

Network Working Group	Q. Wu
Internet-Draft	F. Xia
Intended status: Standards Track	R. Even
Expires: November 07, 2011	Huawei
	May 06, 2011

RTCP Extension for Third-party Loss Report
draft-ietf-avtcore-feedback-supression-rtp-02

Abstract

In a large RTP session using the RTCP feedback mechanism defined in RFC 4585, a media source or middlebox may experience transient overload if some event causes a large number of receivers to send feedback at once. This feedback implosion can be mitigated if the device suffering from overload can send a third party loss report message to the receivers to inhibit further feedback. This memo defines RTCP Extension for third party loss report, to suppress NACK and FIR feedback requests. It also defines associated SDP signaling.

Status of this Memo

This Internet-Draft is submitted in full conformance with the provisions of BCP 78 and BCP 79.

Internet-Drafts are working documents of the Internet Engineering Task Force (IETF). Note that other groups may also distribute working documents as Internet-Drafts. The list of current Internet- Drafts is at <http://datatracker.ietf.org/drafts/current/>.

Internet-Drafts are draft documents valid for a maximum of six months and may be updated, replaced, or obsoleted by other documents at any time. It is inappropriate to use Internet-Drafts as reference material or to cite them other than as "work in progress."

This Internet-Draft will expire on November 07, 2011.

Copyright Notice

Copyright (c) 2011 IETF Trust and the persons identified as the document authors. All rights reserved.

This document is subject to BCP 78 and the IETF Trust's Legal Provisions Relating to IETF Documents (<http://trustee.ietf.org/license-info>) in effect on the date of publication of this document. Please review these documents carefully, as they describe your rights and restrictions with respect to this document. Code Components extracted from this document must include Simplified BSD License text as described in Section 4.e of the Trust Legal Provisions and are provided without warranty as described in the Simplified BSD License.

This document may contain material from IETF Documents or IETF Contributions published or made publicly available before November 10, 2008. The person(s) controlling the copyright in some of this material

may not have granted the IETF Trust the right to allow modifications of such material outside the IETF Standards Process. Without obtaining an adequate license from the person(s) controlling the copyright in such materials, this document may not be modified outside the IETF Standards Process, and derivative works of it may not be created outside the IETF Standards Process, except to format it for publication as an RFC or to translate it into languages other than English.

[Table of Contents](#)

- *1. [Introduction](#)
- *2. [Terminology](#)
- *3. [Protocol Overview](#)
- *4. [Format of RTCP Feedback Messages](#)
 - *4.1. [Transport Layer Feedback: Third-party Loss Report](#)
 - *4.2. [Payload Specific Feedback: Third-party Loss Report](#)
- *5. [SDP Signaling](#)
- *6. [Example Use Cases](#)
 - *6.1. [Source Specific Multicast \(SSM\) use case](#)
 - *6.1.1. [Simple Feedback Model](#)
 - *6.1.2. [Distribution Source Feedback Summary Model](#)
 - *6.2. [Unicast based Rapid Acquisition of Multicast Stream \(RAMS\) use case](#)
 - *6.3. [RTP transport translator use case](#)
 - *6.4. [Multipoint Control Unit \(MCU\) use case](#)
- *7. [Security Considerations](#)
- *8. [IANA Consideration](#)
- *9. [Acknowledgement](#)
- *10. [References](#)
 - *10.1. [Normative References](#)
 - *10.2. [Informative References](#)

*Appendix A. [Appendix A. Change Log](#)

*Appendix A.1. [draft-ietf-avtcore-feedback-suppression-rtp-01](#)

*Appendix A.2. [draft-ietf-avtcore-feedback-suppression-rtp-02](#)

*[Authors' Addresses](#)

1. Introduction

RTCP feedback messages [\[RFC4585\]](#) allow the receivers in an RTP session to report events and ask for action from the media source (or a delegated feedback target defined in SSM [\[RFC5760\]](#)). There are cases where multiple receivers may initiate the same, or an equivalent message towards the same media source. When the receiver count is large, this behavior may cause transient overload of the media source, the network or both. This is known as a "feedback storm" or a "NACK storm". One common cause of such a feedback storm is receivers utilizing RTP retransmission [\[RFC4588\]](#) as a packet loss recovery technique based, sending feedback using RTCP NACK messages [\[RFC4585\]](#) without proper dithering of the retransmission requests.

