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A TCP transport for the Internet Key Exchange
draft-nir-ipsecme-ike-tcp-01

Abstract

This document describes using TCP for IKE messages. This facilitates the transport of large messages over paths where fragments are either dropped, or packet loss makes them unreliable.

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1. Introduction

The Internet Key Exchange (IKE) specified in [RFC2407] and [RFC2408], and IKEv2 as specified in [RFC5996] uses UDP to transport the exchange messages. Some of those messages may be fairly large. Specifically, the 5th and 6th messages of IKEv1 Main Mode, the first and second messages of IKEv1 Aggressive Mode, and the messages of IKEv2 IKE_AUTH exchange can become quite large, as they may contain a chain of certificates, a signature payload (called "Auth" in IKEv2), CRLs, and in the case of IKEv2, some configuration information that is carried in the CFG payload.

When such UDP packets exceed the path MTU, they get fragmented. This increases the probability of packets getting dropped, but the retransmission mechanisms in IKE (as described in section 2.1 of RFC 5996) takes care of that. More recently we have seen a number of service providers dropping fragmented packets. Firewalls and NAT devices need to keep state for each packet where some but not all of the fragments have been received. This creates a burden in terms of memory, especially for high capacity devices such as Carrier-Grade NAT (CGN) or high capacity firewalls.

The BEHAVE working group has an Internet Draft describing required behavior of CGNs ([I-D.ietf-behave-lsn-requirements]). It requires CGNs to comply with [RFC4787], which in section 11 requires NAT devices to support fragments. However, some people deploying IKE have found that some ISPs have begun to drop fragments in preparation for deploying CGNs. While we all hope for a future where all devices comply with the emerging standards, or even a future where CGNs are not required, we have to make IKE work today.

The solution described in this document is to transport the IKE messages over a TCP ([RFC0793]) rather than over UDP. IKE packets (both versions) describe their own length, so they are well-suited for transport over a stream-based connection such as TCP. The Initiator opens a TCP connection to the Responder's port 500, sends the requests and receives the responses, and then closes the connection. TCP can handle arbitrary-length messages, works well with any sized data, and is well supported by all ISP infrastructure.

1.1. Non-Goals of this Specification

Firewall traversal is not a goal of this specification. If a firewall has a policy to block IKE and/or IPsec, hiding the IKE exchange in TCP is not expected to help. Some implementations hide both IKE and IPsec in a TCP connection, usually pretending to be HTTPS by using port 443. This has a significant impact on bandwidth and gateway capacity, and even this is defeated by better firewalls.

SSL VPNs tunnel IP packets over TLS, but the latest firewalls are also TLS proxies, and are able to defeat this as well.

This document is not part of that arms race. It is only meant to allow IKE to work when faced with broken infrastructure that drops large IP packets.

1.2. Conventions Used in This Document

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. The Protocol

2.1. Initiator

An Initiator MAY try IKE using TCP for any request. It opens a TCP connection from an arbitrary port to port 500 of the Responder. When the three-way handshake completes, the Initiator MUST send the request. If the Initiator knows that this request is the last request needed at this time, it SHOULD half-close the TCP connection, although it MAY wait until the last response is received. When all responses have been received, the Initiator MUST close the connection. If the peer has closed the connection before all requests have been transmitted or responded to, the Initiator SHOULD either open a new TCP connection or transmit them over UDP again.

It MUST accept responses sent over IKE within the same connection, but MUST also accept responses over other transports, if the request had been sent over them as well.

2.2. Responder

A Responder MAY accept TCP connections to port 500, and if it does, it MUST accept IKE requests over this connection. Responses to requests received over this connection MUST also go over this connection. If the connection has closed before the Responder had had a chance to respond, it MUST NOT respond over UDP, but MUST instead wait for a retransmission over UDP or over another TCP connection.

