IPv4 and IPv6 especially with respect to security. For example, all layer-2 interactions are now done using Neighbor Discovery Protocol [RFC4861] rather than using Address Resolution Protocol [RFC0826]. Also, there are subtle differences between NAT44 [RFC2993] and NPTv6 [RFC6296] which are explicitly pointed out in the latter’s security considerations section.

IPv6 networks are deployed using a variety of techniques, each of which have their own specific security concerns.

This document complements [RFC4942] by listing all security issues when operating a network utilizing varying transition technologies and updating with ones that have been standardized since 2007. It also provides more recent operational deployment experiences where warranted.

1.1. Requirements Language

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in [RFC2119] when they appear in ALL CAPS. These words may also appear in this document in lower case as plain English words, absent their normative meanings.
2. Generic Security Considerations

2.1. Addressing Architecture

IPv6 address allocations and overall architecture are an important part of securing IPv6. Initial designs, even if intended to be temporary, tend to last much longer than expected. Although initially IPv6 was thought to make renumbering easy, in practice, it may be extremely difficult to renumber without a good IP Addresses Management (IPAM) system.

Once an address allocation has been assigned, there should be some thought given to an overall address allocation plan. With the abundance of address space available, an address allocation may be structured around services along with geographic locations, which then can be a basis for more structured security policies to permit or deny services between geographic regions.

A common question is whether companies should use PI vs PA space [RFC7381], but from a security perspective there is little difference. However, one aspect to keep in mind is who has administrative ownership of the address space and who is technically responsible if/when there is a need to enforce restrictions on routability of the space due to malicious criminal activity.

2.1.1. Statically Configured Addresses

When considering how to assign statically configured addresses it is necessary to take into consideration the effectiveness of perimeter security in a given environment. There is a trade-off between ease of operational deployment where some portions of the IPv6 address could be easily recognizable for operational debugging and troubleshooting versus the risk of scanning; [SCANNING] shows that there are scientifically based mechanisms that make scanning for IPv6 reachable nodes more realizable than expected; see also [RFC7707]. The use of common multicast groups which are defined for important networked devices and the use of commonly repeated addresses could make it easy to figure out which devices are name servers, routers or other critical devices.

While in some environments the security is so poor that obfuscating addresses is considered a benefit; it is a better practice to ensure that perimeter rules are actively checked and enforced and that statically configured addresses follow some logical allocation scheme for ease of operation.
2.1.2. Use of ULAs

ULAs are intended for scenarios where IP addresses will not have global scope so they should not appear in the global BGP routing table. The implicit expectation from the RFC is that all ULAs will be randomly created as /48s. Any use of ULAs that are not created as a /48 violates RFC4193 [RFC4193].

ULAs could be useful for infrastructure hiding as described in RFC4864 [RFC4864]. Alternatively Link-Local addresses RFC7404 [RFC7404] could also be used. Although ULAs are supposed to be used in conjunction with global addresses for hosts that desire external connectivity, a few operators chose to use ULAs in conjunction with some sort of address translation at the border in order to maintain a perception of parity between their IPv4 and IPv6 setup. Some operators believe that stateful IPv6 Network Address and Port Translation (NAPT) provides some security not provided by NPTv6 (the authors of this document do not share this point of view). The use of stateful IPv6 NAPT would be problematic in trying to track specific machines that may source malware although this is less of an issue if appropriate logging is done which includes utilizing accurate timestamps and logging a node’s source ports RFC6302 [RFC6302]. Another typical argument in favor of ULA is that there are too many mistakes made with ACL filters at the edge and the use of ULAs could make things easier to set filters.

The use of ULA does not isolate ‘by magic’ the part of the network using ULA from other parts of the network (including the Internet). Although section 4.1 of RFC4193 [RFC4193] explicitly states "If BGP is being used at the site border with an ISP, the default BGP configuration must filter out any Local IPv6 address prefixes, both incoming and outgoing.", the operational reality is that this guideline is not always followed. As written, RFC4193 makes no changes to default routing behavior of exterior protocols. Therefore, routers will happily forward packets whose source or destination address is ULA as long as they have a route to the destination and there is no ACL blocking those packets. This means that using ULA does not prevent route and packet filters having to be implemented and monitored. This also means that all Internet transit networks should consider ULA as source or destination as bogons packets and drop them.

It is important to carefully weigh the benefits of using ULAs versus utilizing a section of the global allocation and creating a more effective filtering strategy. It is also important to note that the IETF does not recommend the use of ULA and NPTv6.
2.1.3. Point-to-Point Links

RFC6164 [RFC6164] recommends the use of /127 for inter-router point-to-point links. A /127 prevents the ping-pong attack between routers. However, it should be noted that at the time of this writing, there are still many networks out there that follow the advice provided by RFC3627 [RFC3627] (obsoleted and marked Historic by RFC6547 [RFC6547]) and therefore continue to use /64’s and/or /112’s. We recommend that the guidance provided by RFC6164 be followed.

Some environments are also using link-local addressing for point-to-point links. While this practice could further reduce the attack surface against infrastructure devices, the operational disadvantages need also to be carefully considered RFC7404 [RFC7404].

2.1.4. Temporary Addresses - Privacy Extensions for SLAAC

Normal stateless address autoconfiguration (SLAAC) relies on the automatically generated EUI-64 address, which together with the /64 prefix makes up the global unique IPv6 address. The EUI-64 address is generated from the MAC address. Randomly generating an interface ID, as described in [RFC4941], is part of SLAAC with so-called privacy extension addresses and used to address some privacy concerns. Privacy extension addresses a.k.a. temporary addresses may help to mitigate the correlation of activities of a node within the same network, and may also reduce the attack exposure window.

As privacy extension addresses could also be used to obfuscate some malevolent activities (whether on purpose or not), it is advised in scenarios where user attribution is important to rely on a layer-2 authentication mechanism such as IEEE 802.1X [IEEE-802.1X] with the appropriate RADIUS accounting (Section 2.6.1.6) or to disable SLAAC and rely only on DHCPv6. However, in scenarios where anonymity is a strong desire (protecting user privacy is more important than user attribution), privacy extension addresses should be used.

Using privacy extension addresses prevents the operator from building a priori host specific access control lists (ACLs). It must be noted that recent versions of Windows do not use the MAC address anymore to build the stable address but use a mechanism similar to the one described in [RFC7217], this also means that such an ACL cannot be configured based solely on the MAC address of the nodes, diminishing the value of such ACL. On the other hand, different VLANs are often used to segregate users, in this case ACL can rely on a /64 prefix per VLAN rather than a per host ACL entry.
The decision to utilize privacy extension addresses can come down to whether the network is managed versus unmanaged. In some environments full visibility into the network is required at all times which requires that all traffic be attributable to where it is sourced or where it is destined to within a specific network. This situation is dependent on what level of logging is performed. If logging considerations include utilizing accurate timestamps and logging a node’s source ports [RFC6302] then there should always exist appropriate user attribution needed to get to the source of any malware originator or source of criminal activity.

Disabling SLAAC and privacy extensions addresses can be done by sending Router Advertisement with a hint to get addresses via DHCPv6 by setting the M-bit but also disabling SLAAC by resetting all A-bits in all prefix information options sent in the Router Advertisement message.

2.1.5. Privacy consideration of Addresses

However, there are several privacy issues still present with [RFC4941] such as host tracking, and address scanning attacks are still possible. More details are provided in Appendix A. of [RFC7217] and in [RFC7721].

2.1.6. DHCP/DNS Considerations

Many environments use DHCPv6 to allocate addresses to ensure audit-ability and traceability (but see Section 2.6.1.5). A main security concern is the ability to detect and counteract against rogue DHCP servers (Section 2.3.2).

DNS is often used for malware activities and while there are no fundamental differences with IPv4 and IPv6 security concerns, there are specific consideration in DNS64 RFC6147 [RFC6147] environments that need to be understood. Specifically the interactions and potential to interference with DNSsec implementation need to be understood - these are pointed out in detail in Section 2.7.3.2.

2.2. Extension Headers

The extension headers are one of the most critical differentiator between IPv4 and IPv6. They have also become a very controversial topic since forwarding nodes that discard packets containing extension headers are known to cause connectivity failures and deployment problems. Understanding the role of varying extension headers is important and this section enumerates the ones that need careful consideration. The IANA has closed the the existing empty
"Next Header Types" registry to new entries and is redirecting its users to a new "IPv6 Extension Header Types" registry.

A clarification on how intermediate nodes should handle existing packets with extension headers and any extension headers that are defined in the future is found in RFC7045 [RFC7045]. The uniform TLV format to be used for defining future extension headers is described in RFC6564 [RFC6564]. Some observations listed in RFC7872 [RFC7872] seems to indicate that packets with certain extension headers may not traverse the Internet to its intended destination based on operator policies.

It must also be noted that there is no indication in the packet whether the Next Protocol field points to an extension header or to a transport header. This may confuse some filtering rules.

2.2.1. Order and Repetition of Extension Headers

While RFC2460 [RFC2460] defines the order and the maximum repetition of extension headers, there are still IPv6 implementations at the time of writing this document which support a wrong order of headers (such as ESP before routing) or an illegal repetition of headers (such as multiple routing headers). The same applies for options contained in the extension headers (see [I-D.kampanakis-6man-ipv6-eh-parsing]). In some cases, it has lead to nodes crashing when receiving or forwarding wrongly formatted packets.

2.2.2. Hop-by-Hop Extension Header

The hop-by-hop extension header, when present in an IPv6 packet, forces all nodes in the path to inspect this header. This is of course a large avenue for a denial of service as most if not all routers cannot process this kind of packets in hardware but have to ‘punt’ this packet for software processing. See also [I-D.ietf-6man-hbh-header-handling].

2.2.3. Fragmentation Extension Header

The fragmentation extension header is used by the source when it has to fragment packets. RFC7112 [RFC7112] explains why it is important to:

- firewall and security devices should drop first fragment not containing enough of the layer-4 header;
- destination node should ignore first fragment not containing the entire IPv6 header chain.
Else, stateless filtering could be bypassed by an hostile party. RFC6980 [RFC6980] applies the same rule to NDP and the RA-guard function.

2.2.4. IP Security Extension Header

The IPsec [RFC4301] [RFC4301] extension headers (AH [RFC4302] and ESP [RFC4303]) are required if IPsec is to be utilized for network level security functionality.

2.3. Link-Layer Security

IPv6 relies heavily on the Neighbor Discovery protocol (NDP) RFC4861 [RFC4861] to perform a variety of link operations such as discovering other nodes on the link, resolving their link-layer addresses, and finding routers on the link. If not secured, NDP is vulnerable to various attacks such as router/neighbor message spoofing, redirect attacks, Duplicate Address Detection (DAD) DoS attacks, etc. many of these security threats to NDP have been documented in IPv6 ND Trust Models and Threats RFC3756 [RFC3756] and in RFC6583 [RFC6583].

2.3.1. SeND and CGA

SEcure Neighbor Discovery (SeND), as described in RFC3971 [RFC3971], is a mechanism that was designed to secure ND messages. This approach involves the use of new NDP options to carry public key based signatures. Cryptographically Generated Addresses (CGA), as described in RFC3972 [RFC3972], are used to ensure that the sender of a Neighbor Discovery message is the actual "owner" of the claimed IPv6 address. A new NDP option, the CGA option, was introduced and is used to carry the public key and associated parameters. Another NDP option, the RSA Signature option, is used to protect all messages relating to neighbor and Router discovery.

SeND protects against:
- Neighbor Solicitation/Advertisement Spoofing
- Neighbor Unreachability Detection Failure
- Duplicate Address Detection DoS Attack
- Router Solicitation and Advertisement Attacks
- Replay Attacks
- Neighbor Discovery DoS Attacks
SeND does NOT:

- Protect statically configured addresses
- Protect addresses configured using fixed identifiers (i.e. EUI-64)
- Provide confidentiality for NDP communications
- Compensate for an unsecured link - SEND does not require that the addresses on the link and Neighbor Advertisements correspond

However, at this time and after many years after their specifications, CGA and SeND do not have wide support from generic operating systems; hence, their usefulness is limited.

2.3.2. Securing DHCP

Dynamic Host Configuration Protocol for IPv6 (DHCPv6), as detailed in RFC3315 [RFC3315], enables DHCP servers to pass configuration parameters such as IPv6 network addresses and other configuration information to IPv6 nodes. DHCP plays an important role in any large network by providing robust stateful configuration and autoregistration of DNS Host Names.

The two most common threats to DHCP clients come from malicious (a.k.a. rogue) or unintentionally misconfigured DHCP servers. A malicious DHCP server is established with the intent of providing incorrect configuration information to the client to cause a denial of service attack or mount a man in the middle attack. While unintentionally, a misconfigured DHCP server can have the same impact. Additional threats against DHCP are discussed in the security considerations section of RFC3315 [RFC3315].

RFC7610 [RFC7610] specifies a mechanism for protecting connected DHCPv6 clients against rogue DHCPv6 servers. This mechanism is based on DHCPv6 packet-filtering at the layer-2 device; the administrator specifies the interfaces connected to DHCPv6 servers.

It is recommended to use DHCP-shield.

2.3.3. ND/RA Rate Limiting

Neighbor Discovery (ND) can be vulnerable to denial of service (DoS) attacks in which a router is forced to perform address resolution for a large number of unassigned addresses. Possible side effects of this attack preclude new devices from joining the network or even worse rendering the last hop router ineffective due to high CPU.
usage. Easy mitigative steps include rate limiting Neighbor Solicitations, restricting the amount of state reserved for unresolved solicitations, and clever cache/timer management.

RFC6583 [RFC6583] discusses the potential for DoS in detail and suggests implementation improvements and operational mitigation techniques that may be used to mitigate or alleviate the impact of such attacks. Here are some feasible mitigation options that can be employed by network operators today:

- Ingress filtering of unused addresses by ACL, route filtering, longer than /64 prefix; These require static configuration of the addresses.
- Tuning of NDP process (where supported).

Additionally, IPv6 ND uses multicast extensively for signaling messages on the local link to avoid broadcast messages for on-the-wire efficiency. However, this has some side effects on wifi networks, especially a negative impact on battery life of smartphones and other battery operated devices that are connected to such networks. The following drafts are actively discussing methods to rate limit RAs and other ND messages on wifi networks in order to address this issue:

- [I-D.thubert-savi-ra-throttler]
- [I-D.chakrabarti-nordmark-6man-efficient-nd]

### 2.3.4. ND/RA Filtering

Router Advertisement spoofing is a well-known attack vector and has been extensively documented. The presence of rogue RAs, either intentional or malicious, can cause partial or complete failure of operation of hosts on an IPv6 link. For example, a host can select an incorrect router address which can be used as a man-in-the-middle (MITM) attack or can assume wrong prefixes to be used for stateless address configuration (SLAAC). RFC6104 [RFC6104] summarizes the scenarios in which rogue RAs may be observed and presents a list of possible solutions to the problem. RFC6105 [RFC6105] (RA-Guard) describes a solution framework for the rogue RA problem where network segments are designed around switching devices that are capable of identifying invalid RAs and blocking them before the attack packets actually reach the target nodes.

However, several evasion techniques that circumvent the protection provided by RA-Guard have surfaced. A key challenge to this mitigation technique is introduced by IPv6 fragmentation. An
attacker can conceal the attack by fragmenting his packets into multiple fragments such that the switching device that is responsible for blocking invalid RAs cannot find all the necessary information to perform packet filtering in the same packet. RFC7113 [RFC7113] describes such evasion techniques, and provides advice to RA-Guard implementers such that the aforementioned evasion vectors can be eliminated.

Given that the IPv6 Fragmentation Header can be leveraged to circumvent current implementations of RA-Guard, RFC6980 [RFC6980] updates RFC4861 [RFC4861] such that use of the IPv6 Fragmentation Header is forbidden in all Neighbor Discovery messages except "Certification Path Advertisement", thus allowing for simple and effective measures to counter Neighbor Discovery attacks.

The Source Address Validation Improvements (SAVI) working group has worked on other ways to mitigate the effects of such attacks. RFC7513 [RFC7513] would help in creating bindings between a DHCPv4 RFC2131 [RFC2131] /DHCPv6 RFC3315 [RFC3315] assigned source IP address and a binding anchor RFC7039 [RFC7039] on a SAVI device. Also, RFC6620 [RFC6620] describes how to glean similar bindings when DHCP is not used. The bindings can be used to filter packets generated on the local link with forged source IP address.

It is still recommended that RA-Guard be employed as a first line of defense against common attack vectors including misconfigured hosts.

2.3.5. 3GPP Link-Layer Security

The 3GPP link is a point-to-point like link that has no link-layer address. This implies there can only be an end host (the mobile hand-set) and the first-hop router (i.e., a GPRS Gateway Support Node (GGSN) or a Packet Gateway (PGW)) on that link. The GGSN/PGW never configures a non link-local address on the link using the advertised /64 prefix on it. The advertised prefix must not be used for on-link determination. There is no need for an address resolution on the 3GPP link, since there are no link-layer addresses. Furthermore, the GGSN/PGW assigns a prefix that is unique within each 3GPP link that uses IPv6 stateless address autoconfiguration. This avoids the necessity to perform DAD at the network level for every address built by the mobile host. The GGSN/PGW always provides an IID to the cellular host for the purpose of configuring the link-local address and ensures the uniqueness of the IID on the link (i.e., no collisions between its own link-local address and the mobile host’s one).
The 3GPP link model itself mitigates most of the known NDP-related Denial-of-Service attacks. In practice, the GGSN/PGW only needs to route all traffic to the mobile host that falls under the prefix assigned to it. As there is also a single host on the 3GPP link, there is no need to defend that IPv6 address.

See Section 5 of RFC6459 [RFC6459] for a more detailed discussion on the 3GPP link model, NDP on it and the address configuration detail.

2.4. Control Plane Security

RFC6192 [RFC6192] defines the router control plane. This definition is repeated here for the reader’s convenience.

Modern router architecture design maintains a strict separation of forwarding and router control plane hardware and software. The router control plane supports routing and management functions. It is generally described as the router architecture hardware and software components for handling packets destined to the device itself as well as building and sending packets originated locally on the device. The forwarding plane is typically described as the router architecture hardware and software components responsible for receiving a packet on an incoming interface, performing a lookup to identify the packet’s IP next hop and determine the best outgoing interface towards the destination, and forwarding the packet out through the appropriate outgoing interface.

While the forwarding plane is usually implemented in high-speed hardware, the control plane is implemented by a generic processor (named router processor RP) and cannot process packets at a high rate. Hence, this processor can be attacked by flooding its input queue with more packets than it can process. The control plane processor is then unable to process valid control packets and the router can lose OSPF or BGP adjacencies which can cause a severe network disruption.

The mitigation technique is:

- To drop non-legit control packet before they are queued to the RP (this can be done by a forwarding plane ACL) and
- To rate limit the remaining packets to a rate that the RP can sustain. Protocol specific protection should also be done (for example, a spoofed OSPFv3 packet could trigger the execution of the Dijkstra algorithm, therefore the number of Dijkstra execution should be also rate limited).

This section will consider several classes of control packets:
2.4.1. Control Protocols

This class includes OSPFv3, BGP, NDP, ICMP.

An ingress ACL to be applied on all the router interfaces SHOULD be configured such as:

- drop OSPFv3 (identified by Next-Header being 89) and RIPng (identified by UDP port 521) packets from a non link-local address
- allow BGP (identified by TCP port 179) packets from all BGP neighbors and drop the others
- allow all ICMP packets (transit and to the router interfaces)

Note: dropping OSPFv3 packets which are authenticated by IPsec could be impossible on some routers whose ACL are unable to parse the IPsec ESP or AH extension headers.

Rate limiting of the valid packets SHOULD be done. The exact configuration obviously depends on the power of the Route Processor.

2.4.2. Management Protocols

This class includes: SSH, SNMP, syslog, NTP, etc

An ingress ACL to be applied on all the router interfaces SHOULD be configured such as:

- Drop packets destined to the routers except those belonging to protocols which are used (for example, permit TCP 22 and drop all when only SSH is used);
- Drop packets where the source does not match the security policy, for example if SSH connections should only be originated from the NOC, then the ACL should permit TCP port 22 packets only from the NOC prefix.
Rate limiting of the valid packets SHOULD be done. The exact configuration obviously depends on the power of the Route Processor.

2.4.3. Packet Exceptions

This class covers multiple cases where a data plane packet is punted to the route processor because it requires specific processing:

- generation of an ICMP packet-too-big message when a data plane packet cannot be forwarded because it is too large;
- generation of an ICMP hop-limit-expired message when a data plane packet cannot be forwarded because its hop-limit field has reached 0;
- generation of an ICMP destination-unreachable message when a data plane packet cannot be forwarded for any reason;
- processing of the hop-by-hop extension header (see also [I-D.ietf-6man-hbh-header-handling]);
- or more specific to some router implementation: an oversized extension header chain which cannot be processed by the hardware and force the packet to be punted to the generic router CPU.