Another use case involves video Fast Update requests. A storm of these feedback messages can occur in conversational multimedia scenarios like Topo-Video-switch-MCU [\[RFC5117\]](#). In this scenario, packet loss may happen on an upstream link of an intermediate network element such as a Multipoint Control Unit(MCU). Poorly designed receivers that blindly issue fast update requests (i.e., Full Intra Request (FIR) described in [\[RFC5104\]](#)), can cause an implosion of FIR requests from receivers to the same media source.

RTCP feedback storms may cause short term overload, and in extreme cases to pose a possible risk of increasing network congestion on the control channel (e.g. RTCP feedback), the data channel, or both. It is therefore desirable to provide a way of suppressing unneeded feedback. One approach to this, suggested in [\[DVB-IPTV\]](#), involves sending a NACK message to the other clients (or receiver) in the same group as the sender of NACK. However sending multicast NACK to the group can not prevent large amount of unicast NACK addressed to the same media source or middlebox, for example when the NACK is used as a retransmission request [\[RFC4588\]](#). Also NACK is defined as a receiver report sent from a receiver observing a packet loss, therefore it only inform others that sender of NACK detected loss while the case the sender of the feedback has received reports that the indicated packets were lost is not covered. This document specifies a new message for this function. It further is more precise in the intended uses and less likely to be confusing to receivers. It tells receivers explicitly that feedback for a particular packet or frame loss is not needed for a period of time and can provide an early indication before the receiver reacts to the loss and invokes its packet loss repair machinery.

2. Terminology

The keywords "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in [\[RFC2119\]](#).

3. Protocol Overview

This document extends the RTCP feedback messages defined in the Audio-Visual Profile with Feedback (AVPF) and define the Third Party Loss Report message. The Third Party Loss Report message informs the receiver in the downstream path of the middlebox that the sender of the Third Party Loss Report has received reports that the indicated packets were lost and asks a receiver to not send feedback messages for particular packets (indicated by their RTP sequence numbers) independent of whether the receiver detected the packet loss or detected a need for a decoder refresh point.

In order to observe packet loss before the receivers perceive it, one or more intermediate nodes may be placed between the media source and the receivers. These intermediates are variously referred to as Distribution servers, MCUs, RTP translator, or RTP mixers, depending on the precise use case. These intermediaries monitor for packet loss upstream of themselves by checking RTP sequence numbers, just as receivers do. These intermediate nodes need to take into account such factors as the tolerable application delay, packet loss recovery techniques, the network dynamics, and the media type. Loss-repair methods such as retransmission and Forward Error Correction may be used to recover the missing packet. Upon observing (or suspecting) an upstream loss, the intermediary Should send NACK both downstream towards the receivers and upstream towards the media source, to indicate that it has noticed the loss, and to suppress feedback from other downstream receivers. Upon downstream loss is reported to the intermediary, the intermediary SHOULD send the Third Party Loss report to the other downstream receivers which are not aware of the loss reports, to inform those receivers of the loss and suppress their feedback. Therefore the intermediate node can be reasonably certain that it will help the situation by sending a Third Party Loss Report message and NACK message to all the relevant receivers, thereby indicating to the receivers that they should not transmit feedback messages for a period of time.

Alternatively, the media source may directly monitor the amount of feedback requests it receives from downstream, and send the Third Party Loss Report messages to the downstream receivers.

