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Low Delay RTCP Feedback Format

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1. Abstract

Low delay feedback in RTP has gained a lot of interest recently in order to enhance the performance in RTP sessions with two or only a small number of participants. A recent document describes rules that enhance the functionality of the existing RTP timing and allow low delay feedback. While that document describes when it is allowed to send data, we describe here what kind of low delay feedback may be sent. Therefore we define three general RTCP feedback message types as well as some recommendations how the sender SHOULD react to certain feedback. The message types can either be used as is or they can be extended in payload or application specific ways.

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 [RFC-2119](#) [2].

3. Introduction

RTP and its profile define strict rules on the timing behavior of RTCP feedback. It is very important for large multicast groups to define such rules, because the traffic on the feedback channel might explode otherwise. However for small multicast groups or even unicast, a frequently used application of RTP, the rules can be less strict.

A new feedback timing behavior for RTP is defined in [8]. The document describes in certain scenarios when it is allowed to send feedback. These new timing rules scale from unicast connections, where feedback is possible anytime immediately to small multicast groups, where feedback can be sent with low delay according to a credit based scheduling scheme.

While [8] describes the behavior when the receiver is allowed to send feedback, this document describes a framework what is going to be sent. [Section 4](#) describes three general feedback packet types. General formats and semantics are given, but room is left for the extension to specific feedback messages. [Section 5](#) describes some recommendations how the sender SHOULD react if certain feedback is received. Especially congestion control items are discussed.

4 Low Delay Feedback Packets

In this document we define a framework of low delay RTCP feedback messages. Therefore three different types of messages are defined, which are categorized as follows:

- Transport Layer Feedback Messages
- Payload Specific Feedback Messages
- Application Layer Feedback Messages

Transport Layer Feedback Messages are used to transmit protocol related information from the receiver to the sender. This

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	2	3	4	5	6	7	8	9	0	1	2	3	4	5	6	7	8	9
V=2 P E FMT										PT=RTPFB										length																			
										SSRC of packet sender																													
										SSRC of media source																													
:										FCI																				:									
:																																							

The various fields V, P, SSRC and length are defined in the RTP specification [3]. We summarize the meaning of all of the fields below:

version (V): 2 bits

This field identifies the RTP version. The current version is 2.

padding (P): 1 bit

If set, the padding bit indicates that the packet contains additional padding octets at the end which are not part of the control information but are included in the length field.

early packet (E): 1 bit

This bit is set if the packet is sent early. A detailed description can be found in [8].

feedback message type (FMT): 4 bits

This field identifies the type of the feedback. The packet can be used for different purposes. The following types are defined (up to now):

- 0: forbidden
- 1: General NACK
- 2: General ACK
- 3-15: reserved

packet type (PT): 8 bits

This is the RTCP packet type which identifies the packet as being an RTP Feedback Message.

length: 16 bits

The length of this packet in 32-bit words minus one, including the header and any padding. This conforms with the definition of the length field used in RTCP sender and receiver reports [3].

SSRC of packet sender: 16 bits

The synchronization source identifier for the originator of this packet.

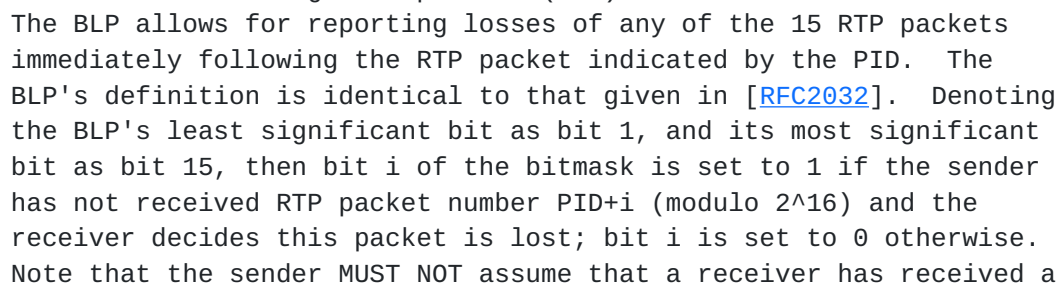
SSRC of media source: 16 bits

The synchronization source identifier of the media source that this

feedback control information (FCI): variable length
Format and semantic of the FCI varies for the different message types. See sections [5.1.1](#) and [5.1.2](#) for the definition of the syntax and the semantic of this field for the general NACK and ACK packets.

In this section the general NACK packet is described. As said above, the general NACK packet is of the type Transport Layer Feedback Message. The basic header of [section 5.1](#) applies for this packet, where FMT is set to 1 (General NACK).