On some routers, not everything can be done by the specialized data plane hardware which requires some packets to be ‘punted’ to the generic RP. This could include for example the processing of a long extension header chain in order to apply an ACL based on layer 4 information. RFC6980 [RFC6980] and more generally RFC7112 [RFC7112] highlights the security implications of oversized extension header chains on routers and updates RFC2460 [RFC2460] such that the first fragment of a packet is required to contain the entire IPv6 header chain.

An ingress ACL cannot help to mitigate a control plane attack using those packet exceptions. The only protection for the RP is to limit the rate of those packet exceptions forwarded to the RP, this means that some data plane packets will be dropped without any ICMP messages back to the source which will cause Path MTU holes. But, there is no other solution.

In addition to limiting the rate of data plane packets queued to the RP, it is also important to limit the generation rate of ICMP messages both the save the RP but also to prevent an amplification attack using the router as a reflector.
2.5. Routing Security

Routing security in general can be broadly divided into three sections:

1. Authenticating neighbors/peers
2. Securing routing updates between peers
3. Route filtering

[RFC7454] covers these sections specifically for BGP in detail.

2.5.1. Authenticating Neighbors/Peers

A basic element of routing is the process of forming adjacencies, neighbor, or peering relationships with other routers. From a security perspective, it is very important to establish such relationships only with routers and/or administrative domains that one trusts. A traditional approach has been to use MD5 HMAC, which allows routers to authenticate each other prior to establishing a routing relationship.

OSPFv3 can rely on IPsec to fulfill the authentication function. However, it should be noted that IPsec support is not standard on all routing platforms. In some cases, this requires specialized hardware that offloads crypto over to dedicated ASICs or enhanced software images (both of which often come with added financial cost) to provide such functionality. An added detail is to determine whether OSPFv3 IPsec implementations use AH or ESP-Null for integrity protection. In early implementations all OSPFv3 IPsec configurations relied on AH since the details weren’t specified in RFC5340 [RFC5340] or RFC2740 [RFC2740] that was obsoleted by the former. However, the document which specifically describes how IPsec should be implemented for OSPFv3 RFC4552 [RFC4552] specifically states that ESP-Null MUST and AH MAY be implemented since it follows the overall IPsec standards wordings. OSPFv3 can also use normal ESP to encrypt the OSPFv3 payload to hide the routing information.

RFC7166 [RFC7166] (which obsoletes RFC6506 [RFC6506]) changes OSPFv3’s reliance on IPsec by appending an authentication trailer to the end of the OSPFv3 packets. This document does not specifically provide for a mechanism that will authenticate the specific originator of a packet. Rather, it will allow a router to confirm that the packet has indeed been issued by a router that had access to the shared authentication key.
With all authentication mechanisms, operators should confirm that implementations can support re-keying mechanisms that do not cause outages. There have been instances where any re-keying cause outages and therefore the tradeoff between utilizing this functionality needs to be weighed against the protection it provides.

2.5.2. Securing Routing Updates Between Peers

IPv6 initially mandated the provisioning of IPsec capability in all nodes. However, in the updated IPv6 Nodes Requirement standard RFC6434 [RFC6434] is now a SHOULD and not MUST implement. Theoretically it is possible, and recommended, that communication between two IPv6 nodes, including routers exchanging routing information be encrypted using IPsec. In practice however, deploying IPsec is not always feasible given hardware and software limitations of various platforms deployed, as described in the earlier section. Additionally, in a protocol such as OSPFv3 where adjacencies are formed on a one-to-many basis, IPsec key management becomes difficult to maintain and is not often utilized.

2.5.3. Route Filtering

Route filtering policies will be different depending on whether they pertain to edge route filtering vs internal route filtering. At a minimum, IPv6 routing policy as it pertains to routing between different administrative domains should aim to maintain parity with IPv4 from a policy perspective e.g.,

- Filter internal-use, non-globally routable IPv6 addresses at the perimeter
- Discard packets from and to bogon and reserved space
- Configure ingress route filters that validate route origin, prefix ownership, etc. through the use of various routing databases, e.g., RADB. There is additional work being done in this area to formally validate the origin ASs of BGP announcements in RFC6810 [RFC6810]

Some good recommendations for filtering can be found from Team CYMRU at [CYMRU].

2.6. Logging/Monitoring

In order to perform forensic research in case of any security incident or to detect abnormal behaviors, network operator should log multiple pieces of information.
This includes:

- logs of all applications when available (for example web servers);
- use of IP Flow Information Export [RFC7011] also known as IPfix;
- use of SNMP MIB [RFC4293];
- use of the Neighbor cache;
- use of stateful DHCPv6 [RFC3315] lease cache, especially when a relay agent [RFC6221] in layer-2 switches is used;
- use of RADIUS [RFC2866] for accounting records.

Please note that there are privacy issues related to how those logs are collected, kept and safely discarded. Operators are urged to check their country legislation.

All those pieces of information will be used for:

- forensic (Section 2.6.2.1) research to answer questions such as who did what and when?
- correlation (Section 2.6.2.3): which IP addresses were used by a specific node (assuming the use of privacy extensions addresses [RFC4941])
- inventory (Section 2.6.2.2): which IPv6 nodes are on my network?
- abnormal behavior detection (Section 2.6.2.4): unusual traffic patterns are often the symptoms of an abnormal behavior which is in turn a potential attack (denial of services, network scan, a node being part of a botnet, ...)

2.6.1. Data Sources

This section lists the most important sources of data that are useful for operational security.

2.6.1.1. Logs of Applications

Those logs are usually text files where the remote IPv6 address is stored in all characters (not binary). This can complicate the processing since one IPv6 address, 2001:db8::1 can be written in multiple ways such as:

- 2001:DB8::1 (in uppercase)
RFC 5952 [RFC5952] explains this problem in detail and recommends the use of a single canonical format (in short use lower case and suppress leading 0). This memo recommends the use of canonical format [RFC5952] for IPv6 addresses in all possible cases. If the existing application cannot log under the canonical format, then this memo recommends the use an external program in order to canonicalize all IPv6 addresses.

For example, this perl script can be used:

```perl
#!/usr/bin/perl -w
use strict;
use warnings;
use Socket;
use Socket6;

my (@words, $word, $binary_address);

## go through the file one line at a time
while (my $line = <STDIN>) {
    chomp $line;
    foreach my $word (split /[\s+]/, $line) {
        $binary_address = inet_pton AF_INET6, $word;
        if ($binary_address) {
            print inet_ntop AF_INET6, $binary_address;
        } else {
            print $word;
        }
        print " ";
    }
    print "\n";
}
```

2.6.1.2. IP Flow Information Export by IPv6 Routers

IPfix [RFC7012] defines some data elements that are useful for security:

- in section 5.4 (IP Header fields): nextHeaderIPv6 and sourceIPv6Address;
- in section 5.6 (Sub-IP fields) sourceMacAddress.
Moreover, IPfix is very efficient in terms of data handling and transport. It can also aggregate flows by a key such as sourceMacAddress in order to have aggregated data associated with a specific sourceMacAddress. This memo recommends the use of IPfix and aggregation on nextHeaderIPv6, sourceIPv6Address and sourceMacAddress.

2.6.1.3. SNMP MIB by IPv6 Routers

RFC 4293 [RFC4293] defines a Management Information Base (MIB) for the two address families of IP. This memo recommends the use of:

- ipIfStatsTable table which collects traffic counters per interface;
- ipNetToPhysicalTable table which is the content of the Neighbor cache, i.e. the mapping between IPv6 and data-link layer addresses.

2.6.1.4. Neighbor Cache of IPv6 Routers

The neighbor cache of routers contains all mappings between IPv6 addresses and data-link layer addresses. It is usually available by two means:

- the SNMP MIB (Section 2.6.1.3) as explained above;
- also by connecting over a secure management channel (such as SSH or HTTPS) and explicitly requesting a neighbor cache dump.

The neighbor cache is highly dynamic as mappings are added when a new IPv6 address appears on the network (could be quite often with privacy extension addresses [RFC4941] or when they are removed when the state goes from UNREACH to removed (the default time for a removal per Neighbor Unreachability Detection [RFC4861] algorithm is 38 seconds for a typical host such as Windows 7). This means that the content of the neighbor cache must periodically be fetched every 30 seconds (to be on the safe side) and stored for later use.

This is an important source of information because it is trivial (on a switch not using the SAVI [RFC7039] algorithm) to defeat the mapping between data-link layer address and IPv6 address. Let us rephrase the previous statement: having access to the current and past content of the neighbor cache has a paramount value for forensic and audit trail.
2.6.1.5. Stateful DHCPv6 Lease

In some networks, IPv6 addresses are managed by stateful DHCPv6 server [RFC3315] that leases IPv6 addresses to clients. It is indeed quite similar to DHCP for IPv4 so it can be tempting to use this DHCP lease file to discover the mapping between IPv6 addresses and data-link layer addresses as it was usually done in the IPv4 era.

It is not so easy in the IPv6 era because not all nodes will use DHCPv6 (there are nodes which can only do stateless autoconfiguration) but also because DHCPv6 clients are identified not by their hardware-client address as in IPv4 but by a DHCP Unique ID (DUID) which can have several formats: some being the data-link layer address, some being data-link layer address prepended with time information or even an opaque number which is useless for operation security. Moreover, when the DUID is based on the data-link address, this address can be of any interface of the client (such as the wireless interface while the client actually uses its wired interface to connect to the network).

If a lightweight DHCP relay agent [RFC6221] is used in the layer-2 switches, then the DHCP server also receives the Interface-ID information which could be save in order to identify the interface of the switches which received a specific leased IPv6 address.

In short, the DHCPv6 lease file is less interesting than in the IPv4 era. DHCPv6 servers that keeps the relayed data-link layer address in addition to the DUID in the lease file do not suffer from this limitation. On a managed network where all hosts support DHCPv6, special care must be taken to prevent stateless autoconfiguration anyway (and if applicable) by sending RA with all announced prefixes without the A-bit set.

The mapping between data-link layer address and the IPv6 address can be secured by using switches implementing the SAVI [RFC7513] algorithms. Of course, this also requires that data-link layer address is protected by using layer-2 mechanism such as [IEEE-802.1X].

2.6.1.6. RADIUS Accounting Log

For interfaces where the user is authenticated via a RADIUS [RFC2866] server, and if RADIUS accounting is enabled, then the RADIUS server receives accounting Acct-Status-Type records at the start and at the end of the connection which include all IPv6 (and IPv4) addresses used by the user. This technique can be used notably for Wi-Fi networks with Wi-Fi Protected Address (WPA) or any other IEEE 802.1X [IEEE-802.1X] wired interface on an Ethernet switch.
2.6.1.7. Other Data Sources

There are other data sources that must be kept exactly as in the IPv4 network:

- historical mapping of IPv6 addresses to users of remote access VPN;
- historical mapping of MAC address to switch interface in a wired network.

2.6.2. Use of Collected Data

This section leverages the data collected as described before (Section 2.6.1) in order to achieve several security benefits.

2.6.2.1. Forensic

The forensic use case is when the network operator must locate an IPv6 address that was present in the network at a certain time or is still currently in the network.

The source of information can be, in decreasing order, neighbor cache, DHCP lease file. Then, the procedure is:

1. based on the IPv6 prefix of the IPv6 address find the router(s) which are used to reach this prefix;

2. based on this limited set of routers, on the incident time and on IPv6 address to retrieve the data-link address from live neighbor cache, from the historical data of the neighbor cache, or from the DHCP lease file;

3. based on the data-link layer address, look-up on which switch interface was this data-link layer address. In the case of wireless LAN, the RADIUS log should have the mapping between user identification and the MAC address.

At the end of the process, the interface where the malicious user was connected or the username that was used by the malicious user is found.

2.6.2.2. Inventory

RFC 7707 [RFC7707] (which obsoletes RFC 5157 [RFC5157]) is about the difficulties to scan an IPv6 network due to the vast number of IPv6 addresses per link. This has the side effect of making the inventory task difficult in an IPv6 network while it was trivial to do in an
IPv4 network (a simple enumeration of all IPv4 addresses, followed by a ping and a TCP/UDP port scan). Getting an inventory of all connected devices is of prime importance for a secure operation of a network.

There are many ways to do an inventory of an IPv6 network.

The first technique is to use the IPfix information and extract the list of all IPv6 source addresses to find all IPv6 nodes that sent packets through a router. This is very efficient but alas will not discover silent node that never transmitted such packets... Also, it must be noted that link-local addresses will never be discovered by this means.

The second way is again to use the collected neighbor cache content to find all IPv6 addresses in the cache. This process will also discover all link-local addresses. See Section 2.6.1.4.

Another way works only for local network, it consists in sending an ICMP ECHO_REQUEST to the link-local multicast address ff02::1 which is all IPv6 nodes on the network. All nodes should reply to this ECHO_REQUEST per [RFC4443].

Other techniques involve enumerating the DNS zones, parsing log files, leveraging service discovery such as mDNS RFC6762 [RFC6762] and RFC6763 [RFC6763].

Other techniques involve enumerating the DNS zones, especially looking at reverse DNS records and CNAMEs. Or scanning for DNS misconfigurations to find DNS servers that send NXDOMAIN instead of NOERROR for non-existing nodes with children, which violates RFC8020 [RFC8020]. Parsing log files and leveraging service discovery such as mDNS RFC6762 [RFC6762] and RFC6763 [RFC6763] are also added techniques.

2.6.2.3. Correlation

In an IPv4 network, it is easy to correlate multiple logs, for example to find events related to a specific IPv4 address. A simple Unix grep command was enough to scan through multiple text-based files and extract all lines relevant to a specific IPv4 address.

In an IPv6 network, this is slightly more difficult because different character strings can express the same IPv6 address. Therefore, the simple Unix grep command cannot be used. Moreover, an IPv6 node can have multiple IPv6 addresses...
In order to do correlation in IPv6-related logs, it is advised to have all logs with canonical IPv6 addresses. Then, the neighbor cache current (or historical) data set must be searched to find the data-link layer address of the IPv6 address. Then, the current and historical neighbor cache data sets must be searched for all IPv6 addresses associated to this data-link layer address: this is the search set. The last step is to search in all log files (containing only IPv6 address in canonical format) for any IPv6 addresses in the search set.

2.6.2.4. Abnormal Behavior Detection

Abnormal behaviors (such as network scanning, spamming, denial of service) can be detected in the same way as in an IPv4 network

- sudden increase of traffic detected by interface counter (SNMP) or by aggregated traffic from IPfix records [RFC7012];

- change of traffic pattern (number of connection per second, number of connection per host...) with the use of IPfix [RFC7012]

2.6.3. Summary

While some data sources (IPfix, MIB, switch CAM tables, logs, ...) used in IPv4 are also used in the secure operation of an IPv6 network, the DHCPv6 lease file is less reliable and the neighbor cache is of prime importance.

The fact that there are multiple ways to express in a character string the same IPv6 address renders the use of filters mandatory when correlation must be done.

2.7. Transition/Coexistence Technologies

Some text

2.7.1. Dual Stack

Dual stack has established itself as the preferred deployment choice for most network operators without an MPLS core where 6PE RFC4798 [RFC4798] is quite common. Dual stacking the network offers many advantages over other transition mechanisms. Firstly, it is easy to turn on without impacting normal IPv4 operations. Secondly, perhaps more importantly, it is easier to troubleshoot when things break. Dual stack allows you to gradually turn IPv4 operations down when your IPv6 network is ready for prime time.
From an operational security perspective, this now means that you have twice the exposure. One needs to think about protecting both protocols now. At a minimum, the IPv6 portion of a dual stacked network should maintain parity with IPv4 from a security policy point of view. Typically, the following methods are employed to protect IPv4 networks at the edge:

- ACLs to permit or deny traffic
- Firewalls with stateful packet inspection

It is recommended that these ACLs and/or firewalls be additionally configured to protect IPv6 communications. Also, given the end-to-end connectivity that IPv6 provides, it is also recommended that hosts be fortified against threats. General device hardening guidelines are provided in Section 2.8.

2.7.2. Transition Mechanisms

There are many tunnels used for specific use cases. Except when protected by IPsec [RFC4301], all those tunnels have a couple of security issues (most of them being described in RFC 6169 [RFC6169]):

- tunnel injection: a malevolent person knowing a few pieces of information (for example the tunnel endpoints and the used protocol) can forge a packet which looks like a legit and valid encapsulated packet that will gladly be accepted by the destination tunnel endpoint, this is a specific case of spoofing;

- traffic interception: no confidentiality is provided by the tunnel protocols (without the use of IPsec), therefore anybody on the tunnel path can intercept the traffic and have access to the clear-text IPv6 packet;

- service theft: as there is no authorization, even a non authorized user can use a tunnel relay for free (this is a specific case of tunnel injection);

- reflection attack: another specific use case of tunnel injection where the attacker injects packets with an IPv4 destination address not matching the IPv6 address causing the first tunnel endpoint to re-encapsulate the packet to the destination... Hence, the final IPv4 destination will not see the original IPv4 address but only one IPv4 address of the relay router.

- bypassing security policy: if a firewall or an IPS is on the path of the tunnel, then it will probably neither inspect not detect an malevolent IPv6 traffic contained in the tunnel.
To mitigate the bypassing of security policies, it could be helpful to block all default configuration tunnels by denying all IPv4 traffic matching:

- IP protocol 41: this will block ISATAP (Section 2.7.2.2), 6to4 (Section 2.7.2.4), 6rd (Section 2.7.2.5) as well as 6in4 (Section 2.7.2.1) tunnels;
- IP protocol 47: this will block GRE (Section 2.7.2.1) tunnels;
- UDP protocol 3544: this will block the default encapsulation of Teredo (Section 2.7.2.3) tunnels.

Ingress filtering [RFC2827] should also be applied on all tunnel endpoints if applicable to prevent IPv6 address spoofing.

As several of the tunnel techniques share the same encapsulation (i.e. IPv4 protocol 41) and embed the IPv4 address in the IPv6 address, there are a set of well-known looping attacks described in RFC 6324 [RFC6324], this RFC also proposes mitigation techniques.

### 2.7.2.1. Site-to-Site Static Tunnels

Site-to-site static tunnels are described in RFC 2529 [RFC2529] and in GRE [RFC2784]. As the IPv4 endpoints are statically configured and are not dynamic they are slightly more secure (bi-directional service theft is mostly impossible) but traffic interception ad tunnel injection are still possible. Therefore, the use of IPsec [RFC4301] in transport mode and protecting the encapsulated IPv4 packets is recommended for those tunnels. Alternatively, IPsec in tunnel mode can be used to transport IPv6 traffic over a non-trusted IPv4 network.

### 2.7.2.2. ISATAP

ISATAP tunnels [RFC5214] are mainly used within a single administrative domain and to connect a single IPv6 host to the IPv6 network. This means that endpoints and the tunnel endpoint are usually managed by a single entity; therefore, audit trail and strict anti-spoofing are usually possible and this raises the overall security.

Special care must be taken to avoid looping attack by implementing the measures of RFC 6324 [RFC6324] and of RFC6964 [RFC6964].

IPsec [RFC4301] in transport or tunnel mode can be used to secure the IPv4 ISATAP traffic to provide IPv6 traffic confidentiality and prevent service theft.
2.7.2.3. Teredo

Teredo tunnels [RFC4380] are mainly used in a residential environment because that can easily traverse an IPv4 NAT-PT device thanks to its UDP encapsulation and they connect a single host to the IPv6 Internet. Teredo shares the same issues as other tunnels: no authentication, no confidentiality, possible spoofing and reflection attacks.

IPsec [RFC4301] for the transported IPv6 traffic is recommended.

The biggest threat to Teredo is probably for IPv4-only network as Teredo has been designed to easily traverse IPv4 NAT-PT devices which are quite often co-located with a stateful firewall. Therefore, if the stateful IPv4 firewall allows unrestricted UDP outbound and accept the return UDP traffic, then Teredo actually punches a hole in this firewall for all IPv6 traffic to the Internet and from the Internet. While host policies can be deployed to block Teredo in an IPv4-only network in order to avoid this firewall bypass, it would be more efficient to block all UDP outbound traffic at the IPv4 firewall if deemed possible (of course, at least port 53 should be left open for DNS traffic).

2.7.2.4. 6to4

6to4 tunnels [RFC3056] require a public routable IPv4 address in order to work correctly. They can be used to provide either one IPv6 host connectivity to the IPv6 Internet or multiple IPv6 networks connectivity to the IPv6 Internet. The 6to4 relay is usually the anycast address defined in RFC3068 [RFC3068] which has been deprecated by RFC7526 [RFC7526], and is no more used by recent Operating Systems. Some security considerations are explained in RFC3694 [RFC3964].