When a receiver gets such a Third Party Loss Report message, it should refrain from sending a feedback request (e.g., NACK or FIR) for the missing packets reported in the message for a period of time. A receiver may still have sent a Feedback message according to the AVPF scheduling algorithm of [\[RFC4585\]](#) before receiving a Third Party Loss Report message, but further feedback messages for those sequence

numbers will be suppressed by this technique for a period of time. Nodes that do not understand the Third Party Loss Report message will ignore it, and might therefore still send feedback according to the AVPF scheduling algorithm of [\[RFC4585\]](#). The media source or intermediate nodes cannot assume that the use of a Third Party Loss Report message actually reduces the amount of feedback it receives. RTCP Third Party Loss Report follows the similar format of message type as RTCP NACK. But unlike RTCP NACK, the third party loss report is defined as an indication that the sender of the feedback has received reports that the indicated packets were lost and conveys the packet receipt/loss events at the sequence number level from the middlebox to the receivers in the downstream path of middlebox while NACK [\[RFC4585\]](#) just indicates that the sender of the NACK observed that these packets were lost. The Third Party Loss Report message is generated by RTP middlebox that has not seen the actual packet loss and sent to the corresponding receivers. Intermediaries downstream of an intermediary receiving the upstream report obviously SHOULD NOT initiate their own additional Third Party Loss Report messages for the same packet sequence numbers. They may either simply forward the Third Party Loss Report message received from upstream, or send its own Third Party Loss Report message that reflects the loss they have been told. The Third Party Loss Report does not have the retransmission request [\[RFC4588\]](#) semantics.

Since Third Party Loss Report interacts strongly with repair timing, it has to work together with feedback to not adversely impact the repair of lost source packets. One example is the middle box gets the retransmitted packet by sending a NACK upstream and sent it downstream. This retransmitted packet was lost on the downstream link. In order to deal with this, the downstream receiver can start a timeout in which it expected to get a retransmission packet. When this timeout expires and there is no retransmitted packet or a new third party loss report message, it can take its normal behavior as if there is no current retransmission suppression. In some cases where the loss was detected and repair initiated much closer to the source, the delay for the receiver to recover from packet loss can be reduced through the combination of intermediary feedback to the source and Third Party Loss Report downstream. In all (properly operating) cases, the risk of increasing network congestion is decreased.

[4. Format of RTCP Feedback Messages](#)

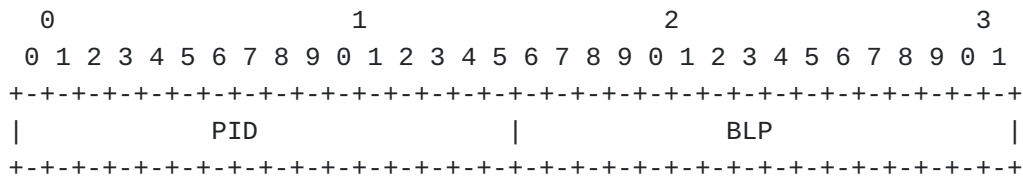
This document registers two new RTCP Feedback messages for Third Party Loss Report. Applications that are employing one or more loss-repair methods MAY use Third Party Loss Report together with their existing loss-repair methods either for every packet they expect to receive, or for an application-specific subset of the RTP packets in a session. In other words, receivers MAY ignore Third Party Loss Report messages, but SHOULD react to them unless they have good reason to still send feedback messages despite having been requested to suppress them.

4.1.1. Transport Layer Feedback: Third-party Loss Report

This Third Party Loss Report message is an extension to the RTCP Transport Layer Feedback Report and identified by RTCP packet type value PT=RTPFB and FMT=TBD.

The FCI field MUST contain one or more entries of transport layer third party loss Early Indication (TLLEI). Each entry applies to a different media source, identified by its SSRC.

The Feedback Control Information (FCI) for TLLEI uses the similar format of message Types defined in the section 6.2.1 of [\[RFC4585\]](#). The format is shown in [Figure 1](#).



Packet ID (PID): 16 bits

The PID field is used to specify a lost packet. The PID field refers to the RTP sequence number of the lost packet.