The responder MUST accept different requests on different transports. Specifically, the Responder MUST NOT rely on subsequent requests coming over the same transport. For example, it is entirely acceptable to have the first two requests on IKE Main Mode come over UDP port 500, while the last request comes over TCP, and the

following Quick Mode request might come over UDP port 4500 (because NAT has been detected).

A responder that receives an IKEv2 Initial request over any other transport MUST send an IKE_TCP_SUPPORTED notification (Section 2.5) in the Initial response. the responder MAY send this notification even if the Initial request was received over TCP.

If the responder has some requests of its own to send, it MUST NOT use a connection that has been opened by a peer. Instead, it MUST either use UDP or else open a new TCP connection to the original Initiator's TCP port 500.

The normal flow of things is that the Initiator opens a connection and closes its side first. The responder closes after sending the last response where the initiator has already half-closed the connection. If, however, a significant amount of time has passed, and neither new requests arrive nor the connection is closed by the initiator, the Responder MAY close or even reset the connection.

This specification makes no recommendation as to how long such a timeout should be, but a few seconds should be enough.

2.3. Transmitter

The transmitter, whether an initiator transmitting a request or a responder transmitting a response MUST NOT retransmit over the same connection. TCP takes care of that. It SHOULD send the IKE header and the IKE payloads with a single command or in rapid succession, because the receiver might block on reading from the socket.

2.4. Receiver

The IKE header is copied from RFC 5996 below for reference:

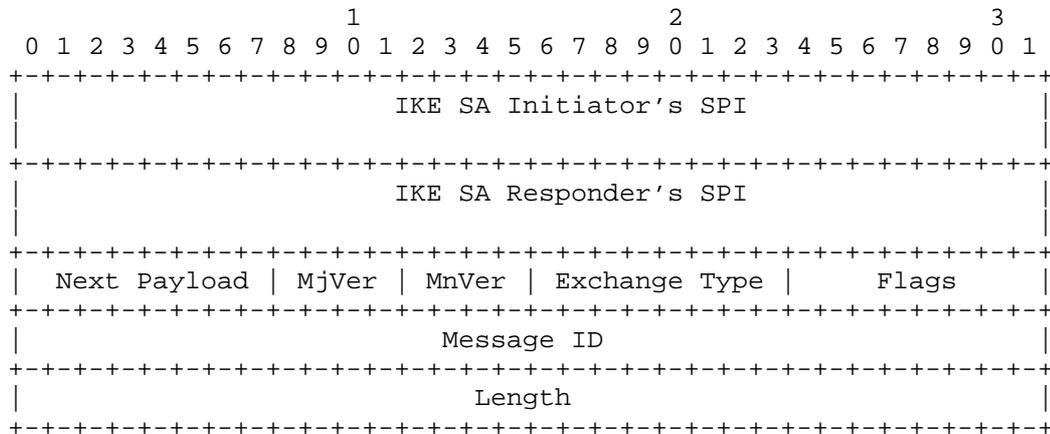


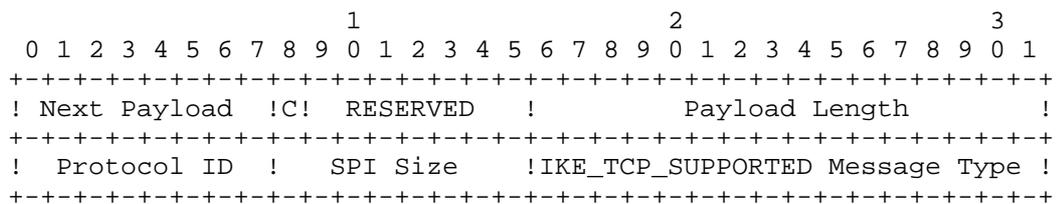
Figure 1: IKE Header Format

The receiver MUST first read in the 28 bytes that make up the IKE header. The Responder then subtracts 28 from the length field, and reads the resulting number of bytes. The combined message, comprised on 28 header bytes and whatever number of payload bytes is processed the same way as regular UDP messages. That includes retransmission detection, with one slight difference: if a retransmitted request is detected, the response is retransmitted as well, but using the current TCP connection rather than whatever other transport had been used for the original transmission of the request.