RFC6343 [RFC6343] points out that if an operator provides well-managed servers and relays for 6to4, non-encapsulated IPv6 packets will pass through well-defined points (the native IPv6 interfaces of those servers and relays) at which security mechanisms may be applied. Client usage of 6to4 by default is now discouraged, and significant precautions are needed to avoid operational problems.

2.7.2.5. 6rd

While 6rd tunnels share the same encapsulation as 6to4 tunnels (Section 2.7.2.4), they are designed to be used within a single SP domain, in other words they are deployed in a more constrained environment than 6to4 tunnels and have little security issues except
lack of confidentiality. The security considerations (Section 12) of RFC5969 [RFC5969] describes how to secure the 6rd tunnels.

IPsec [RFC4301] for the transported IPv6 traffic can be used if confidentiality is important.

2.7.2.6. 6PE and 6VPE

Organizations using MPLS in their core can also use 6PE [RFC4798] and 6VPE RFC4659 [RFC4659] to enable IPv6 access over MPLS. As 6PE and 6VPE are really similar to BGP/MPLS IP VPN described in RFC4364 [RFC4364], the security of these networks is also similar to the one described in RFC4381 [RFC4381]. It relies on:

- Address space, routing and traffic separation with the help of VRF (only applicable to 6VPE);
- Hiding the IPv4 core, hence removing all attacks against P-routers;
- Securing the routing protocol between CE and PE, in the case of 6PE and 6VPE, link-local addresses (see [RFC7404]) can be used and as these addresses cannot be reached from outside of the link, the security of 6PE and 6VPE is even higher than the IPv4 BGP/MPLS IP VPN.

2.7.2.7. DS-Lite

DS-lite is more a translation mechanism and is therefore analyzed further (Section 2.7.3.3) in this document.

2.7.2.8. Mapping of Address and Port

With the tunnel and encapsulation versions of mapping of Address and Port (MAP-E [RFC7597] and MAP-T [RFC7599]), the access network is purely an IPv6 network and MAP protocols are used to give IPv4 hosts on the subscriber network, access to IPv4 hosts on the Internet. The subscriber router does stateful operations in order to map all internal IPv4 addresses and layer-4 ports to the IPv4 address and the set of layer-4 ports received through MAP configuration process. The SP equipment always does stateless operations (either decapsulation or stateless translation). Therefore, as opposed to Section 2.7.3.3 there is no state-exhaustion DoS attack against the SP equipment because there is no state and there is no operation caused by a new layer-4 connection (no logging operation).

The SP MAP equipment MUST implement all the security considerations of [RFC7597]; notably, ensuring that the mapping of the IPv4 address...
and port are consistent with the configuration. As MAP has a predictable IPv4 address and port mapping, the audit logs are easier to manager.

2.7.3. Translation Mechanisms

Translation mechanisms between IPv4 and IPv6 networks are alternative coexistence strategies while networks transition to IPv6. While a framework is described in [RFC6144] the specific security considerations are documented in each individual mechanism. For the most part they specifically mention interference with IPsec or DNSSEC deployments, how to mitigate spoofed traffic and what some effective filtering strategies may be.

2.7.3.1. Carrier-Grade Nat (CGN)

Carrier-Grade NAT (CGN), also called NAT444 CGN or Large Scale NAT (LSN) or SP NAT is described in [RFC6264] and is utilized as an interim measure to prolong the use of IPv4 in a large service provider network until the provider can deploy and effective IPv6 solution. [RFC6598] requested a specific IANA allocated /10 IPv4 address block to be used as address space shared by all access networks using CGN. This has been allocated as 100.64.0.0/10.

Section 13 of [RFC6269] lists some specific security-related issues caused by large scale address sharing. The Security Considerations section of [RFC6598] also lists some specific mitigation techniques for potential misuse of shared address space.

RFC7422 [RFC7422] suggests the use of deterministic address mapping in order to reduce logging requirements for CGN. The idea is to have an algorithm mapping back and forth the internal subscriber to public ports.

2.7.3.2. NAT64/DNS64

Stateful NAT64 translation [RFC6146] allows IPv6-only clients to contact IPv4 servers using unicast UDP, TCP, or ICMP. It can be used in conjunction with DNS64 [RFC6147], a mechanism which synthesizes AAAA records from existing A records. There is also a stateless NAT64 [RFC6145] which is similar for the security aspects with the added benefit of being stateless, so, less prone to a state exhaustion attack.

The Security Consideration sections of [RFC6146] and [RFC6147] list the comprehensive issues. A specific issue with the use of NAT64 is that it will interfere with most IPsec deployments unless UDP
encapsulation is used. DNS64 has an incidence on DNSSEC see section 3.1 of [RFC7050].

2.7.3.3. DS-Lite

Dual-Stack Lite (DS-Lite) [RFC6333] is a transition technique that enables a service provider to share IPv4 addresses among customers by combining two well-known technologies: IP in IP (IPv4-in-IPv6) and Network Address and Port Translation (NAPT).

Security considerations with respect to DS-Lite mainly revolve around logging data, preventing DoS attacks from rogue devices (as the AFTR function is stateful) and restricting service offered by the AFTR only to registered customers.

Section 11 of [RFC6333] describes important security issues associated with this technology.

2.8. General Device Hardening

There are many environments which rely too much on the network infrastructure to disallow malicious traffic to get access to critical hosts. In new IPv6 deployments it has been common to see IPv6 traffic enabled but none of the typical access control mechanisms enabled for IPv6 device access. With the possibility of network device configuration mistakes and the growth of IPv6 in the overall Internet it is important to ensure that all individual devices are hardened against miscreant behavior.

The following guidelines should be used to ensure appropriate hardening of the host, be it an individual computer or router, firewall, load-balancer, server, etc device.

- Restrict access to the device to authorized individuals
- Monitor and audit access to the device
- Turn off any unused services on the end node
- Understand which IPv6 addresses are being used to source traffic and change defaults if necessary
- Use cryptographically protected protocols for device management if possible (SCP, SNMPv3, SSH, TLS, etc)
- Use host firewall capabilities to control traffic that gets processed by upper layer protocols
Use virus scanners to detect malicious programs

3. Enterprises Specific Security Considerations

Enterprises generally have robust network security policies in place to protect existing IPv4 networks. These policies have been distilled from years of experiential knowledge of securing IPv4 networks. At the very least, it is recommended that enterprise networks have parity between their security policies for both protocol versions.

Security considerations in the enterprise can be broadly categorized into two sections - External and Internal.

3.1. External Security Considerations:

The external aspect deals with providing security at the edge or perimeter of the enterprise network where it meets the service providers network. This is commonly achieved by enforcing a security policy either by implementing dedicated firewalls with stateful packet inspection or a router with ACLs. A common default IPv4 policy on firewalls that could easily be ported to IPv6 is to allow all traffic outbound while only allowing specific traffic, such as established sessions, inbound (see also [RFC6092]). Here are a few more things that could enhance the default policy:

- Filter internal-use IPv6 addresses at the perimeter
- Discard packets from and to bogon and reserved space, see also [CYMRU]
- Accept certain ICMPv6 messages to allow proper operation of ND and PMTUD, see also [RFC4890]
- Filter specific extension headers by accepting only the required ones (white list approach) such as ESP, AH (not forgetting the required transport layers: ICMP, TCP, UDP, ...) , where possible at the edge and possibly inside the perimeter; see also [I-D.gont-opsec-ipv6-eh-filtering]
- Filter packets having an illegal IPv6 headers chain at the perimeter (and possible inside as well), see Section 2.2
- Filter unneeded services at the perimeter
- Implement anti-spoofing
- Implement appropriate rate-limiters and control-plane policers
3.2. Internal Security Considerations:

The internal aspect deals with providing security inside the perimeter of the network, including the end host. The most significant concerns here are related to Neighbor Discovery. At the network level, it is recommended that all security considerations discussed in Section 2.3 be reviewed carefully and the recommendations be considered in-depth as well.

As mentioned in Section 2.6.2, care must be taken when running automated IPv6-in-IP4 tunnels.

Hosts need to be hardened directly through security policy to protect against security threats. The host firewall default capabilities have to be clearly understood, especially 3rd party ones which can have different settings for IPv4 or IPv6 default permit/deny behavior. In some cases, 3rd party firewalls have no IPv6 support whereas the native firewall installed by default has it. General device hardening guidelines are provided in Section 2.8.

It should also be noted that many hosts still use IPv4 for transport for things like RADIUS, TACACS+, SYSLOG, etc. This will require some extra level of due diligence on the part of the operator.

4. Service Providers Security Considerations

4.1. BGP

The threats and mitigation techniques are identical between IPv4 and IPv6. Broadly speaking they are:

- Authenticating the TCP session;
- TTL security (which becomes hop-limit security in IPv6);
- Prefix Filtering.

These are explained in more detail in section Section 2.5.

4.1.1. Remote Triggered Black Hole Filtering

RTBH [RFC5635] works identically in IPv4 and IPv6. IANA has allocated 100::/64 as discard prefix RFC6666 [RFC6666].
4.2. Transition Mechanism

SP will typically use transition mechanisms such as 6rd, 6PE, MAP, DS-Lite which have been analyzed in the transition Section 2.7.2 section.

4.3. Lawful Intercept

The Lawful Intercept requirements are similar for IPv6 and IPv4 architectures and will be subject to the laws enforced in varying geographic regions. The local issues with each jurisdiction can make this challenging and both corporate legal and privacy personnel should be involved in discussions pertaining to what information gets logged and what the logging retention policies will be.

The target of interception will usually be a residential subscriber (e.g. his/her PPP session or physical line or CPE MAC address). With the absence of NAT on the CPE, IPv6 has the provision to allow for intercepting the traffic from a single host (a /128 target) rather than the whole set of hosts of a subscriber (which could be a /48, a /60 or /64).

In contrast, in mobile environments, since the 3GPP specifications allocate a /64 per device, it may be sufficient to intercept traffic from the /64 rather than specific /128’s (since each time the device powers up it gets a new IID).

A sample architecture which was written for informational purposes is found in [RFC3924].

5. Residential Users Security Considerations

The IETF Homenet working group is working on how IPv6 residential network should be done; this obviously includes operational security considerations; but, this is still work in progress.

Residential users have usually less experience and knowledge about security or networking. As most of the recent hosts, smartphones, tablets have all IPv6 enabled by default, IPv6 security is important for those users. Even with an IPv4-only ISP, those users can get IPv6 Internet access with the help of Teredo tunnels. Several peer-to-peer programs (notably Bittorrent) support IPv6 and those programs can initiate a Teredo tunnel through the IPv4 residential gateway, with the consequence of making the internal host reachable from any IPv6 host on the Internet. It is therefore recommended that all host security products (personal firewall, ...) are configured with a dual-stack security policy.
If the Residential Gateway has IPv6 connectivity, [RFC7084] (which obsoletes [RFC6204]) defines the requirements of an IPv6 CPE and does not take position on the debate of default IPv6 security policy as defined in [RFC6092]:

- **outbound only**: allowing all internally initiated connections and block all externally initiated ones, which is a common default security policy enforced by IPv4 Residential Gateway doing NAT-PT but it also breaks the end-to-end reachability promise of IPv6. [RFC6092] lists several recommendations to design such a CPE;

- **open/transparent**: allowing all internally and externally initiated connections, therefore restoring the end-to-end nature of the Internet for the IPv6 traffic but having a different security policy for IPv6 than for IPv4.

[RFC6092] REC-49 states that a choice must be given to the user to select one of those two policies.

There is also an alternate solution which has been deployed notably by Swisscom ([I-D.ietf-v6ops-balanced-ipv6-security]: open to all outbound and inbound connections at the exception of an handful of TCP and UDP ports known as vulnerable.

6. Further Reading

There are several documents that describe in more details the security of an IPv6 network; these documents are not written by the IETF but are listed here for your convenience:

1. Guidelines for the Secure Deployment of IPv6 [NIST]

7. Acknowledgements

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8. IANA Considerations

This memo includes no request to IANA.

9. Security Considerations

This memo attempts to give an overview of security considerations of operating an IPv6 network both in an IPv6-only network and in utilizing the most widely deployed IPv4/IPv6 coexistence strategies.

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Abstract

Knowledge and experience on how to operate IPv4 securely is available: whether it is the Internet or an enterprise internal network. However, IPv6 presents some new security challenges. RFC 4942 describes the security issues in the protocol but network managers also need a more practical, operations-minded document to enumerate advantages and/or disadvantages of certain choices.

This document analyzes the operational security issues in several places of a network (enterprises, service providers and residential users) and proposes technical and procedural mitigations techniques.

Status of This Memo

This Internet-Draft is submitted in full conformance with the provisions of BCP 78 and BCP 79.

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1. Introduction

Running an IPv6 network is new for most operators not only because they are not yet used to large scale IPv6 networks but also because there are subtle differences between IPv4 and IPv6 especially with respect to security. For example, all layer-2 interactions are now done using Neighbor Discovery Protocol [RFC4861] rather than using Address Resolution Protocol [RFC0826]. Also, there are subtle differences between NAT44 [RFC2993] and NPTv6 [RFC6296] which are explicitly pointed out in the latter’s security considerations section.

IPv6 networks are deployed using a variety of techniques, each of which have their own specific security concerns.

This document complements [RFC4942] by listing all security issues when operating a network utilizing varying transition technologies and updating with ones that have been standardized since 2007. It also provides more recent operational deployment experiences where warranted.
1.1. Requirements Language

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in [RFC2119] when they appear in ALL CAPS. These words may also appear in this document in lower case as plain English words, absent their normative meanings.

2. Generic Security Considerations

2.1. Addressing Architecture

IPv6 address allocations and overall architecture are an important part of securing IPv6. Initial designs, even if intended to be temporary, tend to last much longer than expected. Although initially IPv6 was thought to make renumbering easy, in practice, it may be extremely difficult to renumber without a good IP Address Management (IPAM) system.

Once an address allocation has been assigned, there should be some thought given to an overall address allocation plan. With the abundance of address space available, an address allocation may be structured around services along with geographic locations, which then can be a basis for more structured security policies to permit or deny services between geographic regions.

A common question is whether companies should use PI vs PA space [RFC7381], but from a security perspective there is little difference. However, one aspect to keep in mind is who has administrative ownership of the address space and who is technically responsible if/when there is a need to enforce restrictions on routability of the space due to malicious criminal activity. Using PA space exposes the organization to a renumbering of the complete network including security policies (based on ACL), audit system, ... in short a complex task which could lead to some temporary security risk if done for a large network and without automation; hence, for large network, PI space should be preferred even if it comes with additional complexities (for example BGP routing) and duties (adding a route6 object in the Regional Internet Registry database).

2.1.1. Statically Configured Addresses

When considering how to assign statically configured addresses it is necessary to take into consideration the effectiveness of perimeter security in a given environment. There is a trade-off between ease of operation (where some portions of the IPv6 address could be easily recognizable for operational debugging and troubleshooting) versus the risk of trivial scanning used for reconnaissance. [SCANNING]
shows that there are scientifically based mechanisms that make scanning for IPv6 reachable nodes more realizable than expected; see also [RFC7707]. The use of common multicast groups which are defined for important networked devices and the use of commonly repeated addresses could make it easy to figure out which devices are name servers, routers or other critical devices; even a simple traceroute will expose most of the routers on a path. There are many scanning techniques and more to come possible, hence, operators should never rely on the ‘impossible to find because my address is random’ paradigm.

While in some unmanaged environments obfuscating addresses could be considered a benefit; it is a better practice to ensure that perimeter rules are actively checked and enforced and that statically configured addresses follow some logical allocation scheme for ease of operation (as simplicity always helps security).

2.1.2. Use of ULAs

Unique Local Addresses (ULAs) RFC4193 [RFC4193] are intended for scenarios where IP addresses are not globally reachable, despite formally having global scope. They must not appear in the routing system outside the administrative domain where they are considered valid. Therefore, packets with ULA source and/or destination addresses MUST be filtered at the domain boundary.

ULAs are assigned within pseudo-random /48 prefixes created as specified in RFC4193 [RFC4193]. They could be useful for infrastructure hiding as described in RFC4864 [RFC4864].

ULAs may be used for internal communication, in conjunction with globally reachable unicast addresses (GUAs) for hosts that also require external connectivity through a firewall. For this reason, no form of address translation is required in conjunction with ULAs.

Using ULAs as described here might simplify the filtering rules needed at the domain boundary, by allowing a regime in which only hosts that require external connectivity possess a globally reachable address. However, this does not remove the need for careful design of the filtering rules. Routers with ULA on their interfaces may also leak their address to the Internet when generating ICMP messages or ICMP error messages can also include ULA address as they contain a copy of the offending packet.
2.1.3. Point-to-Point Links

RFC6164 [RFC6164] in section 5.1 documents the reasons why to use a /127 for inter-router point-to-point links; notably, a /127 prevents the ping-pong attack between routers not implementing correctly RFC4443 [RFC4443]. The previous recommendation of RFC3627 [RFC3627] has been obsoleted and marked Historic by RFC6547 [RFC6547]).

Some environments are also using link-local addressing for point-to-point links. While this practice could further reduce the attack surface against infrastructure devices, the operational disadvantages need also to be carefully considered; see also RFC7404 [RFC7404].

2.1.4. Temporary Addresses - Privacy Extensions for SLAAC

Normal stateless address autoconfiguration (SLAAC) relies on the automatically generated EUI-64 address, which together with the /64 prefix makes up the global unique IPv6 address. The EUI-64 address is generated from the MAC address. Randomly generating an interface ID, as described in [RFC4941], is part of SLAAC with so-called privacy extension addresses and used to address some privacy concerns. Privacy extension addresses a.k.a. temporary addresses may help to mitigate the correlation of activities of a node within the same network, and may also reduce the attack exposure window.

As privacy extension addresses could also be used to obfuscate some malevolent activities (whether on purpose or not), it is advised in scenarios where user attribution is important to rely on a layer-2 authentication mechanism such as IEEE 802.1X [IEEE-802.1X] with the appropriate RADIUS accounting (Section 2.6.1.6) or to disable SLAAC and rely only on DHCPv6. However, in scenarios where anonymity is a strong desire (protecting user privacy is more important than user attribution), privacy extension addresses should be used. When [RFC8064] is available, the stable temporary address are probably a good balance between privacy (among multiple networks) and security/user attribution (within a network).

Using privacy extension addresses prevents the operator from building a priori host specific access control lists (ACLs). It must be noted that recent versions of Windows do not use the MAC address anymore to build the stable address but use a mechanism similar to the one described in [RFC7217], this also means that such an ACL cannot be configured based solely on the MAC address of the nodes, diminishing the value of such ACL. On the other hand, different VLANs are often used to segregate users, in this case ACL can rely on a /64 prefix per VLAN rather than a per host ACL entry.
The decision to utilize privacy extension addresses can come down to whether the network is managed versus unmanaged. In some environments full visibility into the network is required at all times which requires that all traffic be attributable to where it is sourced or where it is destined to within a specific network. This situation is dependent on what level of logging is performed. If logging considerations include utilizing accurate timestamps and logging a node’s source ports [RFC6302] then there should always exist appropriate user attribution needed to get to the source of any malware originator or source of criminal activity.

Disabling SLAAC and privacy extensions addresses can be done for most OS and for non-hacker users by sending RA messages with a hint to get addresses via DHCPv6 by setting the M-bit but also disabling SLAAC by resetting all A-bits in all prefix information options. Hackers will find a way to bypass this mechanism if not enforced at the switch/router level.

2.1.5. Privacy consideration of Addresses

The reader can learn more about privacy considerations for IPv6 addresses in RFC7721 [RFC7721].

2.1.6. DHCP/DNS Considerations

Many environments use DHCPv6 to allocate addresses to ensure auditability and traceability (but see Section 2.6.1.5). A main security concern is the ability to detect and counteract against rogue DHCP servers (Section 2.3.1).

While there are no fundamental differences with IPv4 and IPv6 security concerns about DNS, there are specific consideration in DNS64 RFC6147 [RFC6147] environments that need to be understood. Specifically the interactions and potential to interference with DNSSEC implementation need to be understood – these are pointed out in detail in Section 2.7.3.2.

2.1.7. Using a /64 per host

An interesting approach is using a /64 per host as proposed in RFC8273 [RFC8273]. This allows an easier user attribution (typically based on the host MAC address) as its /64 prefix is stable even if applications, containers within the host can change of IPv6 address within this /64.
2.2. Extension Headers

The extension headers are an important difference between IPv4 and IPv6. The packet structure does make a big difference. For instance, it’s trivial to find (in IPv4-based packets) the upper layer protocol type and protocol header, while in IPv6 it actually isn’t as the extension header chain must be parsed completely. The IANA has closed the existing empty "Next Header Types" registry to new entries and is redirecting its users to a new "IPv6 Extension Header Types" registry per RFC7045 [RFC7045].