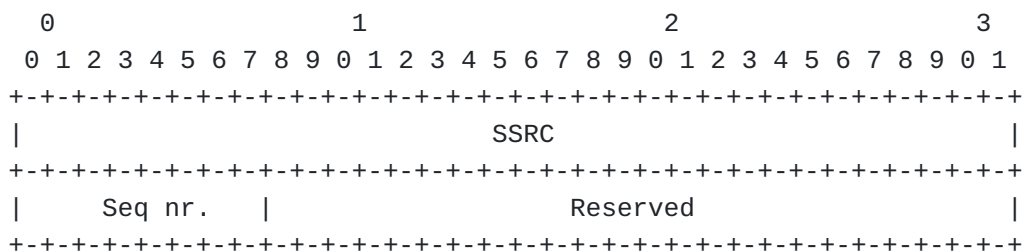
bitmask of proceeding lost packets (BLP): 16 bits

The BLP allows for reporting losses of any of the 16 RTP packets immediately following the RTP packet indicated by the PID. The BLP's definition is identical to that given in [RFC4585](#).

4.2. Payload Specific Feedback: Third-party Loss Report

This message is an extension to the RTP Payload Specific Feedback report and identified by RTP packet type value PT=PSFB and FMT=TBD. The FCI field MUST contain a Payload Specific Third Party Loss Early Indication (PSLEI) entry. Each entry applies to a different media source, identified by its SSRC.

The Feedback Control Information (FCI) for PSLEI uses the similar format of message Types defined in the section 4.3.1.1 of [\[RFC5104\]](#). The format is shown in [Figure 2](#).



SSRC (32 bits):

The SSRC value of the media source that is requested to send a decoder refresh point.

Seq nr:8bits Command sequence number. The sequence number space is unique for each pairing of the SSRC of command source and the SSRC of the command target. The sequence number SHALL be increased by 1 modulo 256 for each new request.

Reserved: 24 bits

All bits SHALL be set to 0 by the media source and SHALL be ignored on reception.

5. SDP Signaling

A new feedback value "tplr" needs to be defined for the Third Party Loss Report message to be used with Session Description Protocol (SDP) [\[RFC4566\]](#) using the Augmented Backus-Naur Form (ABNF) [\[RFC4585\]](#). The "tplr" feedback value SHOULD be used with parameters that indicate the third party loss supported. In this document, we define two such parameter, namely:

In the ABNF for rtcp-fb-val defined in [\[RFC4585\]](#), there is a placeholder called rtcp-fb-id to define new feedback types. "tplr" is defined as a new feedback type in this document, and the ABNF for the parameters for tplr is defined here (please refer to section 4.2 of [\[RFC4585\]](#) for complete ABNF syntax).

*"tllei" denotes support of transport layer third party loss early indication (fsei).

*"pslei" denotes support of payload specific third party loss early indication.

```
rtcp-fb-val          =/ "tplr" rtcp-fb-tplr-param
rtcp-fb-tplr-param   = SP "tllei";transport layer third party loss early indication
                      / SP "pslei";payload specific third party loss early indication
                      / SP token [SP byte-string]
                      ; for future commands/indications
byte-string = <as defined in section 4.2 of [RFC4585] >
```

Refer to Section 4.2 of [\[RFC4585\]](#) for a detailed description and the full syntax of the "rtcp-fb" attribute.

6. Example Use Cases

The operation of feedback suppression is similar for all types of RTP sessions and topologies [\[RFC5117\]](#), however the exact messages used and the scenarios in which suppression is employed differ for various use cases. The following sections outline the intended use cases of using Third Party Loss Report for feedback suppression and give an overview of the particular mechanisms.

6.1. Source Specific Multicast (SSM) use case

In SSM RTP sessions as described in [\[RFC5760\]](#), one or more Media Sources send RTP packets to a Distribution Source. The Distribution Source relays the RTP packets to the receivers using a source-specific multicast group.