The Feedback control information (FCI) field has the following syntax:



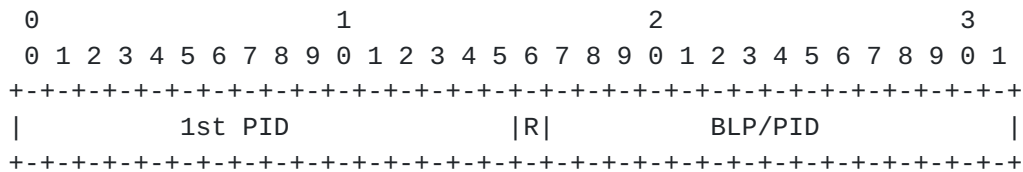
packet because its bitmask was set to 0. For example, the least significant bit of the BLP would be set to 1 if the packet corresponding to the PID and the following packet have been lost. However, the sender cannot infer that packets PID+2 through PID+15 have been received simply because bits 2 through 15 of the BLP are 0; all the sender knows is that the receiver has not reported them as lost at this time.

4.1.2 General ACK

The general ACK packet is also of the type Transport Layer Feedback Message and therefore uses the header as defined in [section 5.1](#) with FMT=2 (general ACK).

The general ACK packet is used to indicate that one or several RTP packets are received correctly. Therefore the received packet(s) are identified by the means of a packet identifier and a bitmask. Acking of a range of consecutive packets is also possible.

The Feedback control information (FCI) field has the following syntax:



Packet ID (1st PID): 16 bits

This PID field is used to specify a correctly received packet. RTP Payload Formats may decide how to identify a packet. Typically RTP sequence number is used for PID as the default format.

Range of ACKs (R): 1 bit

The R-bit indicates that a range of consecutive packets are received correctly. The 1st PID field specifies the first packet of that range and the next field (BLP/PID) will carry the last PID of the range that is acknowledged.

bitmask of following lost packets / Packet ID (BLP/PID): 15 bits

The semantic of this field changes according to the value of the R-

field. If R=1, this field is used to identify the last packet of the acknowledged packet range, as described above.

4.2 Payload Specific Feedback Messages

The basic header format is described below.

The various fields V, P, SSRC and length are defined in the RTP specification [3]. We summarize the meaning of all of the fields below:

If set, the padding bit indicates that the packet contains additional padding octets at the end which are not part of the control information but are included in the length field.

early packet (E): 1 bit

This bit is set if the packet is sent early. A detailed description can be found in [8].

feedback message type (FMT): 4 bits

This field identifies the type of the feedback. Every RTP Payload Format has its own FMT number space, i.e. a certain FMT number can have different meanings for different payload types. It is therefore expected that either the RTP Payload Format documents itself or additional feedback format documents (related to one or several Payload Formats) define the FMT numbers and their meanings. FMT=0 is forbidden.

packet type (PT): 8 bits

This is the RTCP packet type which identifies the packet as being an Payload Specific Feedback Message.

length: 16 bits

The length of this packet in 32-bit words minus one, including the header and any padding. This conforms with the definition of the length field used in RTCP sender and receiver reports [3].

SSRC of packet sender: 16 bits

The synchronization source identifier for the originator of this packet.

SSRC of media source: 16 bits

The synchronization source identifier of the media source that this feedback is related to.

RTP payload type (RTP PT): 7 bits

This field indicates the payload format to which the feedback is related. It is important to use that field, because the session could be used with several payload formats on the forward direction and the semantics of the packet is different for different payload formats.

feedback control information (FCI): variable length

Format and semantic of the FCI varies for the different message types. The type of FCI that is used is indicated by the combination of FMT and RTP PT. The mapping of FMT numbers to FCIs is expected to

be described in additional payload specific documents or the RTP Payload Formats.

4.2.1 Example

This section describes a short example how the payload specific feedback messages can be defined and used.

As an example we consider a predictive coded video application. If packet loss occurs, it is important that the sender gets this information, because it will otherwise send data, which the receiver might not be able to use.

Therefore an additional document can describe a feedback message, which works as a picture or slice loss indication. If the sender receives this message, it might start with a new intra coded frame, which the receiver would be able to use properly.

The document must specify the format of the FCI-field, one or several RTP payload formats, for which this should be used and a FMT number that is not allocated in all of these formats. A different possibility is to define the FMT number and FCI syntax directly in a video RTP Payload Format.

4.3 Application Layer Feedback Messages

These messages are used to transport application defined data directly from the receiver's to the sender's application. The data that is transported is not identified by the feedback message. Therefore the application must be able to identify the messages payload.