They have also become a very controversial topic since forwarding nodes that discard packets containing extension headers are known to cause connectivity failures and deployment problems RFC7872 [RFC7872]. Understanding the role of varying extension headers is important and this section enumerates the ones that need careful consideration.

A clarification on how intermediate nodes should handle existing packets with extension headers and any extension headers that are defined in the future is found in RFC7045 [RFC7045]. The uniform TLV format to be used for defining future extension headers is described in RFC6564 [RFC6564].

It must also be noted that there is no indication in the packet whether the Next Protocol field points to an extension header or to a transport header. This may confuse some filtering rules.

There is work in progress at the IETF about filtering rules for those extension headers: [I-D.ietf-opsec-ipv6-eh-filtering] for transit routers.

2.2.1. Order and Repetition of Extension Headers

While RFC8200 [RFC8200] recommends the order and the maximum repetition of extension headers, there are still IPv6 implementations at the time of writing this document which support a non-recommended order of headers (such as ESP before routing) or an illegal repetition of headers (such as multiple routing headers). The same applies for options contained in the extension headers (see [I-D.kampanakis-6man-ipv6-eh-parsing]). In some cases, it has lead to nodes crashing when receiving or forwarding wrongly formated packets.

A firewall or any edge device able to enforce the recommended order and number of occurrences of extension headers is recommended.
2.2.2. Hop-by-Hop Options Header

The hop-by-hop options header, when present in an IPv6 packet, forces all nodes in the path to inspect this header in the original IPv6 specification RFC2460 [RFC2460]. This was of course a large avenue for a denial of service as most if not all routers cannot process this kind of packets in hardware but have to ‘punt’ this packet for software processing. Section 4.3 of the current Internet Standard for IPv6, RFC8200 [RFC8200], is more sensible to this respect as the processing of hop-by-hop options header is optional.

2.2.3. Fragment Header

The fragment header is used by the source when it has to fragment packets. RFC7112 [RFC7112] and section 4.5 of RFC8200 [RFC8200] explain why it is important to:

- firewall and security devices should drop first fragment not containing an upper-layer header;
- destination nodes should discard first fragments not containing an upper-layer header.

Else, stateless filtering could be bypassed by an hostile party. RFC6980 [RFC6980] applies the same rule to NDP and the RA-guard function described in RFC6105 [RFC6105].

2.2.4. IP Security Extension Header

The IPsec [RFC4301] [RFC4301] extension headers (AH [RFC4302] and ESP [RFC4303]) are required if IPsec is to be utilized for network level security functionality.

2.3. Link-Layer Security

IPv6 relies heavily on the Neighbor Discovery protocol (NDP) RFC4861 [RFC4861] to perform a variety of link operations such as discovering other nodes on the link, resolving their link-layer addresses, and finding routers on the link. If not secured, NDP is vulnerable to various attacks such as router/neighbor message spoofing, redirect attacks, Duplicate Address Detection (DAD) DoS attacks, etc. many of these security threats to NDP have been documented in IPv6 ND Trust Models and Threats RFC3756 [RFC3756] and in RFC6583 [RFC6583].
2.3.1. Securing DHCP

Dynamic Host Configuration Protocol for IPv6 (DHCPv6), as detailed in RFC3315 [RFC3315], enables DHCP servers to pass configuration parameters such as IPv6 network addresses and other configuration information to IPv6 nodes. DHCP plays an important role in any large network by providing robust stateful configuration and autoregistration of DNS Host Names.

The two most common threats to DHCP clients come from malicious (a.k.a. rogue) or unintentionally misconfigured DHCP servers. A malicious DHCP server is established with the intent of providing incorrect configuration information to the client to cause a denial of service attack or mount a man in the middle attack. While unintentionally, a misconfigured DHCP server can have the same impact. Additional threats against DHCP are discussed in the security considerations section of RFC3315 [RFC3315] DHCP-shield.

RFC7610 [RFC7610], DHCPv6-Shield, specifies a mechanism for protecting connected DHCPv6 clients against rogue DHCPv6 servers. This mechanism is based on DHCPv6 packet-filtering at the layer-2 device; the administrator specifies the interfaces connected to DHCPv6 servers. Of course, extension headers could be leveraged to bypass DHCPv6-Shield unless RFC7112 [RFC7112] is enforced. Another way to secure DHCPv6 would be to use the secure DHCPv6 protocol which is currently work in progress per [I-D.ietf-dhc-sedhcpv6], but, with no real deployment known by the authors of this document.

It is recommended to use DHCP-shield and to analyze the log generated by this security feature.

2.3.2. ND/RA Rate Limiting

Neighbor Discovery (ND) can be vulnerable to denial of service (DoS) attacks in which a router is forced to perform address resolution for a large number of unassigned addresses. Possible side effects of this attack preclude new devices from joining the network or even worse rendering the last hop router ineffective due to high CPU usage. Easy mitigative steps include rate limiting Neighbor Solicitations, restricting the amount of state reserved for unresolved solicitations, and clever cache/timer management.

RFC6583 [RFC6583] discusses the potential for DoS in detail and suggests implementation improvements and operational mitigation techniques that may be used to mitigate or alleviate the impact of such attacks. Here are some feasible mitigation options that can be employed by network operators today:
Ingress filtering of unused addresses by ACL, route filtering, longer than /64 prefix; These require static configuration of the addresses.

Tuning of NDP process (where supported).

Using /127 on point-to-point link per RFC6164 [RFC6164].

Additionally, IPv6 ND uses multicast extensively for signaling messages on the local link to avoid broadcast messages for on-the-wire efficiency. However, this has some side effects on wifi networks, especially a negative impact on battery life of smartphones and other battery operated devices that are connected to such networks. The following drafts are actively discussing methods to rate limit RAs and other ND messages on wifi networks in order to address this issue:

- [I-D.thubert-savi-ra-throttler]
- [I-D.chakrabarti-nordmark-6man-efficient-nd]

2.3.3. ND/RA Filtering

Router Advertisement spoofing is a well-known attack vector and has been extensively documented. The presence of rogue RAs, either intentional or malicious, can cause partial or complete failure of operation of hosts on an IPv6 link. For example, a host can select an incorrect router address which can be used as a man-in-the-middle (MITM) attack or can assume wrong prefixes to be used for stateless address configuration (SLAAC). RFC6104 [RFC6104] summarizes the scenarios in which rogue RAs may be observed and presents a list of possible solutions to the problem. RFC6105 [RFC6105] (RA-Guard) describes a solution framework for the rogue RA problem where network segments are designed around switching devices that are capable of identifying invalid RAs and blocking them before the attack packets actually reach the target nodes.

However, several evasion techniques that circumvent the protection provided by RA-Guard have surfaced. A key challenge to this mitigation technique is introduced by IPv6 fragmentation. An attacker can conceal the attack by fragmenting his packets into multiple fragments such that the switching device that is responsible for blocking invalid RAs cannot find all the necessary information to perform packet filtering in the same packet. RFC7113 [RFC7113] describes such evasion techniques, and provides advice to RA-Guard implementers such that the aforementioned evasion vectors can be eliminated.
Given that the IPv6 Fragmentation Header can be leveraged to circumvent current implementations of RA-Guard, RFC6980 [RFC6980] updates RFC4861 [RFC4861] such that use of the IPv6 Fragmentation Header is forbidden in all Neighbor Discovery messages except "Certification Path Advertisement", thus allowing for simple and effective measures to counter Neighbor Discovery attacks.

The Source Address Validation Improvements (SAVI) working group has worked on other ways to mitigate the effects of such attacks. RFC7513 [RFC7513] would help in creating bindings between a DHCPv4 RFC2131 [RFC2131] /DHCPv6 RFC3315 [RFC3315] assigned source IP address and a binding anchor RFC7039 [RFC7039] on a SAVI device. Also, RFC6620 [RFC6620] describes how to glean similar bindings when DHCP is not used. The bindings can be used to filter packets generated on the local link with forged source IP address.

It is still recommended that RA-Guard be employed as a first line of defense against common attack vectors including misconfigured hosts. The generated log should also be analyzed to act on violations.

2.3.4. 3GPP Link-Layer Security

The 3GPP link is a point-to-point like link that has no link-layer address. This implies there can only be an end host (the mobile hand-set) and the first-hop router (i.e., a GPRS Gateway Support Node (GGSN) or a Packet Gateway (PGW)) on that link. The GGSN/PGW never configures a non link-local address on the link using the advertised /64 prefix on it. The advertised prefix must not be used for on-link determination. There is no need for an address resolution on the 3GPP link, since there are no link-layer addresses. Furthermore, the GGSN/PGW assigns a prefix that is unique within each 3GPP link that uses IPv6 stateless address autoconfiguration. This avoids the necessity to perform DAD at the network level for every address built by the mobile host. The GGSN/PGW always provides an IID to the cellular host for the purpose of configuring the link-local address and ensures the uniqueness of the IID on the link (i.e., no collisions between its own link-local address and the mobile host’s one).

The 3GPP link model itself mitigates most of the known NDP-related Denial-of-Service attacks. In practice, the GGSN/PGW only needs to route all traffic to the mobile host that falls under the prefix assigned to it. As there is also a single host on the 3GPP link, there is no need to defend that IPv6 address.

See Section 5 of RFC6459 [RFC6459] for a more detailed discussion on the 3GPP link model, NDP on it and the address configuration detail.
2.3.5. SeND and CGA

SEcure Neighbor Discovery (SeND), as described in RFC3971 [RFC3971], is a mechanism that was designed to secure ND messages. This approach involves the use of new NDP options to carry public key based signatures. Cryptographically Generated Addresses (CGA), as described in RFC3972 [RFC3972], are used to ensure that the sender of a Neighbor Discovery message is the actual "owner" of the claimed IPv6 address. A new NDP option, the CGA option, was introduced and is used to carry the public key and associated parameters. Another NDP option, the RSA Signature option, is used to protect all messages relating to neighbor and Router discovery.

SeND protects against:

- Neighbor Solicitation/Advertisement Spoofing
- Neighbor Unreachability Detection Failure
- Duplicate Address Detection DoS Attack
- Router Solicitation and Advertisement Attacks
- Replay Attacks
- Neighbor Discovery DoS Attacks

SeND does NOT:

- Protect statically configured addresses
- Protect addresses configured using fixed identifiers (i.e. EUI-64)
- Provide confidentiality for NDP communications
- Compensate for an unsecured link - SEND does not require that the addresses on the link and Neighbor Advertisements correspond

However, at this time and after many years after their specifications, CGA and SeND do not have wide support from generic operating systems; hence, their usefulness is limited.

2.4. Control Plane Security

RFC6192 [RFC6192] defines the router control plane. This definition is repeated here for the reader’s convenience.
Modern router architecture design maintains a strict separation of forwarding and router control plane hardware and software. The router control plane supports routing and management functions. It is generally described as the router architecture hardware and software components for handling packets destined to the device itself as well as building and sending packets originated locally on the device. The forwarding plane is typically described as the router architecture hardware and software components responsible for receiving a packet on an incoming interface, performing a lookup to identify the packet’s IP next hop and determine the best outgoing interface towards the destination, and forwarding the packet out through the appropriate outgoing interface.

While the forwarding plane is usually implemented in high-speed hardware, the control plane is implemented by a generic processor (named router processor RP) and cannot process packets at a high rate. Hence, this processor can be attacked by flooding its input queue with more packets than it can process. The control plane processor is then unable to process valid control packets and the router can lose OSPF or BGP adjacencies which can cause a severe network disruption.

The mitigation technique is:

- To drop non-legit control packet before they are queued to the RP (this can be done by a forwarding plane ACL) and
- To rate limit the remaining packets to a rate that the RP can sustain. Protocol specific protection should also be done (for example, a spoofed OSPFv3 packet could trigger the execution of the Dijkstra algorithm, therefore the number of Dijkstra execution should be also rate limited).

This section will consider several classes of control packets:

- Control protocols: routing protocols: such as OSPFv3, BGP and by extension Neighbor Discovery and ICMP
- Management protocols: SSH, SNMP, IPfix, etc
- Packet exceptions: which are normal data packets which requires a specific processing such as generating a packet-too-big ICMP message or having the hop-by-hop options header.
2.4.1. Control Protocols

This class includes OSPFv3, BGP, NDP, ICMP.

An ingress ACL to be applied on all the router interfaces SHOULD be configured such as:

- drop OSPFv3 (identified by Next-Header being 89) and RIPng (identified by UDP port 521) packets from a non link-local address
- allow BGP (identified by TCP port 179) packets from all BGP neighbors and drop the others
- allow all ICMP packets (transit and to the router interfaces)

Note: dropping OSPFv3 packets which are authenticated by IPsec could be impossible on some routers whose ACL are unable to parse the IPsec ESP or AH extension headers.

Rate limiting of the valid packets SHOULD be done. The exact configuration obviously depends on the power of the Route Processor.

2.4.2. Management Protocols

This class includes: SSH, SNMP, syslog, NTP, etc

An ingress ACL to be applied on all the router interfaces SHOULD be configured such as:

- Drop packets destined to the routers except those belonging to protocols which are used (for example, permit TCP 22 and drop all when only SSH is used);
- Drop packets where the source does not match the security policy, for example if SSH connections should only be originated from the NOC, then the ACL should permit TCP port 22 packets only from the NOC prefix.

Rate limiting of the valid packets SHOULD be done. The exact configuration obviously depends on the power of the Route Processor.

2.4.3. Packet Exceptions

This class covers multiple cases where a data plane packet is punted to the route processor because it requires specific processing:

- generation of an ICMP packet-too-big message when a data plane packet cannot be forwarded because it is too large;
- generation of an ICMP hop-limit-expired message when a data plane packet cannot be forwarded because its hop-limit field has reached 0;
- generation of an ICMP destination-unreachable message when a data plane packet cannot be forwarded for any reason;
- processing of the hop-by-hop options header, new implementations follow section 4.3 of RFC8200 [RFC8200] where this processing is optional;
- or more specific to some router implementation: an oversized extension header chain which cannot be processed by the hardware and force the packet to be punted to the generic router CPU.

On some routers, not everything can be done by the specialized data plane hardware which requires some packets to be ‘punted’ to the generic RP. This could include for example the processing of a long extension header chain in order to apply an ACL based on layer 4 information. RFC6980 [RFC6980] and more generally RFC7112 [RFC7112] highlights the security implications of oversized extension header chains on routers and updates RFC2460 [RFC2460] such that the first fragment of a packet is required to contain the entire IPv6 header chain.

An ingress ACL cannot help to mitigate a control plane attack using those packet exceptions. The only protection for the RP is to limit the rate of those packet exceptions forwarded to the RP, this means that some data plane packets will be dropped without any ICMP messages back to the source which may cause Path MTU holes.

In addition to limiting the rate of data plane packets queued to the RP, it is also important to limit the generation rate of ICMP messages both the save the RP but also to prevent an amplification attack using the router as a reflector.

2.5. Routing Security

Routing security in general can be broadly divided into three sections:

1. Authenticating neighbors/peers
2. Securing routing updates between peers
3. Route filtering

[RFC7454] covers these sections specifically for BGP in detail.
2.5.1. Authenticating Neighbors/Peers

A basic element of routing is the process of forming adjacencies, neighbor, or peering relationships with other routers. From a security perspective, it is very important to establish such relationships only with routers and/or administrative domains that one trusts. A traditional approach has been to use MD5 HMAC, which allows routers to authenticate each other prior to establishing a routing relationship.

OSPFv3 can rely on IPsec to fulfill the authentication function. However, it should be noted that IPsec support is not standard on all routing platforms. In some cases, this requires specialized hardware that offloads crypto over to dedicated ASICs or enhanced software images (both of which often come with added financial cost) to provide such functionality. An added detail is to determine whether OSPFv3 IPsec implementations use AH or ESP-Null for integrity protection. In early implementations all OSPFv3 IPsec configurations relied on AH since the details weren’t specified in RFC5340 [RFC5340] or RFC2740 [RFC2740] that was obsoleted by the former. However, the document which specifically describes how IPsec should be implemented for OSPFv3 RFC4552 [RFC4552] specifically states that ESP-Null MUST and AH MAY be implemented since it follows the overall IPsec standards wording. OSPFv3 can also use normal ESP to encrypt the OSPFv3 payload to hide the routing information.

RFC7166 [RFC7166] (which obsoletes RFC6506 [RFC6506] changes OSPFv3’s reliance on IPsec by appending an authentication trailer to the end of the OSPFv3 packets; it does not specifically authenticate the specific originator of an OSPFv3 packet; rather, it allows a router to confirm that the packet has indeed been issued by a router that had access to the shared authentication key.

With all authentication mechanisms, operators should confirm that implementations can support re-keying mechanisms that do not cause outages. There have been instances where any re-keying cause outages and therefore the tradeoff between utilizing this functionality needs to be weighed against the protection it provides.

2.5.2. Securing Routing Updates Between Peers

IPv6 initially mandated the provisioning of IPsec capability in all nodes. However, in the updated IPv6 Nodes Requirement standard RFC6434 [RFC6434] is now a ‘SHOULD’ and no more a ‘MUST’ implement. Theoretically it is possible, and recommended, that communication between two IPv6 nodes, including routers exchanging routing information be encrypted using IPsec. In practice however, deploying
IPsec is not always feasible given hardware and software limitations of various platforms deployed, as described in the earlier section.

2.5.3. Route Filtering

Route filtering policies will be different depending on whether they pertain to edge route filtering vs internal route filtering. At a minimum, IPv6 routing policy as it pertains to routing between different administrative domains should aim to maintain parity with IPv4 from a policy perspective e.g.,

- Filter internal-use, non-globally routable IPv6 addresses at the perimeter
- Discard packets from and to bogon and reserved space (see RFC8190 [RFC8190])
- Configure ingress route filters that validate route origin, prefix ownership, etc. through the use of various routing databases, e.g., RADB. There is additional work being done in this area to formally validate the origin ASs of BGP announcements in RFC6810 [RFC6810]

Some good recommendations for filtering can be found from Team CYMRU at [CYMRU].

2.6. Logging/Monitoring

In order to perform forensic research in case of any security incident or to detect abnormal behaviors, network operators should log multiple pieces of information.

This includes:

- logs of all applications when available (for example web servers);
- use of IP Flow Information Export [RFC7011] also known as IPfix;
- use of SNMP MIB [RFC4293];
- use of the Neighbor cache;
- use of stateful DHCPv6 [RFC3315] lease cache, especially when a relay agent [RFC6221] in layer-2 switches is used;
- use of RADIUS [RFC2866] for accounting records.
Please note that there are privacy issues related to how those logs are collected, kept and safely discarded. Operators are urged to check their country legislation.

All those pieces of information will be used for:

- forensic (Section 2.6.2.1) investigations such as who did what and when?
- correlation (Section 2.6.2.3): which IP addresses were used by a specific node (assuming the use of privacy extensions addresses [RFC4941])
- inventory (Section 2.6.2.2): which IPv6 nodes are on my network?
- abnormal behavior detection (Section 2.6.2.4): unusual traffic patterns are often the symptoms of a abnormal behavior which is in turn a potential attack (denial of services, network scan, a node being part of a botnet, ...)

2.6.1. Data Sources

This section lists the most important sources of data that are useful for operational security.

2.6.1.1. Logs of Applications

Those logs are usually text files where the remote IPv6 address is stored in all characters (not binary). This can complicate the processing since one IPv6 address, 2001:db8::1 can be written in multiple ways such as:

- 2001:DB8::1 (in uppercase)
- 2001:0db8::0001 (with leading 0)
- and many other ways including the reverse DNS mapping into a FQDN (which should not be trusted).

RFC 5952 [RFC5952] explains this problem in detail and recommends the use of a single canonical format (in short use lower case and suppress leading 0). This memo recommends the use of canonical format [RFC5952] for IPv6 addresses in all possible cases. If the existing application cannot log under the canonical format, then this memo recommends the use an external program in order to canonicalize all IPv6 addresses.

For example, this perl script can be used:
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#!/usr/bin/perl -w
use strict ;
use warnings ;
use Socket ;
use Socket6 ;
my (@words, $word, $binary_address) ;
## go through the file one line at a time
while (my $line = <STDIN>) {
chomp $line;
foreach my $word (split /[\s+]/, $line) {
$binary_address = inet_pton AF_INET6, $word ;
if ($binary_address) {
print inet_ntop AF_INET6, $binary_address ;
} else {
print $word ;
}
print " " ;
}
print "\n" ;
}
2.6.1.2.

IP Flow Information Export by IPv6 Routers

IPfix [RFC7012] defines some data elements that are useful for
security:
o

in section 5.4 (IP Header fields): nextHeaderIPv6 and
sourceIPv6Address;

o

in section 5.6 (Sub-IP fields) sourceMacAddress.