In order to avoid the forms of Feedback implosion described in section 1, the distribution source should be told that the indicated packets were lost. How the distribution source know the indicated packets were lost is beyond of scope of this document. When one downstream receiver reports loss, the distribution source creates a Third Party Loss Report and sent it to all the RTP receivers, over the multicast channel. Another possibility is when there may be multiple distribution sources placed between the media source and the receivers, each distribution source may send its own Third Party Loss report to downstream receivers respectively when downstream loss is reported to each distribution source. And also the upstream distribution source may inform downstream distribution sources in the path of the detected packet loss using the Third Party Loss Report messages. In response, if the upstream Third Party Loss Report reports the different event, the downstream distribution sources forward Third Party Loss Report received from upstream to all the RTP receivers, over the multicast channel. If the same event is reported both from upstream distribution source and from downstream receiver, the downstream distribution source may suppress creating and sending its own report to the relevant RTP receivers. This Third Party Loss Report message tells the receivers that the sender of the third party loss report has received reports that the indicated packets were lost. The distribution source then can (optionally) ask for the lost packets from the media source or itself on behalf of all the RTP receivers. The lost packets will either be forthcoming from distribution source, or it irretrievably lost such that there is nothing to be gained by the receiver sending a NACK to the media source.

The distribution source must be able to communicate with all group members in order for either mechanism to be effective at suppressing feedback.

As outlined in the [\[RFC5760\]](#), there are two Unicast Feedback models that may be used for reporting, - the Simple Feedback model and the Distribution Source Feedback Summary Model. The RTCP Feedback extension for Third Party Loss Report specified in the [Section 4](#) of this document will work in both Feedback models. Details of operation in each are specified below.

[6.1.1.](#) Simple Feedback Model

In the simple Feedback Model, NACKs from the receiver observing the loss will be reflected to the other receivers, and there's no need for distribution source to create the third-party loss report. The distribution source that has not seen the actual packet loss should pass through any Third Party Loss Report message it receives from the upstream direction.

This RTCP Third Party Loss Report message lets the receivers know that the sender of the Third party Loss Report has received reports that the indicated packets were lost and feedback for this packet loss is not needed and should not be sent to the media source(s). If the media source(s) are part of the SSM group for RTCP packet reflection, the Distribution Source must filter this packet out. If the media source(s) are not part of the SSM group for RTCP packets, the Distribution Source must not forward this RTCP Third Party Loss Report message to the media source(s).

[6.1.2.](#) Distribution Source Feedback Summary Model

In the distribution source feedback summary model, there may be multiple distribution sources who see the actual packet loss. In this section, we focus on this generic case to discuss the distribution Source Feedback Summary Model.

The distribution source A must listen on the RTP channel for data. When the distribution source A observes RTP packets from a media source are not consecutive by checking the sequence number of packets, the distribution source A generates the NACK message, and then send it to receivers in the downstream path via the multicast channel.