2.5. IKE_TCP_SUPPORTED Notification

This notification is sent by a responder over non-TCP transports to inform the initiator that this specification is supported.

The Notify payload is formatted as follows:



- o Protocol ID (1 octet) MUST be 0.
- o SPI Size (1 octet) MUST be zero, in conformance with section 3.10 of [RFC5996].

- o IKE_TCP_SUPPORTED Notify Message Type (2 octets) - MUST be xxxxx, the value assigned for IKE_TCP_SUPPORTED. TBA by IANA.

3. Operational Considerations

Most IKE messages are relatively short. Quick Mode in IKEv1, and all but the IKE_AUTH exchange in IKEv2 are comprised of short messages that fit in a single packet on most networks. It is only the IKE_AUTH exchange in IKEv2, and the two last messages of Main Mode that are long. UDP has advantages in lower latency and lower resource consumption, so it makes sense to use UDP whenever TCP is not required.

The requirements in Section 2.2 mean that different requests may be sent over different transports. So the initiator can choose the transport on a per-request basis. So one obvious policy would be to do everything over UDP except the specific requests that tend to become too big. This way the first messages use UDP, and the Initiator can set up the TCP connection at the same time, eliminating the latency penalty of using TCP. This may not always be the most efficient policy, though. It means that the first messages sent over TCP are relatively large ones, and TCP slow start may cause an extra roundtrip, because the message may exceed the transmission window. An initiator using this policy MUST NOT go to TCP if the responder has not indicated support by sending the IKE_TCP_SUPPORTED notification (Section 2.5) in the Initial response.

An alternative method, that is probably easier for the Initiator to implement, is to do an entire "mission" using the same transport. So if TCP is needed and an IKE SA has not yet been created, the Initiator will open a TCP connection, and perform all 2-4 requests needed to set up a child SA over the same connection.

Yet another policy would be to begin by using UDP, and at the same time set up the TCP connection. If at any point the TCP handshake completes, the next requests go over that connection. This method can be used to auto-discover support of TCP on the responder. This is easier for the user than configuring which peers support TCP, but has the potential of wasting resources, as TCP connections may finish the three-way handshake just when IKE over UDP has finished. The requirements from the responder ensure that all these policies will work.

3.1. Liveness Check

The TCP connections described in this document are short-lived. We do not expect them to stay for the lifetime of the SA, but to get

torn down by either side within seconds of the SA being set up. Because of this, they are not well-suited for the transport of short requests such as those for liveness check.

Although liveness checks MAY be sent over TCP, this is not recommended.

4. Security Considerations

Most of the security considerations for IKE over TCP are the same as those for UDP as in RFC 5996.

For the Responder, listening to TCP port 500 involves all the risks of maintaining any TCP server. Precautions against DoS attacks, such as SYN cookies are RECOMMENDED.

5. IANA Considerations

IANA is requested to assign a notify message type from the status types range (16418-40959) of the "IKEv2 Notify Message Types" registry with name "IKE_TCP_SUPPORTED"

No IANA action is required for the TCP port, as TCP port 500 is already allocated to "ISAKMP".

6. References

6.1. Normative References

- [RFC2119] Bradner, S., "Key words for use in RFCs to Indicate Requirement Levels", BCP 14, RFC 2119, March 1997.
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- [RFC5996] Kaufman, C., Hoffman, P., Nir, Y., and P. Eronen, "Internet Key Exchange Protocol Version 2 (IKEv2)", RFC 5996, September 2010.

6.2. Informative References

- [I-D.ietf-behave-lsn-requirements]
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- [RFC0793] Postel, J., "Transmission Control Protocol", STD 7,
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- [RFC4787] Audet, F. and C. Jennings, "Network Address Translation
(NAT) Behavioral Requirements for Unicast UDP", BCP 127,
RFC 4787, January 2007.

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