Moreover, IPfix is very efficient in terms of data handling and
transport. It can also aggregate flows by a key such as
sourceMacAddress in order to have aggregated data associated with a
specific sourceMacAddress. This memo recommends the use of IPfix and
aggregation on nextHeaderIPv6, sourceIPv6Address and
sourceMacAddress.
2.6.1.3.

SNMP MIB by IPv6 Routers

RFC 4293 [RFC4293] defines a Management Information Base (MIB) for
the two address families of IP. This memo recommends the use of:
o

ipIfStatsTable table which collects traffic counters per
interface;

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2.6.1.4. Neighbor Cache of IPv6 Routers

The neighbor cache of routers contains all mappings between IPv6 addresses and data-link layer addresses. It is usually available by two means:

- the SNMP MIB (Section 2.6.1.3) as explained above;
- using NETCONF RFC6241 [RFC6241] to collect the state of the neighbor cache;
- also by connecting over a secure management channel (such as SSH) and explicitly requesting a neighbor cache dump via the Command Line Interface or any other monitoring mechanism.

The neighbor cache is highly dynamic as mappings are added when a new IPv6 address appears on the network (could be quite often with privacy extension addresses [RFC4941] or when they are removed when the state goes from UNREACH to removed (the default time for a removal per Neighbor Unreachability Detection [RFC4861] algorithm is 38 seconds for a typical host such as Windows 7). This means that the content of the neighbor cache must periodically be fetched every 30 seconds (to be on the safe side) and stored for later use.

This is an important source of information because it is trivial (on a switch not using the SAVI [RFC7039] algorithm) to defeat the mapping between data-link layer address and IPv6 address. Let us rephrase the previous statement: having access to the current and past content of the neighbor cache has a paramount value for forensic and audit trail.

Using the approach of one /64 per host (Section 2.1.7) replaces the neighbor cache dumps by a mere caching of the allocated /64 prefix when combined with strict enforcement rule on the router and switches to prevent IPv6 spoofing.

2.6.1.5. Stateful DHCPv6 Lease

In some networks, IPv6 addresses are managed by stateful DHCPv6 server [RFC3315] that leases IPv6 addresses to clients. It is indeed quite similar to DHCP for IPv4 so it can be tempting to use this DHCP lease file to discover the mapping between IPv6 addresses and data-link layer addresses as it was usually done in the IPv4 era.
It is not so easy in the IPv6 era because not all nodes will use DHCPv6 (there are nodes which can only do stateless autoconfiguration) but also because DHCPv6 clients are identified not by their hardware-client address as in IPv4 but by a DHCP Unique ID (DUID) which can have several formats: some being the data-link layer address, some being data-link layer address prepended with time information or even an opaque number which is useless for operation security. Moreover, when the DUID is based on the data-link address, this address can be of any interface of the client (such as the wireless interface while the client actually uses its wired interface to connect to the network).

If a lightweight DHCP relay agent [RFC6221] is used in the layer-2 switches, then the DHCP server also receives the Interface-ID information which could be save in order to identify the interface of the switches which received a specific leased IPv6 address. Also, if a ‘normal’ (not lightweight) relay agent adds the data-link layer address in the option for Relay Agent Remote-ID [RFC4649] or RFC6939 [RFC6939], then the DHCPv6 server can keep track of the data-link and leased IPv6 addresses.

In short, the DHCPv6 lease file is less interesting than in the IPv4 era. DHCPv6 servers that keep the relayed data-link layer address in addition to the DUID in the lease file do not suffer from this limitation.

The mapping between data-link layer address and the IPv6 address can be secured by using switches implementing the SAVI [RFC7513] algorithms. Of course, this also requires that data-link layer address is protected by using layer-2 mechanism such as [IEEE-802.1X].

2.6.1.6. RADIUS Accounting Log

For interfaces where the user is authenticated via a RADIUS [RFC2866] server, and if RADIUS accounting is enabled, then the RADIUS server receives accounting Acct-Status-Type records at the start and at the end of the connection which include all IPv6 (and IPv4) addresses used by the user. This technique can be used notably for Wi-Fi networks with Wi-Fi Protected Address (WPA) or any other IEEE 802.1X [IEEE-802.1X]wired interface on an Ethernet switch.

2.6.1.7. Other Data Sources

There are other data sources that must be kept exactly as in the IPv4 network:
o historical mapping of IPv6 addresses to users of remote access VPN;

o historical mapping of MAC address to switch interface in a wired network.

2.6.2. Use of Collected Data

This section leverages the data collected as described before (Section 2.6.1) in order to achieve several security benefits.

2.6.2.1. Forensic

The forensic use case is when the network operator must locate an IPv6 address that was present in the network at a certain time or is still currently in the network.

The source of information can be, in decreasing order, neighbor cache, DHCP lease file. Then, the procedure is:

1. based on the IPv6 prefix of the IPv6 address find the router(s) which are used to reach this prefix (assuming that anti-spoofing mechanisms are used);

2. based on this limited set of routers, on the incident time and on IPv6 address to retrieve the data-link address from live neighbor cache, from the historical data of the neighbor cache,

3. based on the incident time and on the IPv6 address, retrieve the data-link address from the DHCP lease file (Section 2.6.1.5);

4. based on the data-link layer address, look-up on which switch interface was this data-link layer address. In the case of wireless LAN, the RADIUS log should have the mapping between user identification and the MAC address. If a Configuration Management Data Base (CMDB) is used, the mapping between the data-link layer address and a switch port.

At the end of the process, the interface the host originating malicious activity or the username which was abused for malicious activity has been determined.

2.6.2.2. Inventory

RFC 7707 [RFC7707] (which obsoletes RFC 5157 [RFC5157]) is about the difficulties for an attacker to scan an IPv6 network due to the vast number of IPv6 addresses per link (and why in some case it can still be done). While the huge addressing space can sometime be perceived
as a ‘protection’, it also make the inventory task difficult in an IPv6 network while it was trivial to do in an IPv4 network (a simple enumeration of all IPv4 addresses, followed by a ping and a TCP/UDP port scan). Getting an inventory of all connected devices is of prime importance for a secure operation of a network.

There are many ways to do an inventory of an IPv6 network.

The first technique is to use the IPfix information and extract the list of all IPv6 source addresses to find all IPv6 nodes that sent packets through a router. This is very efficient but alas will not discover silent node that never transmitted such packets... Also, it must be noted that link-local addresses will never be discovered by this means.

The second way is again to use the collected neighbor cache content to find all IPv6 addresses in the cache. This process will also discover all link-local addresses. See Section 2.6.1.4.

Another way works only for local network, it consists in sending an ICMP ECHO_REQUEST to the link-local multicast address ff02::1 which is all IPv6 nodes on the network. All nodes should reply to this ECHO_REQUEST per [RFC4443].

Other techniques involve obtaining data from DNS, parsing log files, leveraging service discovery such as mDNS RFC6761 [RFC6762] and RFC6763 [RFC6763].

Enumerating DNS zones, especially looking at reverse DNS records and CNAMEs, is another common method employed by various tools. As already mentioned in RFC7707 [RFC7707], this allows an attacker to prune the IPv6 reverse DNS tree, and hence enumerate it in a feasible time. Furthermore, authoritative servers that allow zone transfers (AXFR) may be a further information source.

2.6.2.3. Correlation

In an IPv4 network, it is easy to correlate multiple logs, for example to find events related to a specific IPv4 address. A simple Unix grep command was enough to scan through multiple text-based files and extract all lines relevant to a specific IPv4 address.

In an IPv6 network, this is slightly more difficult because different character strings can express the same IPv6 address. Therefore, the simple Unix grep command cannot be used. Moreover, an IPv6 node can have multiple IPv6 addresses.
In order to do correlation in IPv6-related logs, it is advised to have all logs with canonical IPv6 addresses. Then, the neighbor cache current (or historical) data set must be searched to find the data-link layer address of the IPv6 address. Then, the current and historical neighbor cache data sets must be searched for all IPv6 addresses associated to this data-link layer address: this is the search set. The last step is to search in all log files (containing only IPv6 address in canonical format) for any IPv6 addresses in the search set.

2.6.2.4. Abnormal Behavior Detection

Abnormal behaviors (such as network scanning, spamming, denial of service) can be detected in the same way as in an IPv4 network:

- sudden increase of traffic detected by interface counter (SNMP) or by aggregated traffic from IPfix records [RFC7012];
- change of traffic pattern (number of connection per second, number of connection per host...) with the use of IPfix [RFC7012]

2.6.3. Summary

While some data sources (IPfix, MIB, switch CAM tables, logs, ...) used in IPv4 are also used in the secure operation of an IPv6 network, the DHCPv6 lease file is less reliable and the neighbor cache is of prime importance.

The fact that there are multiple ways to express in a character string the same IPv6 address renders the use of filters mandatory when correlation must be done.

2.7. Transition/Coexistence Technologies

As it is expected that network will not run in a pure IPv6-only way, the different transition mechanisms must be deployed and operated in a secure way. This section proposes operational guidelines for the most known and deployed transition techniques.

2.7.1. Dual Stack

Dual stack is often the first deployment choice for most existing network operators without an MPLS core where 6PE RFC4798 [RFC4798] is quite common. Dual stacking the network offers some advantages over other transition mechanisms. Firstly, the impact on existing IPv4 operations is reduced. Secondly, in the absence of tunnels or address translation, the IPv4 and IPv6 traffics are native (easier to observe) and should have the same network processing (path, quality...).
of service, ...). Dual stack allows you to gradually turn IPv4 operations down when your IPv6 network is ready for prime time. On the other hand, the operators have to manage two networks with the added complexities.

From an operational security perspective, this now means that you have twice the exposure. One needs to think about protecting both protocols now. At a minimum, the IPv6 portion of a dual stacked network should maintain parity with IPv4 from a security policy point of view. Typically, the following methods are employed to protect IPv4 networks at the edge:

- ACLs to permit or deny traffic
- Firewalls with stateful packet inspection

It is recommended that these ACLs and/or firewalls be additionally configured to protect IPv6 communications. Also, given the end-to-end connectivity that IPv6 provides, it is also recommended that hosts be fortified against threats. General device hardening guidelines are provided in Section 2.8.

For many years, all host operating systems have IPv6 enabled by default, so, it is possible even in an ‘IPv4-only’ network to attack layer-2 adjacent victims over IPv6 link-local address or over a global IPv6 address is rogue RA or rogue DHCPv6 addresses are provided by an attacker.

2.7.2. Transition Mechanisms

There are many tunnels used for specific use cases. Except when protected by IPsec [RFC4301], all those tunnels have a couple of security issues (most of them being described in RFC 6169 [RFC6169]):

- tunnel injection: a malevolent person knowing a few pieces of information (for example the tunnel endpoints and the used protocol) can forge a packet which looks like a legit and valid encapsulated packet that will gladly be accepted by the destination tunnel endpoint, this is a specific case of spoofing;
- traffic interception: no confidentiality is provided by the tunnel protocols (without the use of IPsec), therefore anybody on the tunnel path can intercept the traffic and have access to the clear-text IPv6 packet; combined with the absence of authentication, a man in the middle attack can also be mounted;
service theft: as there is no authorization, even a non authorized user can use a tunnel relay for free (this is a specific case of tunnel injection);

reflection attack: another specific use case of tunnel injection where the attacker injects packets with an IPv4 destination address not matching the IPv6 address causing the first tunnel endpoint to re-encapsulate the packet to the destination... Hence, the final IPv4 destination will not see the original IPv4 address but only one IPv4 address of the relay router.

bypassing security policy: if a firewall or an IPS is on the path of the tunnel, then it will probably neither inspect not detect an malevolent IPv6 traffic contained in the tunnel.

To mitigate the bypassing of security policies, it is recemended to block all default configuration tunnels by denying all IPv4 traffic matching:

- IP protocol 41: this will block ISATAP (Section 2.7.2.2), 6to4 (Section 2.7.2.7), 6rd (Section 2.7.2.3) as well as 6in4 (Section 2.7.2.1) tunnels;

- IP protocol 47: this will block GRE (Section 2.7.2.1) tunnels;

- UDP protocol 3544: this will block the default encapsulation of Teredo (Section 2.7.2.6) tunnels.

Ingress filtering [RFC2827] should also be applied on all tunnel endpoints if applicable to prevent IPv6 address spoofing.

As several of the tunnel techniques share the same encapsulation (i.e. IPv4 protocol 41) and embed the IPv4 address in the IPv6 address, there are a set of well-known looping attacks described in RFC 6324 [RFC6324], this RFC also proposes mitigation techniques.

2.7.2.1. Site-to-Site Static Tunnels

Site-to-site static tunnels are described in RFC 2529 [RFC2529] and in GRE [RFC2784]. As the IPv4 endpoints are statically configured and are not dynamic they are slightly more secure (bi-directional service theft is mostly impossible) but traffic interception and tunnel injection are still possible. Therefore, the use of IPsec [RFC4301] in transport mode and protecting the encapsulated IPv4 packets is recommended for those tunnels. Alternatively, IPsec in tunnel mode can be used to transport IPv6 traffic over a non-trusted IPv4 network.
2.7.2.2. ISATAP

ISATAP tunnels [RFC5214] are mainly used within a single administrative domain and to connect a single IPv6 host to the IPv6 network. This means that endpoints and the tunnel endpoint are usually managed by a single entity; therefore, audit trail and strict anti-spoofing are usually possible and this raises the overall security.

Special care must be taken to avoid looping attack by implementing the measures of RFC 6324 [RFC6324] and of RFC6964 [RFC6964].

IPsec [RFC4301] in transport or tunnel mode can be used to secure the IPv4 ISATAP traffic to provide IPv6 traffic confidentiality and prevent service theft.

2.7.2.3. 6rd

While 6rd tunnels share the same encapsulation as 6to4 tunnels (Section 2.7.2.7), they are designed to be used within a single SP domain, in other words they are deployed in a more constrained environment than 6to4 tunnels and have little security issues except lack of confidentiality. The security considerations (Section 12) of RFC5969 [RFC5969] describes how to secure the 6rd tunnels.

IPsec [RFC4301] for the transported IPv6 traffic can be used if confidentiality is important.

2.7.2.4. 6PE and 6VPE

Organizations using MPLS in their core can also use 6PE [RFC4798] and 6VPE RFC4659 [RFC4659] to enable IPv6 access over MPLS. As 6PE and 6VPE are really similar to BGP/MPLS IP VPN described in RFC4364 [RFC4364], the security of these networks is also similar to the one described in RFC4381 [RFC4381]. It relies on:

- Address space, routing and traffic separation with the help of VRF (only applicable to 6VPE);
- Hiding the IPv4 core, hence removing all attacks against P-routers;
- Securing the routing protocol between CE and PE, in the case of 6PE and 6VPE, link-local addresses (see [RFC7404]) can be used and as these addresses cannot be reached from outside of the link, the security of 6PE and 6VPE is even higher than the IPv4 BGP/MPLS IP VPN.
2.7.2.5. DS-Lite

DS-lite is more a translation mechanism and is therefore analyzed further (Section 2.7.3.3) in this document.

2.7.2.6. Teredo

Teredo tunnels [RFC4380] are mainly used in a residential environment because that can easily traverse an IPv4 NAT-PT device thanks to its UDP encapsulation and they connect a single host to the IPv6 Internet. Teredo shares the same issues as other tunnels: no authentication, no confidentiality, possible spoofing and reflection attacks.

IPsec [RFC4301] for the transported IPv6 traffic is recommended.

The biggest threat to Teredo is probably for IPv4-only network as Teredo has been designed to easily traverse IPv4 NAT-PT devices which are quite often co-located with a stateful firewall. Therefore, if the stateful IPv4 firewall allows unrestricted UDP outbound and accept the return UDP traffic, then Teredo actually punches a hole in this firewall for all IPv6 traffic to the Internet and from the Internet. While host policies can be deployed to block Teredo in an IPv4-only network in order to avoid this firewall bypass, it would be more efficient to block all UDP outbound traffic at the IPv4 firewall if deemed possible (of course, at least port 53 should be left open for DNS traffic).

Teredo is now mostly never used and it is no more automated in most environment, so, it is less of a threat.

2.7.2.7. 6to4

6to4 tunnels [RFC3056] require a public routable IPv4 address in order to work correctly. They can be used to provide either one IPv6 host connectivity to the IPv6 Internet or multiple IPv6 networks connectivity to the IPv6 Internet. The 6to4 relay is usually the anycast address defined in RFC3068 [RFC3068] which has been deprecated by RFC7526 [RFC7526], and is no more used by recent Operating Systems. Some security considerations are explained in RFC3694 [RFC3964].

RFC6343 [RFC6343] points out that if an operator provides well-managed servers and relays for 6to4, non-encapsulated IPv6 packets will pass through well-defined points (the native IPv6 interfaces of those servers and relays) at which security mechanisms may be applied. Client usage of 6to4 by default is now discouraged, and significant precautions are needed to avoid operational problems.
2.7.2.8. Mapping of Address and Port

With the encapsulation and translation versions of mapping of Address and Port (MAP-E [RFC7597] and MAP-T [RFC7599]), the access network is purely an IPv6 network and MAP protocols are used to give IPv4 hosts on the subscriber network, access to IPv4 hosts on the Internet. The subscriber router does stateful operations in order to map all internal IPv4 addresses and layer-4 ports to the IPv4 address and the set of layer-4 ports received through MAP configuration process. The SP equipment always does stateless operations (either decapsulation or stateless translation). Therefore, as opposed to Section 2.7.3.3 there is no state-exhaustion DoS attack against the SP equipment because there is no state and there is no operation caused by a new layer-4 connection (no logging operation).

The SP MAP equipment MUST implement all the security considerations of [RFC7597]; notably, ensuring that the mapping of the IPv4 address and port are consistent with the configuration. As MAP has a predictable IPv4 address and port mapping, the audit logs are easier to manager.

2.7.3. Translation Mechanisms

Translation mechanisms between IPv4 and IPv6 networks are alternative coexistence strategies while networks transition to IPv6. While a framework is described in [RFC6144] the specific security considerations are documented in each individual mechanism. For the most part they specifically mention interference with IPsec or DNSSEC deployments, how to mitigate spoofed traffic and what some effective filtering strategies may be.

2.7.3.1. Carrier-Grade Nat (CGN)

Carrier-Grade NAT (CGN), also called NAT444 CGN or Large Scale NAT (LSN) or SP NAT is described in [RFC6264] and is utilized as an interim measure to prolong the use of IPv4 in a large service provider network until the provider can deploy and effective IPv6 solution. [RFC6598] requested a specific IANA allocated /10 IPv4 address block to be used as address space shared by all access networks using CGN. This has been allocated as 100.64.0.0/10.

Section 13 of [RFC6269] lists some specific security-related issues caused by large scale address sharing. The Security Considerations section of [RFC6598] also lists some specific mitigation techniques for potential misuse of shared address space. Some Law Enforcement Agencies have identified CGN as impeding their cyber-crime investigations (for example Europol press release on CGN [europol-cgn]).
RFC7422 [RFC7422] suggests the use of deterministic address mapping in order to reduce logging requirements for CGN. The idea is to have an algorithm mapping back and forth the internal subscriber to public ports.

2.7.3.2. NAT64/DNS64

Stateful NAT64 translation [RFC6146] allows IPv6-only clients to contact IPv4 servers using unicast UDP, TCP, or ICMP. It can be used in conjunction with DNS64 [RFC6147], a mechanism which synthesizes AAAA records from existing A records. There is also a stateless NAT64 [RFC6145] which is similar for the security aspects with the added benefit of being stateless, so, less prone to a state exhaustion attack.

The Security Consideration sections of [RFC6146] and [RFC6147] list the comprehensive issues. A specific issue with the use of NAT64 is that it will interfere with most IPsec deployments unless UDP encapsulation is used. DNS64 has an incidence on DNSSEC see section 3.1 of [RFC7050].

2.7.3.3. DS-Lite

Dual-Stack Lite (DS-Lite) [RFC6333] is a transition technique that enables a service provider to share IPv4 addresses among customers by combining two well-known technologies: IP in IP (IPv4-in-IPv6) and Network Address and Port Translation (NAPT).

Security considerations with respect to DS-Lite mainly revolve around logging data, preventing DoS attacks from rogue devices (as the AFTR function is stateful) and restricting service offered by the AFTR only to registered customers.

Section 11 of [RFC6333] describes important security issues associated with this technology.

2.8. General Device Hardening

There are many environments which rely too much on the network infrastructure to disallow malicious traffic to get access to critical hosts. In new IPv6 deployments it has been common to see IPv6 traffic enabled but none of the typical access control mechanisms enabled for IPv6 device access. With the possibility of network device configuration mistakes and the growth of IPv6 in the overall Internet it is important to ensure that all individual devices are hardened against miscreant behavior.
The following guidelines should be used to ensure appropriate hardening of the host, be it an individual computer or router, firewall, load-balancer, server, etc device.