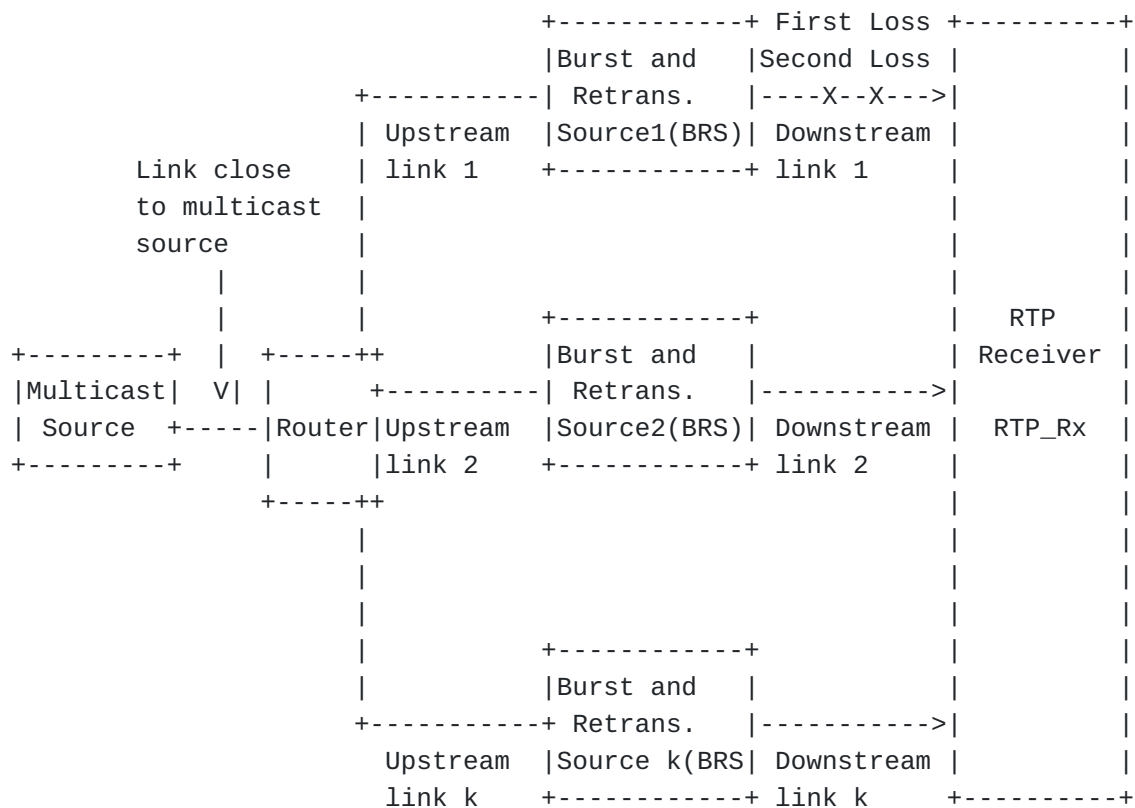
The Distribution Source B must also listen for RTCP data sent to the RTCP port. Upon receiving the RTCP Third Party Loss Report from the Distribution Source A, the Distribution Source B needs to check whether it sees the same event reported both from upstream distribution source A and downstream receiver. If the upstream Third Party Loss Report reports the different event, the distribution source B passes through any Third Party Loss Report message it receives from the upstream direction. If the same event is reported from both distribution source A and downstream receiver of distribution source B, the distribution source B may suppress creating and sending its own report with the same event to the downstream RTP receiver.

Also the Distribution Source B may create and send its own Third Party Loss Report described in the [Section 4](#) to the group over the multicast

RTCP channel in response to NACKs received from downstream. if downstream loss is reported using NACK to the distribution source B.

6.2. Unicast based Rapid Acquisition of Multicast Stream (RAMS) use case

The typical RAMS architecture [\[I-D.ietf-avt-rapid-acquisition-for-rtsp\]](#) may have several Burst/Retransmission Sources(BRS) behind the multicast source (MS) These BRSes will receive the multicast SSM stream from the media source. If one of the BRSes receives downstream loss report (i.e., First loss in [Figure 4](#)) on its downstream link, but the others BRSes have not, as the packet loss took place on the SSM tree branch that does not impact the other BRSes. In such case, the BRSes not being impacted are not aware of downstream loss at their downstream link, therefore these BRSes will not create new Third Party Loss Report message and send it to receivers in their downstream path. If the BRS impacted by packet loss has been told the actual packet loss, the BRS MAY choose to create new Third Party Loss Report message and send it to the receivers in the downstream link. Note that BRS must use its own SSRC as packet sender SSRC for transmitting the feedback suppress message.



The BRS may also send a NACK upstream to request the retransmitted packet. Upon receiving the retransmitted packet, the BRS sent it downstream. Note that this retransmitted packet may get lost (i.e., second loss in the [Figure 4](#)) on the downstream link. In order to deal

with this issue, the downstream receiver can start a timeout clock in which it expected to get a retransmission packet. When this timeout expires and there is no retransmitted packet or a new Third Party Loss Report message, it can take its normal behavior as if there is no current retransmission suppression in place.

6.3. RTP transport translator use case

A Transport Translator (Topo-Trn-Translator), as defined in [\[RFC5117\]](#) is typically forwarding the RTP and RTCP traffic between RTP clients, for example converting between multicast and unicast for domains that do not support multicast. The translator can identify packet loss from the upstream and send the Third Party Loss Report message to the unicast receivers. Note that the translator must be a participant in the session and can then use its own SSRC as packet sender SSRC for transmitting the Third Party Loss Report message