Usually applications define their own set of messages, e.g. NEWPRED messages in MPEG-4 or feedback messages in H.263/Annex N,U. These messages do not need any additional information from the RTCP message and thus this message has the following simple 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	2	3	4	5	6	7	8	9	0	1								
V=2 P E FMT										PT=AFB										length																			
										SSRC of packet sender																													
										SSRC of media source																													
:										Application Message										:																			

:

:

The various fields V, P, SSRC and length are defined in the RTP specification [3]. We summarize the meaning of all of the fields below:

version (V): 2 bits

This field identifies the RTP version. The current version is 2.

early packet (E): 1 bit

This bit is set if the packet is sent early. A detailed description can be found in [8].

padding (P): 1 bit

If set, the padding bit indicates that the packet contains additional padding octets at the end which are not part of the control information but are included in the length field. The last octet of the padding is a count of how many padding octets should be ignored.

feedback message type(FMT): 4 bits

This field identifies the type of the feedback. The packet can be used for different purposes. No type is defined up to now:

0: forbidden

1-15: reserved

packet type (PT): 8 bits

This is the RTCP packet type which identifies the packet as being an Application Feedback Message.

length: 16 bits

The length of this packet in 32-bit words minus one, including the header and any padding. This conforms with the definition of the length field used in RTCP sender and receiver reports [3].

SSRC of packet sender: 16 bits

The synchronization source identifier for the originator of this packet.

SSRC of media source: 16 bits

The synchronization source identifier of the media source that this feedback is related to.

Application Message (FCI): variable length

This field contains the original application message that should be transported from the receiver to the source. The format is application dependent. The length of this field is variable. If the application data is not four-byte aligned, padding must be added.

4.3.1 Example

An example usage of the Application Feedback Message is NEWPRED for MPEG-4. The NEWPRED messages are defined in the MPEG-4 specification. The MPEG-4 entity at the receiver sends the message, which is transported in the application feedback packet. It is then given to the MPEG-4 entity at the sender, which can identify the information as a NEWPRED message and knows what to do with it.

Normally one NEWPRED message is mapped onto one AFM-RTCP packet, and several messages could be continuously mapped onto one AFM-RTCP packet. In the case several messages are mapped onto one AFM-RTCP packet, padding should only be required on the last individual message. One message SHALL NOT be fragmented into different AFM-RTCP packets.

5. Reaction to Feedback

In the previous sections the feedback messages were defined as well as the timing rules when it is allowed to send these messages. We defined only the feedback messages, because we cannot foresee all purposes for which application designers would want to use the low delay feedback and thus the reaction to the messages is generally left to the application. However this section describes some rules and recommendations of what the sender SHOULD or MAY do in the event of receiving a low delay feedback message.

5.1 Congestion Control

Congestion control for real-time or near real-time traffic, especially for unicast, is not new. Recently a mechanism called TFRC

(TCP-Friendly-Rate-Control), [9], was introduced in the TSV working group for standardization. This mechanism controls the sending bit rate according to monitored network conditions in a TCP friendly way, i.e. in the long-term, the RTP traffic shares the bandwidth fair with TCP connections. Other mechanisms are subject of research everywhere and may be applicable for the use in RTP.

Low delay feedback supports the use of these congestion control algorithms. Due to the frequent feedback messages, it is possible for the sender to monitor the network state closely and therefore it is possible to react to upcoming congestion in time. This minimizes the congestion related packet loss and serves the network stability.

A congestion control algorithm that shares the available bandwidth fair with competing TCP connections, e.g. TFRC, SHOULD be used if the low delay RTP session is transmitted in an best effort environment.

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5.2 Reactions to Feedback Messages

As described in [section 4](#) feedback messages are either processed directly by the RTP entity or the application that uses RTP. However the exact behavior is implementation, negotiation and environment dependent. In this section we describe how the sender may react and what he should not do.

At reception of a Transport Layer Feedback Message, the RTP entity evaluates the message and reacts to it. At detection of a packet loss, either by receiving a NACK, missing an ACK or other means, the RTP entity MAY invoke error resilience features such as retransmissions. However if the session is transmitted in an best effort environment, the sender SHOULD NOT increase the sending bit rate at detection of a packet loss. Possible reactions are to retransmit important data, while non-important packets that were scheduled for transmission are discarded in the sender.

At reception of a Payload Type or Application Defined Message the application related part is given to the application, while the attached receiver report (all feedback is sent at least in minimum compound packets) SHOULD be evaluated in the RTP entity and MAY serve as the input for the congestion control algorithm (see previous section for details about congestion control).

6. Security Considerations

Security is not considered in this draft.

7. References

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