- Restrict access to the device to authorized individuals
- Monitor and audit access to the device
- Turn off any unused services on the end node
- Understand which IPv6 addresses are being used to source traffic and change defaults if necessary
- Use cryptographically protected protocols for device management if possible (SCP, SNMPv3, SSH, TLS, etc)
- Use host firewall capabilities to control traffic that gets processed by upper layer protocols
- Use virus scanners to detect malicious programs

3. Enterprises Specific Security Considerations

Enterprises generally have robust network security policies in place to protect existing IPv4 networks. These policies have been distilled from years of experiential knowledge of securing IPv4 networks. At the very least, it is recommended that enterprise networks have parity between their security policies for both protocol versions.

Security considerations in the enterprise can be broadly categorized into two sections - External and Internal.

3.1. External Security Considerations:

The external aspect deals with providing security at the edge or perimeter of the enterprise network where it meets the service providers network. This is commonly achieved by enforcing a security policy either by implementing dedicated firewalls with stateful packet inspection or a router with ACLs. A common default IPv4 policy on firewalls that could easily be ported to IPv6 is to allow all traffic outbound while only allowing specific traffic, such as established sessions, inbound (see also [RFC6092]). Here are a few more things that could enhance the default policy:

- Filter internal-use IPv6 addresses at the perimeter
o Discard packets from and to bogon and reserved space, see also [CYMRU]

o Accept certain ICMPv6 messages to allow proper operation of ND and PMTUD, see also [RFC4890]

o Filter specific extension headers by accepting only the required ones (white list approach) such as ESP, AH (not forgetting the required transport layers: ICMP, TCP, UDP, ...) , where possible at the edge and possibly inside the perimeter; see also [I-D.gont-opsec-ipv6-eh-filtering]

o Filter packets having an illegal IPv6 headers chain at the perimeter (and possible inside as well), see Section 2.2

o Filter unneeded services at the perimeter

o Implement anti-spoofing

o Implement appropriate rate-limiters and control-plane policers

3.2. Internal Security Considerations:

The internal aspect deals with providing security inside the perimeter of the network, including the end host. The most significant concerns here are related to Neighbor Discovery. At the network level, it is recommended that all security considerations discussed in Section 2.3 be reviewed carefully and the recommendations be considered in-depth as well.

As mentioned in Section 2.6.2, care must be taken when running automated IPv6-in-IP4 tunnels.

Hosts need to be hardened directly through security policy to protect against security threats. The host firewall default capabilities have to be clearly understood, especially 3rd party ones which can have different settings for IPv4 or IPv6 default permit/deny behavior. In some cases, 3rd party firewalls have no IPv6 support whereas the native firewall installed by default has it. General device hardening guidelines are provided in Section 2.8

It should also be noted that many hosts still use IPv4 for transport for things like RADIUS, TACACS+, SYSLOG, etc. This will require some extra level of due diligence on the part of the operator.
4. Service Providers Security Considerations

4.1. BGP

The threats and mitigation techniques are identical between IPv4 and IPv6. Broadly speaking they are:

- Authenticating the TCP session;
- TTL security (which becomes hop-limit security in IPv6);
- Prefix Filtering.

These are explained in more detail in section Section 2.5.

4.1.1. Remote Triggered Black Hole Filtering

RTBH [RFC5635] works identically in IPv4 and IPv6. IANA has allocated 100::/64 as discard prefix RFC6666 [RFC6666].

4.2. Transition Mechanism

SP will typically use transition mechanisms such as 6rd, 6PE, MAP, DS-Lite which have been analyzed in the transition Section 2.7.2 section.

4.3. Lawful Intercept

The Lawful Intercept requirements are similar for IPv6 and IPv4 architectures and will be subject to the laws enforced in varying geographic regions. The local issues with each jurisdiction can make this challenging and both corporate legal and privacy personnel should be involved in discussions pertaining to what information gets logged and what the logging retention policies will be.

The target of interception will usually be a residential subscriber (e.g. his/her PPP session or physical line or CPE MAC address). With the absence of NAT on the CPE, IPv6 has the provision to allow for intercepting the traffic from a single host (a /128 target) rather than the whole set of hosts of a subscriber (which could be a /48, a /60 or /64).

In contrast, in mobile environments, since the 3GPP specifications allocate a /64 per device, it may be sufficient to intercept traffic from the /64 rather than specific /128’s (since each time the device powers up it gets a new IID).
A sample architecture which was written for informational purposes is found in [RFC3924].

5. Residential Users Security Considerations

The IETF Homenet working group is working on how IPv6 residential network should be done; this obviously includes operational security considerations; but, this is still work in progress.

Residential users have usually less experience and knowledge about security or networking. As most of the recent hosts, smartphones, tablets have all IPv6 enabled by default, IPv6 security is important for those users. Even with an IPv4-only ISP, those users can get IPv6 Internet access with the help of Teredo tunnels. Several peer-to-peer programs (notably Bittorrent) support IPv6 and those programs can initiate a Teredo tunnel through the IPv4 residential gateway, with the consequence of making the internal host reachable from any IPv6 host on the Internet. It is therefore recommended that all host security products (personal firewall, ...) are configured with a dual-stack security policy.

If the Residential Gateway has IPv6 connectivity, [RFC7084] (which obsoletes [RFC6204]) defines the requirements of an IPv6 CPE and does not take position on the debate of default IPv6 security policy as defined in [RFC6092]:

- outbound only: allowing all internally initiated connections and block all externally initiated ones, which is a common default security policy enforced by IPv4 Residential Gateway doing NAT-PT but it also breaks the end-to-end reachability promise of IPv6. [RFC6092] lists several recommendations to design such a CPE;

- open/transparent: allowing all internally and externally initiated connections, therefore restoring the end-to-end nature of the Internet for the IPv6 traffic but having a different security policy for IPv6 than for IPv4.

[RFC6092] REC-49 states that a choice must be given to the user to select one of those two policies.

There is also an alternate solution which has been deployed notably by Swisscom: open to all outbound and inbound connections at the exception of an handful of TCP and UDP ports known as vulnerable.
6. Further Reading

There are several documents that describe in more details the security of an IPv6 network; these documents are not written by the IETF but are listed here for your convenience:

1. Guidelines for the Secure Deployment of IPv6 [NIST]

7. Acknowledgements

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8. IANA Considerations

This memo includes no request to IANA.

9. Security Considerations

This memo attempts to give an overview of security considerations of operating an IPv6 network both in an IPv6-only network and in utilizing the most widely deployed IPv4/IPv6 coexistence strategies.

10. References

10.1. Normative References


10.2. Informative References


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Separating Crypto Negotiation and Communication
draft-kuehlewind-taps-crypto-sep-00

Abstract

Due to the latency involved in connection setup and security handshakes, there is an increasing deployment of cryptographic session resumption mechanisms. While cryptographic context and endpoint capabilities need to be be known before encrypted application data can be sent, there is otherwise no technical constraint that the crypto handshake must be performed on the same transport connection. This document recommends a logical separation between the mechanism(s) used to negotiate capabilities and set up encryption context (handshake protocol), the application of encryption and authentication state to data (record protocol), and the associated transport connection(s).

Status of This Memo

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Secure transport protocols are generally composed of three pieces:

1. A transport protocol to control the transfer of data.
2. A record protocol to frame, encrypt and/or authenticate data.
3. A handshake protocol to negotiate cryptographic secrets.

For ease of deployment and standardization, among other reasons, these constituents are often tightly coupled. For example, in TLS [RFC5246], the handshake protocol depends on the record protocol, and vice versa. However, more recent transport protocols such as QUIC [I-D.ietf-quic-tls] keep these pieces separate. QUIC uses TLS to negotiate secrets, and _exports_ those secrets to encrypt packets directly.
Separating these pieces is important, as new secure transport protocols increasingly rely on session resumption mechanisms where cryptographic context can be resumed to transmit application data with the first packet without delay for connection setup and negotiation. In the case where there is no cryptographic context available when an application expresses the need to transmit data to a certain endpoint, it must first run the handshake protocol on a transport connection before being able to transmit application data. If the handshake protocol can be separated from the other components, then it can use another transport connection to establish secrets without blocking the application’s main transport connection. This also opens up the possibility to run the handshake protocol well in advance of the need to send application data, to avoid unnecessary delays. For example, a client system could maintain a database of endpoints it is likely to communicate with, and establish keying material with a handshake protocol at periodic intervals to ensure fresh keys for new transport connections.

[I-D.moskowitz-sse] proposes a similar approach. However while [I-D.moskowitz-sse] proposes a new protocol to negotiate and maintain long-term cryptographic sessions, this document relies on the use of existing protocols and only discusses requirements for the evolution of these protocols and exchange of information within one endpoint locally.

2. Terminology

- **Transport Protocol**: A protocol that can transport messages between two endpoints. This may represent the service offered to applications to allow them to send and receive data before encryption; and also represent the protocol that can transmit handshake data and encrypted records.

- **Handshake Protocol**: A protocol that can validate and authenticate endpoints, encrypt and authenticate its negotiation, and ultimately generate keying material.

- **Record Protocol**: A protocol that can use keying material to transform messages. A record will generally add a frame around application data, and authenticate and/or encrypt the data.

- **Keying Material**: One or more pre-shared keys that can be used to encrypt and authenticate data, generated by a handshake protocol and used by a record protocol.
3. Protocol Interfaces

In traditional models in which the protocols are not separated out into the three elements of handshake, record, and transport protocols, there are two basic approaches to the interactions:

1. The transport protocol provides data to the security protocol and gets back an encrypted version of the data to be sent (handshake and record protocols are combined)

2. The security protocol provides keying material to the transport protocol, and the transport protocol is responsible for encrypting data (transport and record protocols are combined)

By teasing apart all three portions as separate protocols, there end up being six interface points:

Application Data

```
+----V----+      (1)      +---------------+
|               |    Transport   |                |   Handshake   |
|               +----------------+    (2)    +-----+-----^---+
|                      (6)                        (3) (4)
|                      +---------------+      (5)
|                      |    Record     |           |
|---------------------->               +-----------+
|                              +---------------+
```

Figure 1: Secure Transport Protocol Components and Interactions

1. A transport protocol depends upon a handshake protocol to establish keying material to protect application data being sent through the transport. The main interface it relies upon is starting the handshake, or ensuring that the material is ready.

2. A handshake protocol depends upon a transport protocol in order to send and receive negotiation messages with the remote peer.

3. A handshake protocol sends its keying material and cryptographic context to the record protocol to use
4. A record protocol may signal state expiration events to a handshake protocol

5. A transport protocol uses a record protocol to send and receive application data

6. A record protocol uses a transport protocol to send and receive encrypted data

3.1. Handshake-Transport Interface

Note that for the purposes of this interface description, it is assumed that the application is primarily interacting with the transport protocol, and thus the handshake protocol interacts with the application primarily through the abstraction of the transport protocol.

- Start negotiation: The interface MUST provide an indication to start the protocol handshake for key negotiation, and have a way to be notified when the handshake is complete.

- Identity constraints: The interface MUST allow the application to constrain the identities that it will accept a connection to, such as the hostname it expects to be provided in certificate SAN.

- Local identities: The interface MUST allow the local identity to be set via a raw private key or interface to one to perform cryptographic operations such as signing and decryption.

- State changes: The interface SHOULD provide a way for the transport to be notified of important state changes during the protocol execution and session lifetime, e.g., when the handshake begins, ends, or when a key update occurs.

- Validation: The interface MUST provide a way for the application to participate in the endpoint authentication and validation, which can either be specified as parameters to define how the peer’s authentication can be validated, or when the protocol provides the authentication information for the application to inspect directly.

- Caching domain and lifetime: The application SHOULD be able to specify the instances of the protocol that can share cached keys, as well as the lifetime of cached resources.

- The protocol SHOULD allow applications to negotiate application protocols and related information.
The protocol SHOULD allow applications to specify negotiable cryptographic algorithm suites.

The protocol SHOULD expose the peer’s identity information.

3.2. Handshake-Record Interface

- Key export: The interface MUST provide a way to export keying material from a handshake protocol to a record protocol with well-defined cryptographic properties, e.g., "forward-secure" or "perfectly forward secure"

- Key lifetime and rotation: The interface MUST provide a way for the handshake protocol to define key lifetime bounds in terms of _time_ or _bytes encrypted_ and, additionally, provide a way to forcefully update cryptographic session keys at will. The record protocol MUST be able to signal back to the handshake protocol that a lifetime has been reached and that rotation is required. These values SHOULD be configurable by the application.

3.3. Transport-Record Interface

- Transform data: The interface MUST provide a way to send raw application data from the transport protocol to a record protocol to transform it based on the keying material. This data is then sent out by the transport protocol. The same applies for inbound data, in which inbound transport data is transformed by the record protocol into raw application data.

- Reliability: The transport MUST specify if messages are transmitted reliable and in order.

- Maximum message size (optional): The transport may specify a maximum message size for the encrypted data if e.g. a datagram transport is used

4. Existing Mappings

In this section we document existing mappings between common transport security protocols and the three components described in Section I.

- TLS/DTLS: TLS [RFC5246] and DTLS [RFC6347] is a combination of a handshake and record protocol, with a dependency on some underlying transport.
The emerging QUIC standard is decomposed into the three pieces outlined in Section I [I-D.ietf-quic-tls]. TLS is used as the handshake protocol running on a dedicated QUIC stream, a QUIC-specific record protocol encrypts and encapsulates stream frames, and the main QUIC component handles the transport of these frames.
IKEv2 + ESP: IKEv2 [RFC7296] is a handshake protocol commonly used to establish keys for use in IPsec (often VPN) deployments. It is already a distinct protocol from its commonly paired record protocol, which is ESP [RFC4303]. ESP encrypts and authenticates IP datagrams, and sends them as datagrams over a transport mechanism such, e.g., IP or UDP.

5. Benefits of Separation
5.1. Reducing Connection Latency

One of the clearest benefits of separating the handshake protocol from the record protocol is that the handshake can be performed out-of-band from the application’s data transfer. This should essentially reduce the number of RTTs required before being able to send data by the full length of the handshake (which is commonly 1 or 2 RTTs in the best cases for TLS 1.2 and IKEv2, potentially more if cookie challenges or extended authentication are required).

To avoid long-lived transport connections that wouldn’t be actively used, and thus would be vulnerable to timeouts on NATs or firewalls, an obvious approach to separating the handshake and record protocols is to use different transport connections for the early handshake and the data transfer. However, this approach of using separate connections will not always save RTTs if the handshake and data transfer are back-to-back. Each connection may require its own transport protocol handshake, and if the data transfer must wait for two transport protocols to establish and the cryptographic handshake to be finished before sending, then it may experience higher latency. Implementations SHOULD avoid this by either allowing the handshake and record protocols to share a single transport connection or open two connections in parallel when the handshake protocol has not pre-fetched keys. Latency benefits, however, can even be achieved when ensuring that this scenario does not occur by always having the handshake protocol refresh the keys whenever old ones are near expiry.

5.2. Protocol Flexibility

Separation of the handshake, record, and transport protocols also allows for more flexible composition of protocols with one another. If a deployment uses a handshake protocol like TLS, which requires a stream-based transport protocol like TCP, separation of protocols will allow it to use the resulting keys for record protocols that run on datagram transport protocols like UDP.

This flexibility may be useful for implementations that are optimizing for packet size by choosing minimal/lightweight record protocols, while being able to use commonly supported handshake protocols like TLS. One example here is the approach of a VPN tunnel that uses ESP or Diet-ESP [I-D.mglt-ipsecme-diet-esp] to encrypt datagrams, but uses TLS for establishing keys.
5.3. Protocol Capability Negotiation

Enabling the use of a different transport protocol for the actual data transmission than for the cryptographic handshakes opens also the possibility to negotiate protocol capabilities for the data transmission. For TLS, usually TCP is the appropriate transport protocol to use, as it is also widely supported by endpoints. Allowing an endpoint to indicate the support of other, new transport protocols within the TCP connection that is used for the handshake, provides a dynamic transition path to enable easy deployment of new protocols.

6. IANA Considerations

This document has on request to IANA.

7. Security Considerations

(editor’s note: this section will be added later. However, this document discusses the use of cryptographic context for transport connections and as such it has security relevant consideration within the whole document.)

8. Acknowledgments

This work is partially supported by the European Commission under Horizon 2020 grant agreement no. 688421 Measurement and Architecture for a Middleboxed Internet (MAMI), and by the Swiss State Secretariat for Education, Research, and Innovation under contract no. 15.0268. This support does not imply endorsement.

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Abstract

Secure transport protocols often consist of three logically distinct components: transport, control (handshake), and record protection. Typically, such a protocol contains a single module that is responsible for all three functions. However, in many cases, this coupling is unnecessary. For example, while cryptographic context and endpoint capabilities need to be known before encrypted application data can be sent on a specific transport connection, there is otherwise no technical constraint that a cryptographic handshake must be performed on said connection. This document recommends a logical separation between transport, control, and record components of secure transport protocols. We compare existing protocols such as Transport Layer Security, QUIC, and IKEv2+ESP in the context of this logical separation.

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This Internet-Draft will expire on January 1, 2019.
1. Introduction

Secure transport protocols are generally composed of three pieces:

1. A transport protocol to handle the transfer of data.
2. A record protocol to frame, encrypt and/or authenticate data
3. A control protocol to perform cryptographic handshakes, negotiate shared secrets, and maintain state during the lifetime of cryptographic session including session resumption and key

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refreshment. (In the context of TLS, the control protocol is called the handshake protocol.)

For ease of deployment and standardization, among other reasons, these constituents are often tightly coupled. For example, in TLS [RFC5246], the control protocol depends on the record protocol, and vice versa. However, more recent transport protocols such as QUIC [I-D.ietf-quic-tls] keep these pieces separate. For example, QUIC uses TLS to negotiate secrets, and exports those secrets to encrypt packets independent of TLS.

Separating these pieces is important, as new secure transport protocols increasingly rely on session resumption mechanisms where cryptographic context can be resumed to transmit application data with the first packet without delay for connection setup and negotiation. In the case where there is no cryptographic context available when an application expresses the need to transmit data to a certain endpoint, it must first run the control protocol on a transport connection before being able to transmit application data. If the control protocol can be separated from the other components, then it can use another transport connection to establish secrets without blocking the application’s main transport connection. This also opens up the possibility to run the control protocol well in advance of the need to send application data, to avoid unnecessary delays. For example, a client system could maintain a database of endpoints it is likely to communicate with, and establish keying material with a control protocol at periodic intervals to ensure fresh keys for new transport connections.

[I-D.moskowitz-sse] proposes a similar approach. However while [I-D.moskowitz-sse] proposes a new protocol to negotiate and maintain long-term cryptographic sessions, this document relies on the use of existing protocols and only discusses requirements for the evolution of these protocols and exchange of information within one endpoint locally.

2. Terminology

- **Transport Protocol**: A protocol that can transport messages between two endpoints. This may represent the service offered to applications to allow them to send and receive data before encryption; and also represent the protocol that can transmit control data and encrypted records.

- **Control Protocol**: A protocol that performs a cryptographic handshake and, in addition, can validate and authenticate endpoints, encrypt and authenticate its negotiation, and ultimately generate keying material.
o Record Protocol: A protocol that can use keying material to transform messages. A record will generally add a frame around application data, and authenticate and/or encrypt the data.

o Keying Material: A shared secret from which pre-shared keys can be derived and subsequently used to encrypt and authenticate data, generated by a control protocol and used by a record protocol.

3. Protocol Interfaces

In traditional models in which the protocols are not separated out into the three elements of control, record, and transport protocols, there are two basic approaches to the interactions:

1. The transport protocol provides data to the security protocol and gets back an encrypted version of the data to be sent (control and record protocols are combined).

2. The security protocol provides keying material to the transport protocol, and the transport protocol is responsible for encrypting data (transport and record protocols are combined).

By teasing apart all three portions as separate protocols, there end up being six interface points:

```
Application Data
     ^
    (1) +---------------->                       +---------------+
       |     Control                   Transport     |
       |                                 +-------------+    (3)
       |                                         (4)
       |                                         +---------+
       |                                         | (5) Record
       +-------------+----------------->             +----------------+
```

Figure 1: Secure Transport Protocol Components and Interactions

1. A transport protocol depends upon a control protocol to establish keying material to protect application data being sent through the transport. The main interface it relies upon is starting the
control channel, or handshake, or ensuring that the material is ready.

2. A control protocol depends upon a transport protocol in order to send and receive negotiation messages with the remote peer.