6.4. Multipoint Control Unit (MCU) use case

In point to multipoint topologies using video switching MCU (Topo-Video-switch-MCU) [\[RFC5117\]](#), the MCU typically forwards a single media stream to each participant, selected from the available input streams. The selection of the input stream is often based on voice activity in the audio-visual conference, but other conference management mechanisms (like presentation mode or explicit floor control) exist as well. In this case the MCU may detect packet loss from the sender or may decide to switch to a new source. In both cases the receiver may lose synchronization with the video stream and may send a FIR request. If the MCU itself can detect the mis-synchronization of the video, the MCU can send the FIR suppression message to the receivers and send a FIR request to the video source. As suggested in RFC 5117, this topology is better implemented as an Topo-mixer, in which case the mixer's SSRC is used as packet sender SSRC for transmitting Third Party Loss Report message.

7. Security Considerations

The defined messages have certain properties that have security implications. These must be addressed and taken into account by users of this protocol.

Spoofed or maliciously created feedback messages of the type defined in this specification can have the following implications:

Sending Third Party Loss Report with wrong sequence number of lost packet that makes missing RTP packets can not be compensated.

To prevent these attacks, there is a need to apply authentication and integrity protection of the feedback messages. This can be accomplished against threats external to the current RTP session using the RTP profile that combines Secure RTP [\[RFC3711\]](#) and AVPF into SAVPF [\[RFC5124\]](#).

Note that middleboxes that are not visible at the RTP layer that wish to send Third Party Loss Reports on behalf of the media source can only do so if they spoof the SSRC of the media source. This is difficult in case SRTP is in use. If the middlebox is visible at the RTP layer, this is not an issue, provided the middlebox is part of the security context for the session.

Also note that endpoints that receive a Third Party Loss Report would be well-advised to ignore it, unless it is authenticated via SRTCP or similar. Accepting un-authenticated Third Party Loss Report can lead to a denial of service attack, where the endpoint accepts poor quality media that could be repaired.

8. IANA Consideration

New feedback type and New parameters for RTCP Third Party Loss Report are subject to IANA registration. For general guidelines on IANA considerations for RTCP feedback, refer to [\[RFC4585\]](#).

This document assigns one new feedback type value x in the RTCP feedback report registry to "Third Party Loss Report" with the following registrations format:

Name:	TPLR
Long Name:	Third Party Loss Report
Value:	TBD
Reference:	This document.

This document also assigns the parameter value y in the RTCP TPLR feedback report Registry to "Transport Layer Third Party Loss Early Indication", with the following registrations format:

Name:	TLLEI
Long name:	Transport Layer Third Party Loss Early Indication
Value:	TBD
Reference:	this document.

This document also assigns the parameter value z in the RTCP TPLR feedback report Registry to "Payload Specific Third Party Loss Early Indication", with the following registrations format:

Name:	PSLEI
Long name:	Payload Specific Third Party Loss Early Indication
Value:	TBD
Reference:	this document.

Qin Wu
sunseawq@huawei.com
101 Software Avenue, Yuhua District
Nanjing, Jiangsu 210012, China

The contact information for the registrations is:

9. Acknowledgement

The authors would like to thank David R Oran, Ali C. Begen, Colin Perkins, Tom VAN CAENEGEM, Ingemar Johansson S, Bill Ver Steeg, Jonathan Lennox, WeeSan Lee for their valuable comments and suggestions on this document.