3. A control protocol sends its keying material and cryptographic context to the record protocol to use.

4. A record protocol may signal state expiration events to a control protocol.

5. A transport protocol uses a record protocol to send and receive application data.

6. A record protocol uses a transport protocol to send and receive encrypted data.

3.1. Control-Transport Interface

Note that for the purposes of this interface description, it is assumed that the application is primarily interacting with the transport protocol, and thus the control protocol interacts with the application primarily through the abstraction of the transport protocol. Since security protocol interfaces often require pre-connection and active behavior on behalf of clients, we further categorize the following interfaces based on whether they are meant for passive configuration or active control.

3.1.1. Passive Configuration Interface

- Start negotiation: The interface MUST provide an indication to start the protocol handshake for key negotiation, and have a way to be notified when the handshake is complete.

- Identity constraints: The interface MUST allow the application to constrain the identities that it will accept a connection to, such as the hostname it expects to be provided in certificate SAN.

- Local identities: The interface MUST allow the local identity to be set via a raw private key or interface to one to perform cryptographic operations such as signing and decryption.

- Caching domain and lifetime: The application SHOULD be able to specify the instances of the protocol that can share cached keys, as well as the lifetime of cached resources.
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- Pre-shared keying material: The application SHOULD be able to specify
  pre-share keying material to use to bootstrap connections. The control
  protocol can pass this directly to the record protocol for use.

- The protocol SHOULD allow applications to negotiate application
  protocols and related information.

- The protocol SHOULD allow applications to specify negotiable
  cryptographic algorithm suites.

3.1.2. Active Control and Introspection Interface

- State changes: The interface SHOULD provide a way for the transport to
  be notified of important state changes during the protocol execution
  and session lifetime, e.g., when the handshake begins, ends, or when a
  key update occurs.

- Validation: The interface MUST provide a way for the application to
  participate in the endpoint authentication and validation, which can
  either be specified as parameters to define how the peer’s
  authentication can be validated, or when the protocol provides the
  authentication information for the application to inspect directly.

- The protocol SHOULD expose the peer’s identity information during
  and after connection establishment.

3.2. Control-Record Interface

- Key export: The interface MUST provide a way to export keying
  material from a control protocol to a record protocol with well-
  defined cryptographic properties, e.g., "forward-secure."

- Key lifetime and rotation: The interface MUST provide a way for
  the control protocol to define key lifetime bounds in terms of
  _time_ or _bytes encrypted_ and, additionally, provide a way to
  forcefully update cryptographic session keys at will. The record
  protocol MUST be able to signal back to the control protocol that
  a lifetime has been reached and that rotation is required. These
  values SHOULD be configurable by the application.

3.3. Transport-Record Interface

- Transform data: The interface MUST provide a way to send raw
  application data from the transport protocol to a record protocol
  to transform it based on the keying material. This data is then
  sent out by the transport protocol. The same applies for inbound
data, in which inbound transport data is transformed by the record protocol into raw application data.

- Reliability: The transport MUST specify if messages are transmitted reliable and in order.

- Maximum message size (optional): The transport may specify a maximum message size for the encrypted data if e.g. a datagram transport is used

4. Existing Mappings

In this section we document existing mappings between common transport security protocols and the three components described in Section I.

- TLS/DTLS: TLS [RFC5246] and DTLS [RFC6347] is a combination of a control (handshake) and record protocol, with a dependency on some underlying transport.

- QUIC + TLS: The emerging QUIC standard is decomposed into the three pieces outlined in Section I [I-D.ietf-quic-tls]. TLS is used as the control protocol running on a dedicated QUIC stream, a QUIC-specific record protocol encrypts and encapsulates stream frames, and the main QUIC component handles the transport of these frames.
IKEv2 + ESP: IKEv2 [RFC7296] is a control protocol commonly used to establish keys for use in IPsec (often VPN) deployments. It is already a distinct protocol from its commonly paired record protocol, which is ESP [RFC4303]. ESP encrypts and authenticates IP datagrams, and sends them as datagrams over a transport mechanism such, e.g., IP or UDP.

OpenVPN [OpenVPN]: OpenVPN consists of two separate stacks - one for TLS, which is used for key exchange and derivation, and the other as an interface to tunnel IP packets over UDP. A common multiplexing layer is used to send TLS and OpenVPN framed packets over an unreliable transport layer. OpenVPN adds a reliability
layer to TLS to ensure packets are sent and processed in order. Running over TCP naturally provides this reliability. After the TLS connection finishes, OpenVPN extracts encryption and authentication keys from TLS, via the PRF, and uses them to encrypt and authenticate IP packets. Packets are framed using a simple length-type-value envelope, wherein the type specifies the contents of the packet, e.g., channel control (TLS ciphertext) bytes.

DTLS-SRTP: DTLS [RFC5764] is commonly used as a way to perform mutual authentication and key agreement for SRTP [RFC5763]. (Here, certificates marshal public keys between endpoints. Thus, self-signed certificates may be used if peers do not mutually trust one another, as is common on the Internet.) When DTLS is used, certificate fingerprints are transmitted out-of-band using SIP. Peers typically verify that DTLS-offered certificates match that which are offered over SIP. This prevents active attacks on RTP, but not on the signaling (SIP or WebRTC) channel.
5. Benefits of Separation

5.1. Reducing Connection Latency

One of the clearest benefits of separating the control protocol from the record protocol is that the cryptographic handshake can be performed out-of-band from the application’s data transfer. This should essentially reduce the number of RTTs required before being able to send data by the full length of the handshake (which is commonly 1 or 2 RTTs in the best cases for TLS 1.2 and IKEv2, potentially more if cookie challenges or extended authentication are required).

To avoid long-lived transport connections that wouldn’t be actively used, and thus would be vulnerable to timeouts on NATs or firewalls, an obvious approach to separating the control and record protocols is to use different transport connections for the early handshake and the data transfer. However, this approach of using separate connections will not always save RTTs if the cryptographic handshake and data transfer are back-to-back. Each connection may require its own transport protocol handshake, and if the data transfer must wait for two transport protocols to establish and the cryptographic handshake to be finished before sending, then it may experience higher latency. Implementations SHOULD avoid this by either allowing the control and record protocols to share a single transport connection or open two connections in parallel when the control protocol has not pre-fetched keys. Latency benefits, however, can even be achieved when ensuring that this scenario does not occur by always having the control protocol refresh the keys whenever old ones are near expiry.
5.2. Protocol Flexibility

Separation of the control, record, and transport protocols also allows for more flexible composition of protocols with one another. If a deployment uses a control protocol like TLS, which requires a stream-based transport protocol like TCP, separation of protocols will allow it to use the resulting keys for record protocols that run on datagram transport protocols like UDP.

This flexibility may be useful for implementations that are optimizing for packet size by choosing minimal/lightweight record protocols, while being able to use commonly supported control protocols like TLS. One example here is the approach of a VPN tunnel that uses ESP or Diet-ESP [I-D.mglt-ipsecme-diet-esp] to encrypt datagrams, but uses TLS for establishing keys. This design is similar to that used by OpenVPN [OpenVPN], as described above.

5.3. Protocol Capability and Upgrade Negotiation

Enabling the use of a different transport protocol for the actual data transmission than for the cryptographic handshakes opens also the possibility to negotiate protocol capabilities for the data transmission. For TLS, usually TCP is the appropriate transport protocol to use, as it is also widely supported by endpoints. Allowing an endpoint to indicate the support of other, new transport protocols within the TCP connection that is used for the cryptographic handshake, provides a dynamic transition path to enable easy deployment of new protocols. Another example is providing an upgrade path from TCP+TLS to QUIC. If TLS could negotiate the use of other transport layers, such as QUIC, applications could perform an abbreviated upgrade from TCP+TLS connections to QUIC, i.e., without doing a full QUIC handshake.

6. Transport Service Architecture Integration

The Transport Services Architecture ([I-D.ietf-taps-arch]) describes a system that can provide transport security functionality behind a common interface. Such systems and their APIs provide applications with the ability to establish connections for sending and receiving data. The lifetime of a connection is comprised of a pre-establishment configuration stage, established (connected) stage, and terminated stage. Pre-establishment properties configured include: Local and Remote Endpoint, protocol selection properties, and specific protocol options. Applications configure security protocols during pre-establishment using the passive interfaces described in Section Section 3.1. Active control interfaces are exercised during connection establishment, i.e., from pre-establishment to established states. Applications can query connection metadata or state...
information, e.g., peer identity information, during and after connection establishment.

7. IANA Considerations

This document has on request to IANA.

8. Security Considerations

(editor’s note: this section will be added later. However, this document discusses the use of cryptographic context for transport connections and as such it has security relevant consideration within the whole document.)

9. Acknowledgments

This work is partially supported by the European Commission under Horizon 2020 grant agreement no. 688421 Measurement and Architecture for a Middleboxed Internet (MAMI), and by the Swiss State Secretariat for Education, Research, and Innovation under contract no. 15.0268. This support does not imply endorsement. Thanks to Brian Trammell for reviewing this draft.

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Abstract

This document identifies a need for improvement of the unicast Reverse Path Filtering techniques (uRPF) [BCP84] for source address validation (SAV) [BCP38]. The strict uRPF is inflexible about directionality, the loose uRPF is oblivious to directionality, and the current feasible-path uRPF attempts to strike a balance between the two [BCP84]. However, as shown in this draft, the existing feasible-path uRPF still has shortcomings. This document proposes an enhanced feasible-path uRPF technique, which aims to be more flexible (in a meaningful way) about directionality than the feasible-path uRPF. It can potentially alleviate ISPs’ concerns about the possibility of disrupting service for their customers, and encourage greater deployment of uRPF techniques.

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1. Introduction

This internet draft identifies a need for improvement of the unicast Reverse Path Filtering (uRPF) techniques [RFC2827] for source address validation (SAV) [RFC3704]. The strict uRPF is inflexible about directionality, the loose uRPF is oblivious to directionality, and the current feasible-path uRPF attempts to strike a balance between the two [RFC3704]. However, as shown in this draft, the existing feasible-path uRPF still has short comings. Even with the feasible-path uRPF, ISPs are often apprehensive that they may be dropping customers’ data packets with legitimate source addresses.

This document proposes an enhanced feasible-path uRPF technique, which aims to be more flexible (in a meaningful way) about directionality than the feasible-path uRPF. It is based on the
principle that if BGP updates for multiple prefixes with the same origin AS were received on different interfaces (at border routers), then incoming data packets with source addresses in any of those prefixes should be accepted on any of those interfaces (described in Section 3.1). For some challenging ISP-customer scenarios (see Section 3.3), we further propose (a) Forming a list of all unique prefixes in the collection of routes received on all customer interfaces; and (b) Including that list in the RPF list of each customer interface (described in Section 3.4). Implementation considerations are discussed in Section 3.5.

Note: Definition of Reverse Path Filtering (RPF) list: The list of permissible source address prefixes for incoming data packets on a given interface.

The proposed techniques are expected to add greater operational robustness and efficacy to uRPF, while minimizing ISPs’ concerns about accidental service disruption for their customers. It is expected that this will encourage more deployment of uRPF so as to realize its DDoS prevention benefits network wide.

1.1. Requirements Language

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in RFC 2119 [RFC2119].

2. Review of Existing Source Address Validation Techniques

There are various existing techniques for mitigation against DDoS attacks with spoofed addresses [RFC2827] [RFC3704]. There are also some techniques used for mitigating reflection attacks [RRL] [TA14-017A], which are used to amplify the impact in DDoS attacks. Employing a combination of these preventive techniques in enterprise and ISP border routers, DNS servers, broadband and wireless access networks, and data centers provides reasonably effective protection against DDoS attacks.

Source address validation (SAV) is performed in network edge devices such as border routers, Cable Modem Termination Systems (CMTS), Digital Subscriber Line Access Multiplexers (DSLAM), and Packet Data Network (PDN) gateways in mobile networks. Ingress Access Control List (ACL) and unicast Reverse Path Filtering (uRPF) are techniques employed for implementing SAV [RFC2827] [RFC3704] [ISOC].
2.1. SAV using Access Control List

Ingress/egress Access Control Lists (ACLs) are maintained which list acceptable (or alternatively, unacceptable) prefixes for the source addresses in the incoming Internet Protocol (IP) packets. Any packet with a source address that does not match the filter is dropped. The ACLs for the ingress/egress filters need to be maintained to keep them up to date. Updating the ACLs is an operator driven manual process, and hence operationally difficult or infeasible.

Typically, the egress ACLs in access aggregation devices (e.g. CMTS, DSLAM) permit source addresses only from the address spaces (prefixes) that are associated with the interface on which the customer network is connected. Ingress ACLs are typically deployed on border routers, and drop ingress packets when the source address is spoofed (i.e. belongs to obviously disallowed prefix blocks, RFC 1918 prefixes, or provider’s own prefixes).

2.2. SAV using Strict Unicast Reverse Path Filtering

In the strict unicast Reverse Path Filtering (uRPF) method, an ingress packet at border router is accepted only if the Forwarding Information Base (FIB) contains a prefix that encompasses the source address and forwarding information for that destination prefix points back to the interface over which the packet was received. In other words, the reverse path for routing to that source address (if it were used as a destination address) should use the same interface over which the packet was received. It is well known that this method has limitations when networks are multi-homed and there is asymmetric routing of packets. Asymmetric routing occurs (see Figure 1) when a customer AS announces one prefix (P1) to one transit provider (ISP-a) and a different prefix (P2) to another transit provider (ISP-b), but routes data packets with source addresses in the second prefix (P2) to the first transit provider (ISP-a) or vice versa.
Consider data packets received at AS2
(1) from AS1 with source address in P2, or
(2) from AS3 that originated from AS1
with source address in P1:
* Strict uRPF fails
* Feasible-path uRPF fails
* Loose uRPF works (but ineffective in IPv4)
* Enhanced Feasible-path uRPF works best

Figure 1: Scenario 1 for illustration of efficacy of uRPF schemes.

2.3. SAV using Feasible-Path Unicast Reverse Path Filtering

The feasible-path uRPF helps partially overcome the problem
identified with the strict uRPF in the multi-homing case. The
feasible-path uRPF is similar to the strict uRPF, but in addition to
inserting the best-path prefix, additional prefixes from alternative
announced routes are also included in the RPF table. This method
relies on announcements for the same prefixes (albeit some may be
prepended to effect lower preference) propagating to all routers
performing feasible-path uRPF checks. Therefore, in the multi-homing
scenario, if the customer AS announces routes for both prefixes (P1,
P2) to both transit providers (with suitable prepends if needed for
traffic engineering), then the feasible-path uRPF method works (see
Figure 2). It should be mentioned that the feasible-path uRPF works
in this scenario only if customer routes are preferred at AS2 and AS3
over a shorter non-customer route.
Consider data packets received at AS2 via AS3 that originated from AS1 and have source address in P1:

* Feasible-path uRPF works (if customer route to P1 is preferred at AS3 over shorter path)
* Feasible-path uRPF fails (if shorter path to P1 is preferred at AS3 over customer route)
* Loose uRPF works (but ineffective in IPv4)
* Enhanced Feasible-path uRPF works best

Figure 2: Scenario 2 for illustration of efficacy of uRPF schemes.

However, the feasible-path uRPF method has limitations as well. One form of limitation naturally occurs when the recommendation of propagating the same prefixes to all routers is not followed. Another form of limitation can be described as follows. In Scenario 2 (described above, illustrated in Figure 2), it is possible that the second transit provider (ISP-b or AS3) does not propagate the prepended route for prefix P1 to the first transit provider (ISP-a or AS2). This is because AS3’s decision policy permits giving priority to a shorter route to prefix P1 via a peer (AS2) over a longer route learned directly from the customer (AS1). In such a scenario, AS3 would not send any route announcement for prefix P1 to AS2. Then a data packet with source address in prefix P1 that originates from AS1 and traverses via AS3 to AS2 will get dropped at AS2.

2.4. SAV using Loose Unicast Reverse Path Filtering

In the loose unicast Reverse Path Filtering (uRPF) method, an ingress packet at the border router is accepted only if the FIB has one or more prefixes that encompass the source address. That is, a packet is dropped if no route exists in the FIB for the source address. Loose uRPF sacrifices directionality. This method is not effective for prevention of address spoofing since there is little unrouted address space in IPv4. It only drops packets if the spoofed address
is unreachable in the current FIB (e.g. RFC 1918, unallocated, allocated but currently not routed).


3.1. Description of the Method

Enhanced feasible-path uRPF adds greater operational robustness and efficacy to existing uRPF methods discussed in Section 2. The proposed technique is based on the principle that if BGP updates for multiple prefixes with the same origin AS were received on different interfaces (at border routers), then incoming data packets with source addresses in any of those prefixes should be accepted on any of those interfaces. It can be best explained with an example as follows:

Let us say, a border router of ISP-A has in its Adj-RIB-in the set of prefixes {Q1, Q2, Q3} each of which has AS-x as its origin and AS-x is in ISP-A’s customer cone. Further, the border router received a route for prefix Q1 over a customer facing interface, while it learned routes for prefixes Q2 and Q3 from a lateral peer and an upstream transit provider, respectively. All these routes passed route filtering and/or origin validation (i.e. the origin AS-x is deemed legitimate). That is, the route announcements are considered legitimate. In this example scenario, the enhanced feasible-path uRPF method allows source addresses to belong in {Q1, Q2, Q3} on any of the three specific interfaces in question (customer, peer, provider) on which the three routes were learned.

Thus, enhanced feasible-path uRPF defines feasible paths in a more generalized but precise way (as compared to feasible-path uRPF). In the above example, routes for prefixes Q2 and Q3 were not received on a customer facing interface at the border router, yet data packets with source addresses in Q2 or Q3 are accepted by the router if they come in on the same customer interface on which the route for prefix Q1 was received (based on these prefix routes having the same origin AS).

Looking back at Scenarios 1 and 2 (Figure 1 and Figure 2), the enhanced feasible-path uRPF provides comparable or better performance than the other uRPF methods. Scenario 3 (Figure 3) further illustrates the enhanced feasible-path uRPF method with a more concrete example. In this scenario, the focus is on operation of the feasible-path uRPF at ISP4 (AS4). ISP4 learns a route for prefix P1 via a customer-to-provider (C2P) interface from customer ISP2 (AS2). This route for P1 has origin AS1. ISP4 also learns a route for P2 via another C2P interface from customer ISP3 (AS3). Additionally, AS4 learns an alternate route for P2 via a peer-to-peer (p2p)
interface from ISP 5 (AS5). Both routes for P2 have the same origin AS (i.e. AS1) as does the route for P1. Using the proposed enhanced feasible-path uRPF scheme, given the commonality of the origin AS across the above-mentioned routes for P1 and P2, AS4 would permit source addresses belonging to either P1 or P2 in data packets received on any of the three interfaces (from AS2, AS3, and AS5).

Consider that data packets (sourced from AS1) may be received at AS4 with source address in P1 or P2 via any of the neighbors (AS2, AS3, AS5):
* Feasible-path uRPF fails
* Loose uRPF works (but not desirable)
* Enhanced Feasible-path uRPF works best

Figure 3: Scenario 3 for illustration of efficacy of uRPF schemes.

Based on the above, the proposed enhanced feasible-path uRPF method would reduce ISP concerns about possible service disruption affecting their customers and encourage greater adoption of uRPF.

3.2. Operational Recommendations

The following operational recommendations will make the operation of the proposed enhanced feasible-path uRPF robust:

For multi-homed stub AS:
A multi-homed stub AS SHOULD announce at least one of the prefixes it originates to each of its transit provider ASes.

For non-stub AS:

- A non-stub AS SHOULD also announce at least one of the prefixes it originates to each of its transit provider ASes.
- Additionally, from the routes it has learned from customers, a non-stub AS SHOULD announce at least one route per origin AS to each of its transit provider ASes.

(Note: It is worth noting that in the above recommendations if "at least one" is replaced with "all", then even traditional feasible-path uRPF will work as desired.)

3.3. A Challenging Scenario

It should be observed that in the absence of ASes adhering the above recommendations, the following example scenarios may be constructed which pose a challenge for the enhanced feasible-path uRPF (as well as for traditional feasible-path uRPF). In the scenario illustrated in Figure 4, since routes for neither P1 nor P2 are propagated on the AS2-AS4 interface, the enhanced feasible-path uRPF at AS4 will reject data packets received on that interface with source addresses in P1 or P2.
3.4. Overcoming the Above Challenge: Algorithm with Full Flexibility Across Customer Cone

Adding further flexibility to the enhanced feasible-path uRPF method can help address the potential limitation identified above using the scenario in Figure 4 (Section 3.3). In the following, "route" refers to a route currently existing in the Adj-RIB-in. Including the additional degree of flexibility, the modified algorithm can be described as follows:

- Let $I = \{I_1, I_2, ..., I_n\}$ represent the set of all directly-connected customer interfaces at customer-facing edge routers in a transit provider’s AS.
- Let $P = \{P_1, P_2, ..., P_m\}$ represent the set of all unique prefixes for which routes were received over the interfaces in Set $I$.
- Let $A = \{A_{S1}, A_{S2}, ..., A_{S_k}\}$ represent the set of all unique origin ASes seen in the routes that were received over the interfaces in Set $I$.
- Let $Q = \{Q_1, Q_2, ..., Q_j\}$ represent the set of all unique prefixes for which routes were received over peer or provider interfaces such that each of the routes has its origin AS belonging in Set $A$.