10. References

10.1. Normative References

[RFC5760]	Ott, J., Chesterfield, J. and E. Schooler, " RTP Control Protocol (RTCP) Extensions for Single-Source Multicast Sessions with Unicast Feedback ", RFC 5760, February 2010.
[RFC2119]	Bradner, S., " Key words for use in RFCs to Indicate Requirement Levels ", BCP 14, RFC 2119, March 1997.
[RFC4585]	Ott, J., Wenger, S., Sato, N., Burmeister, C. and J. Rey, " Extended RTP Profile for Real-time Transport Control Protocol (RTCP)-Based Feedback (RTP/AVPF) ", RFC 4585, July 2006.
[RFC3550]	Schulzrinne, H., Casner, S., Frederick, R. and V. Jacobson, " RTP: A Transport Protocol for Real-Time Applications ", STD 64, RFC 3550, July 2003.
[RFC5117]	Westerlund, M. and S. Wenger, " RTP Topologies ", RFC 5117, January 2008.
[RFC4588]	Rey, J., Leon, D., Miyazaki, A., Varsa, V. and R. Hakenberg, " RTP Retransmission Payload Format ", RFC 4588, July 2006.
[RFC4566]	Handley, M., Jacobson, V. and C. Perkins, " SDP: Session Description Protocol ", RFC 4566, July 2006.
[RFC5234]	Crocker, D. and P. Overell, " Augmented BNF for Syntax Specifications: ABNF ", STD 68, RFC 5234, January 2008.
[RFC5104]	Wenger, S., Chandra, U., Westerlund, M. and B. Burman, " Codec Control Messages in the RTP Audio-Visual Profile with Feedback (AVPF) ", RFC 5104, February 2008.
[RFC3711]	Baughner, M., McGrew, D., Naslund, M., Carrara, E. and K. Norrman, " The Secure Real-time Transport Protocol (SRTP) ", RFC 3711, March 2004.
[RFC5124]	Ott, J. and E. Carrara, " Extended Secure RTP Profile for Real-time Transport Control Protocol (RTCP)-Based Feedback (RTP/SAVPF) ", RFC 5124, February 2008.

10.2. Informative References

[RFC5740]	
-----------	--

	Adamson, B., Bormann, C., Handley, M. and J. Macker, "NACK-Oriented Reliable Multicast (NORM) Transport Protocol", November 2009.
[DVB-IPTV]	ETSI Standard, "Digital Video Broadcasting(DVB); Transport of MPEG-2 TS Based DVB Services over IP Based Networks", ETSI TS 102 034, V1.4.1 , August 2009.
[I-D.ietf-avt-rapid-acquisition-for-rtp]	Steeg, B., Begen, A., Caenegem, T. and Z. Vax, "Unicast- Based Rapid Acquisition of Multicast RTP Sessions", November 2010.
[I-D.hunt-avt-monarch-01]	Hunt, G. and P. Arden, "Monitoring Architectures for RTP", August 2008.
[I-D.ietf-pmol-metrics-framework-02]	Clark, A., "Framework for Performance Metric Development", .

[Appendix A.](#) **Appendix A. Change Log**

Note to the RFC-Editor: please remove this section prior to publication as an RFC.

[Appendix A.1.](#) **draft-ietf-avtcore-feedback-suppression-rtp-01**

The following are the major changes compared to previous version:

- *Remove the merge report from SSM use case and additional text to address report merging issue.
- *Revise section 3 and section 6 to address FEC packet dealing issue and Leave how to repair packet loss beyond the scope.
- *Modify the SSM use case and RAMS use case to focus on uses.
- *Other Editorial changes.

[Appendix A.2.](#) **draft-ietf-avtcore-feedback-suppression-rtp-02**

The following are the major changes compared to previous version:

- *In Appendix A, fix typo: Appendix A. Appendix A. -> Appendix A.
- *In Section 4.1, fix typo: Section 4.3.1.1 of section [RFC5104]-> section 6.2.1 of [RFC4585].
- *In Section 3: Clarify how to deal with downstream loss using Third party loss report and upstream loss using NACK.
- *Update title and abstract to focus on third party loss report.

*In Section 6.1: Update this section to explain how third party loss report is used to deal with downstream loss.

*In section 6.1.2: Update this section to explain how third party loss report is used to deal with downstream loss.

*In section 6.2: Rephrase the text to discuss how BRS deal with the third party loss report.

Authors' Addresses

Qin Wu Wu Huawei 101 Software Avenue, Yuhua District Nanjing,
Jiangsu 210012 China EMail: sunseawq@huawei.com

Frank Xia Xia Huawei 1700 Alma Dr. Suite 500 Plano, TX 75075 USA
Phone: +1 972-509-5599 EMail: xiayangsong@huawei.com

Roni Even Even Huawei 14 David Hamelech Tel Aviv 64953 Israel EMail:
even.roni@huawei.com