Figure 4: Illustration of a challenging scenario.
Then, \( \text{Set } Z = \text{Union}(P, Q) \) represents the RPF list for each customer-facing edge router in the AS in question. That is, over each interface in Set I, the edge router SHOULD permit only those ingress data packets that have SA in any of the prefixes in Set Z.

When this algorithmic flexibility is incorporated, then the type of limitation identified in Figure 4 (Section 3.3) goes away. This should significantly reduce the possibility of blocking legitimate customer-data packets in uRPF implementations.

3.5. Implementation Considerations

The existing RPF checks in edge routers take advantage of existing line card implementations to perform the RPF functions. For implementation of the proposed technique, the general necessary feature would be to extend the line cards to take arbitrary RPF lists that are not necessarily the same as the existing FIB contents. For example, in the proposed method, the RPF lists are constructed by applying a set of rules to all received BGP routes (not just those selected as best path and installed in FIB).

3.5.1. Impact on FIB Memory Size Requirement

The proposed technique requires that there should be FIB memory (i.e., TCAM) available to store the RPF lists in line cards. For an ISP’s AS, the RPF list size for each line card will roughly and conservatively equal the total number of prefixes in its customer cone (assuming the algorithm in Section 3.4 is used). The following table shows the measured customer cone sizes for various types of ISPs [sriram-ripe63]:

Sriram, et al. Expires May 3, 2018
<table>
<thead>
<tr>
<th>Type of ISP</th>
<th>Measured Customer Cone Size in # Prefixes (in turn this is an estimate for RPF list size on line card)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Very Large Global ISP</td>
<td>32392</td>
</tr>
<tr>
<td>Very Large Global ISP</td>
<td>29528</td>
</tr>
<tr>
<td>Large Global ISP</td>
<td>20038</td>
</tr>
<tr>
<td>Mid-size Global ISP</td>
<td>8661</td>
</tr>
<tr>
<td>Regional ISP (in Asia)</td>
<td>1101</td>
</tr>
</tbody>
</table>

Table 1: Customer cone sizes (# prefixes) for various types of ISPs.

For some super large global ISPs that are at the core of the Internet, the customer cone size (# prefixes) can be as high as a few hundred thousand [caida]. But uRPF is most effective when deployed at ASes at the edges of the Internet where the customer cone sizes are smaller as shown in Table 1.

A very large global ISP’s router line card is likely to have a FIB size large enough to accommodate 2 to 6 million routes [cisco1]. Similarly, the line cards in routers corresponding to a large global ISP, a mid-size global ISP, and a regional ISP are likely to have FIB sizes large enough to accommodate about 1 million, 0.5 million, and 100K routes, respectively [cisco2]. Comparing these FIB size numbers with the corresponding RPF list size numbers in Table 1, it can be surmised that the conservatively estimated RPF list size is only a small fraction of the anticipated FIB memory size under various ISP scenarios.

4. Security Considerations

This document offers a technique to improve the robustness features of uRPF and thus improve the security of the Internet as a whole. The proposed technique does not warrant any additional security considerations.

5. IANA Considerations

This document does not request new capabilities or attributes. It does not create any new IANA registries.
6. Acknowledgements

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Abstract

This document identifies a need for improvement of the unicast Reverse Path Filtering techniques (uRPF) [BCP84] for source address validation (SAV) [BCP38]. The strict uRPF is inflexible about directionality, the loose uRPF is oblivious to directionality, and the current feasible-path uRPF attempts to strike a balance between the two [BCP84]. However, as shown in this draft, the existing feasible-path uRPF still has short comings. This document describes an enhanced feasible-path uRPF technique, which aims to be more flexible (in a meaningful way) about directionality than the feasible-path uRPF. It can potentially alleviate ISPs’ concerns about the possibility of disrupting service for their customers, and encourage greater deployment of uRPF techniques.

Status of This Memo

This Internet-Draft is submitted in full conformance with the provisions of BCP 78 and BCP 79.

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1. Introduction

This internet draft identifies a need for improvement of the unicast Reverse Path Filtering (uRPF) techniques [RFC2827] for source address validation (SAV) [RFC3704]. The strict uRPF is inflexible about directionality, the loose uRPF is oblivious to directionality, and the current feasible-path uRPF attempts to strike a balance between the two [RFC3704]. However, as shown in this draft, the existing feasible-path uRPF still has short comings. Even with the feasible-path uRPF, ISPs are often apprehensive that they may be dropping customers’ data packets with legitimate source addresses.
This document describes an enhanced feasible-path uRPF technique, which aims to be more flexible (in a meaningful way) about directionality than the feasible-path uRPF. It is based on the principle that if BGP updates for multiple prefixes with the same origin AS were received on different interfaces (at border routers), then incoming data packets with source addresses in any of those prefixes should be accepted on any of those interfaces (presented in Section 3). For some challenging ISP-customer scenarios (see Section 3.3), this document also describes a more relaxed version of the enhanced feasible-path uRPF technique (presented in Section 3.4). Implementation considerations are discussed in Section 3.5.

Note: Definition of Reverse Path Filtering (RPF) list: The list of permissible source address prefixes for incoming data packets on a given interface.

Note: Throughout this document, the routes in consideration are assumed to have been vetted based on prefix filtering [RFC7454] and possibly (in the future) origin validation [RFC6811].

The enhanced feasible-path uRPF methods described here are expected to add greater operational robustness and efficacy to uRPF, while minimizing ISPs’ concerns about accidental service disruption for their customers. It is expected that this will encourage more deployment of uRPF to help realize its DDoS prevention benefits network wide.

1.1. Requirements Language

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in RFC 2119 [RFC2119].

2. Review of Existing Source Address Validation Techniques

There are various existing techniques for mitigation against DDoS attacks with spoofed addresses [RFC2827] [RFC3704]. There are also some techniques used for mitigating reflection attacks [RRL] [TA14-017A], which are used to amplify the impact in DDoS attacks. Employing a combination of these preventive techniques (as applicable) in enterprise and ISP border routers, broadband and wireless access network, data centers, and DNS servers provides reasonably effective protection against DDoS attacks.

Source address validation (SAV) is performed in network edge devices such as border routers, Cable Modem Termination Systems (CMTS), Digital Subscriber Line Access Multiplexers (DSLAM), and Packet Data Network (PDN) gateways in mobile networks. Ingress Access Control
List (ACL) and unicast Reverse Path Filtering (uRPF) are techniques employed for implementing SAV [RFC2827] [RFC3704] [ISOC].

2.1. SAV using Access Control List

Ingress/egress Access Control Lists (ACLs) are maintained which list acceptable (or alternatively, unacceptable) prefixes for the source addresses in the incoming Internet Protocol (IP) packets. Any packet with a source address that does not match the filter is dropped. The ACLs for the ingress/egress filters need to be maintained to keep them up to date. Updating the ACLs is an operator driven manual process, and hence operationally difficult or infeasible.

Typically, the egress ACLs in access aggregation devices (e.g. CMTS, DSLAM) permit source addresses only from the address spaces (prefixes) that are associated with the interface on which the customer network is connected. Ingress ACLs are typically deployed on border routers, and drop ingress packets when the source address is spoofed (i.e. belongs to obviously disallowed prefix blocks, RFC 1918 prefixes, or provider’s own prefixes).

2.2. SAV using Strict Unicast Reverse Path Filtering

In the strict unicast Reverse Path Filtering (uRPF) method, an ingress packet at border router is accepted only if the Forwarding Information Base (FIB) contains a prefix that encompasses the source address, and forwarding information for that prefix points back to the interface over which the packet was received. In other words, the reverse path for routing to the source address (if it were used as a destination address) should use the same interface over which the packet was received. It is well known that this method has limitations when networks are multi-homed and there is asymmetric routing of packets. Asymmetric routing occurs (see Figure 1) when a customer AS announces one prefix (P1) to one transit provider (ISP-a) and a different prefix (P2) to another transit provider (ISP-b), but routes data packets with source addresses in the second prefix (P2) to the first transit provider (ISP-a) or vice versa.
Consider data packets received at AS2
(1) from AS1 with source address in P2, or
(2) from AS3 that originated from AS1
with source address in P1:
* Strict uRPF fails
* Feasible-path uRPF fails
* Loose uRPF works (but ineffective in IPv4)
* Enhanced Feasible-path uRPF works best

Figure 1: Scenario 1 for illustration of efficacy of uRPF schemes.

2.3. SAV using Feasible-Path Unicast Reverse Path Filtering

The feasible-path uRPF helps partially overcome the problem
identified with the strict uRPF in the multi-homing case. The
feasible-path uRPF is similar to the strict uRPF, but in addition to
inserting the best-path prefix, additional prefixes from alternative
announced routes are also included in the RPF table. This method
relies on announcements for the same prefixes (albeit some may be
prepended to effect lower preference) propagating to all routers
performing feasible-path uRPF checks. Therefore, in the multi-homing
scenario, if the customer AS announces routes for both prefixes (P1,
P2) to both transit providers (with suitable prepends if needed for
traffic engineering), then the feasible-path uRPF method works (see
Figure 2). It should be mentioned that the feasible-path uRPF works
in this scenario only if customer routes are preferred at AS2 and AS3
over a shorter non-customer route.
Consider data packets received at AS2 via AS3 that originated from AS1 and have source address in P1:

- Feasible-path uRPF works (if customer route to P1 is preferred at AS3 over shorter path)
- Feasible-path uRPF fails (if shorter path to P1 is preferred at AS3 over customer route)
- Loose uRPF works (but ineffective in IPv4)
- Enhanced Feasible-path uRPF works best

Figure 2: Scenario 2 for illustration of efficacy of uRPF schemes.

However, the feasible-path uRPF method has limitations as well. One form of limitation naturally occurs when the recommendation of propagating the same prefixes to all routers is not followed. Another form of limitation can be described as follows. In Scenario 2 (described above, illustrated in Figure 2), it is possible that the second transit provider (ISP-b or AS3) does not propagate the prepended route for prefix P1 to the first transit provider (ISP-a or AS2). This is because AS3’s decision policy permits giving priority to a shorter route to prefix P1 via a peer (AS2) over a longer route learned directly from the customer (AS1). In such a scenario, AS3 would not send any route announcement for prefix P1 to AS2. Then a data packet with source address in prefix P1 that originates from AS1 and traverses via AS3 to AS2 will get dropped at AS2.

2.4. SAV using Loose Unicast Reverse Path Filtering

In the loose unicast Reverse Path Filtering (uRPF) method, an ingress packet at the border router is accepted only if the FIB has one or more prefixes that encompass the source address. That is, a packet is dropped if no route exists in the FIB for the source address. Loose uRPF sacrifices directionality. This method is not effective for prevention of address spoofing since there is little unrouted address space in IPv4. It only drops packets if the spoofed address
3. SAV using Enhanced Feasible-Path uRPF

3.1. Description of the Method

Enhanced feasible-path uRPF adds greater operational robustness and efficacy to existing uRPF methods discussed in Section 2. The technique is based on the principle that if BGP updates for multiple prefixes with the same origin AS were received on different interfaces (at border routers), then incoming data packets with source addresses in any of those prefixes should be accepted on any of those interfaces. It can be best explained with an example as follows:

Let us say, a border router of ISP-A has in its Adj-RIB-Ins [RFC4271]. the set of prefixes {Q1, Q2, Q3} each of which has AS-x as its origin and AS-x is in ISP-A’s customer cone. Further, the border router received a route for prefix Q1 over a customer facing interface, while it learned routes for prefixes Q2 and Q3 from a lateral peer and an upstream transit provider, respectively. In this example scenario, the enhanced feasible-path uRPF method requires Q1, Q2, and Q3 be included in the RPF list for the customer interface in consideration. Loose uRPF (see Section 2.4) is recommended to be applied to the peer and provider interfaces in consideration.

Thus, enhanced feasible-path uRPF defines feasible paths for customer interfaces in a more generalized but precise way (as compared to feasible-path uRPF).

Looking back at Scenarios 1 and 2 (Figure 1 and Figure 2), the enhanced feasible-path uRPF provides comparable or better performance than the other uRPF methods. Scenario 3 (Figure 3) further illustrates the enhanced feasible-path uRPF method with a more concrete example. In this scenario, the focus is on operation of the feasible-path uRPF at ISP4 (AS4). ISP4 learns a route for prefix P1 via a customer-to-provider (C2P) interface from customer ISP2 (AS2). This route for P1 has origin AS1. ISP4 also learns a route for P2 via another C2P interface from customer ISP3 (AS3). Additionally, AS4 learns a route for P3 via a peer-to-peer (p2p) interface from ISP5 (AS5). Routes for all three prefixes have the same origin AS (i.e. AS1). Using the enhanced feasible-path uRPF scheme, given the commonality of the origin AS across the routes for P1, P2 and P3, AS4 includes all of these prefixes to the RPF list for the customer interfaces (from AS2 and AS3).
Consider that data packets (sourced from AS1) may be received at AS4 with source address in P1, P2 or P3 via any of the neighbors (AS2, AS3, AS5):
* Feasible-path uRPF fails
* Loose uRPF works (but not desirable)
* Enhanced Feasible-path uRPF works best

Figure 3: Scenario 3 for illustration of efficacy of uRPF schemes.

3.1.1. Algorithm A: Enhanced Feasible-Path uRPF

The underlying algorithm in the solution method described above can be specified as follows (to be implemented in a transit AS):

1. Create the list of unique origin ASes considering only the routes in the Adj-RIB-Ins of customer interfaces. Call it Set A = {AS1, AS2, ..., ASn}.

2. Considering all routes in Adj-RIB-Ins for all interfaces (customer, lateral peer, and provider), form the set of unique prefixes that have a common origin AS1. Call it Set X1.

3. Include set X1 in Reverse Path Filter (RPF) list on all customer interfaces on which one or more of the prefixes in set X1 were received.
4. Repeat Steps 2 and 3 for each of the remaining ASes in Set A (i.e., for AS\textsubscript{i}, where \(i = 2, \ldots, n\)).

3.2. Operational Recommendations

The following operational recommendations will make the operation of the enhanced feasible-path uRPF robust:

For multi-homed stub AS:

- A multi-homed stub AS SHOULD announce at least one of the prefixes it originates to each of its transit provider ASes.

For non-stub AS:

- A non-stub AS SHOULD also announce at least one of the prefixes it originates to each of its transit provider ASes.

- Additionally, from the routes it has learned from customers, a non-stub AS SHOULD announce at least one route per origin AS to each of its transit provider ASes.

(Note: It is worth noting that in the above recommendations if "at least one" is replaced with "all", then even traditional feasible-path uRPF will work as effectively.)

3.3. A Challenging Scenario

It should be observed that in the absence of ASes adhering the above recommendations, the following example scenario may be constructed which poses a challenge for the enhanced feasible-path uRPF (as well as for traditional feasible-path uRPF). In the scenario illustrated in Figure 4, since routes for neither P1 nor P2 are propagated on the AS2-AS4 interface, the enhanced feasible-path uRPF at AS4 will reject data packets received on that interface with source addresses in P1 or P2.
3.4. Algorithm B: Enhanced Feasible-Path uRPF with Additional Flexibility Across Customer Cone

Adding further flexibility to the enhanced feasible-path uRPF method can help address the potential limitation identified above using the scenario in Figure 4 (Section 3.3). In the following, "route" refers to a route currently existing in the Adj-RIB-in. Including the additional degree of flexibility, the modified algorithm (implemented in a transit AS) can be described as follows (we call this Algorithm B):

1. Create the set of all directly-connected customer interfaces. Call it Set I = {I1, I2, ..., Ik}.
2. Create the set of all unique prefixes for which routes exist in Adj-RIB-Ins for the interfaces in Set I. Call it Set P = {P1, P2, ..., Pm}.
3. Create the set of all unique origin ASes seen in the routes that exist in Adj-RIB-Ins for the interfaces in Set I. Call it Set A = {AS1, AS2, ..., ASn}.
4. Create the set of all unique prefixes for which routes exist in Adj-RIB-Ins of all lateral peer and provider interfaces such that...
each of the routes has its origin AS belonging in Set A. Call it 
Set Q = \{Q1, Q2, \ldots, Qj\}.

5. Then, Set Z = \text{Union}(P, Q) represents the RPF list for every 
customer interface in Set I.

6. Apply loose uRPF method for SAV on all peer and provider 
interfaces.

When Algorithm B (which is more flexible than Algorithm A) is 
employed, the type of limitation identified in Figure 4 (Section 3.3) 
goes away.

3.5. Implementation Considerations

The existing RPF checks in edge routers take advantage of existing 
line card implementations to perform the RPF functions. For 
implementation of the enhanced feasible-path uRPF, the general 
necessary feature would be to extend the line cards to take arbitrary 
RPF lists that are not necessarily the same as the existing FIB 
contents. In the algorithms (Section 3.1.1 and Section 3.4) 
described here, the RPF lists are constructed by applying a set of 
rules to all received BGP routes (not just those selected as best 
path and installed in FIB).

3.5.1. Impact on FIB Memory Size Requirement

The techniques described here require that there should be FIB memory 
(i.e., TCAM) available to store the RPF lists in line cards. For an 
ISP’s AS, the RPF list size for each line card will roughly and 
conservatively equal the total number of prefixes in its customer 
cone (assuming the algorithm in Section 3.4 is used). (Note: Most 
ISP customer cone scenarios would not require the algorithm in 
Section 3.4, but instead be served best by the algorithm in 
Section 3.1.1, which requires much less FIB memory.) The following 
table shows the measured customer cone sizes for various types of 
ISPs [sriram-ripe63]:
### Table 1: Customer cone sizes (# prefixes) for various types of ISPs.

<table>
<thead>
<tr>
<th>Type of ISP</th>
<th>Measured Customer Cone Size in # Prefixes (in turn this is an estimate for RPF list size on line card)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Very Large Global ISP</td>
<td>32392</td>
</tr>
<tr>
<td>Very Large Global ISP</td>
<td>29528</td>
</tr>
<tr>
<td>Large Global ISP</td>
<td>20038</td>
</tr>
<tr>
<td>Mid-size Global ISP</td>
<td>8661</td>
</tr>
<tr>
<td>Regional ISP (in Asia)</td>
<td>1101</td>
</tr>
</tbody>
</table>

For some super large global ISPs that are at the core of the Internet, the customer cone size (# prefixes) can be as high as a few hundred thousand [CAIDA]. But uRPF is most effective when deployed at ASes at the edges of the Internet where the customer cone sizes are smaller as shown in Table 1.

A very large global ISP’s router line card is likely to have a FIB size large enough to accommodate 2 to 6 million routes [cisco1]. Similarly, the line cards in routers corresponding to a large global ISP, a mid-size global ISP, and a regional ISP are likely to have FIB sizes large enough to accommodate about 1 million, 0.5 million, and 100K routes, respectively [cisco2]. Comparing these FIB size numbers with the corresponding RPF list size numbers in Table 1, it can be surmised that the conservatively estimated RPF list size is only a small fraction of the anticipated FIB memory size under relevant ISP scenarios.

#### 3.6. Summary of Recommendations

Depending on the scenario, an ISP or enterprise AS operator should follow one of the following recommendations concerning uRPF/SAV:

1. For directly connected networks, i.e., subnets directly connected to the AS and not multi-homed, the AS in consideration SHOULD perform ACL-based SAV.

2. For a directly connected single-homed stub AS (customer), the AS in consideration SHOULD perform SAV based on the strict uRPF method.
3. For all other scenarios:

* If the scenario does not involve complexity such as NO_EXPORT of routes (see Section 3.3, Figure 4), then the enhanced feasible-path uRPF method in Algorithm A (see Section 3.1.1) SHOULD be applied.

* Else, if the scenario involves the aforementioned complexity, then the enhanced feasible-path uRPF method in Algorithm B (see Section 3.4) SHOULD be applied.

4. Security Considerations

The security considerations in BCP 38 [RFC2827] and BCP 84 [RFC3704] apply for this document as well. In addition, AS operator should apply the uRPF method that performs best (i.e., with zero or insignificant possibility of dropping legitimate data packets) for the type of peer (customer, provider, etc.) and the nature of customer cone scenario that apply (see Section 3.1.1 and Section 3.4).

5. IANA Considerations

This document does not request new capabilities or attributes. It does not create any new IANA registries.

6. Acknowledgements

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7. Informative References


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