RTP Congestion Control Feedback

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Reporting on draft-ietf-avtcore-cc-feedback-message-09, that’s is co-authored with Zahed Sarker, Varun Singh, and Michael A. Ramalho
Draft Status

- draft-ietf-avtcore-cc-feedback-message-08 went to IESG review
- Comments from TSVART, GenART, IANA, Benjamin Kaduk, Éric Vyncke, Erik Kline, Magnus Westerlund, Martin Duke, and Robert Wilton
- Mostly minor editorial nits – a small number of technical changes to review
Editorial Changes

- Add to Abstract: “An effective RTP congestion control algorithm requires more fine-grained feedback on packet loss, timing, and ECN marks than is provided by the standard RTP Control Protocol (RTCP) Sender Report (SR) and Receiver Report (RR) packets”
- Don’t reference the terminology defined in RFCs 3551 and 3611, since those RFCs don’t define terminology
- Clarify: “RTCP packets do not contain a sequence number, so loss of feedback packets has to be inferred based on the time since the last feedback packet.”
- Clarify that it’s the media sender that should reduce its rate if multiple feedback packets are lost
- Since draft-alvestrand-rmcat-remb expired long ago, reference “Mechanisms that convey the receiver's estimate of the maximum available bit-rate” rather than that explicit draft
Packet Format Description

The contents of each 16-bit packet metric block comprises the L, ECN, and ATO fields as follows:

- **L**: 1 bit. This field is a boolean to indicate if the packet was received. 0 represents that the packet was not yet received, and all the subsequent bits (ECN and ATO) are also set to 0. 1 represents that the packet was received and the subsequent bits in the block need to be parsed.

- **Received (R)**: 1 bit. This field is a boolean to indicate if the packet was received. 0 represents that the packet was not yet received, and the subsequent 15-bits (ECN and ATO) in this 16-bit packet metric block are also set to 0 and MUST be ignored. 1 represents that the packet was received and the subsequent bits in the block need to be parsed.

- Rename field L → R to match usage

- Clarify that if the packet is lost, the bits indicating its arrival time and ECN mark are set to zero and MUST be ignored
Relation to RFC 6679 ECN Feedback

The RTP ECN Feedback Packet is not useful when ECN is used with the RTP Congestion Control Feedback Packet defined in this memo since it provides duplicate information. Accordingly, when congestion control feedback is to be used with RTP and ECN, the SDP offer generated MUST include an "a=ecn-capable-rtp:" attribute to negotiate ECN support, along with an "a=rtp-fb:" attribute with the "ack" parameter "ccfb" to indicate that the RTP Congestion Control Feedback Packet is to be used for feedback. The "a=rtp-fb:" attribute MUST NOT include the "nack" parameter "ecn", so the RTP ECN Feedback Packet will not be used.

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- Change ECN feedback signalling, to allow more graceful fallback to RFC 6679 format
- Mandate that the guidelines in RFC 6679 for checking that ECN actually works on the path need to be followed
Clarify Security Considerations

11. Security Considerations

The security considerations of the RTP specification [RFC3550], the applicable RTP profile (e.g., [RFC3551], [RFC3711], or [RFC4585]), and the RTP congestion control algorithm that is in use (e.g., [RFC8698], [RFC8298], [I-D.ietf-rmcat-gcc], or [RFC8382]) apply.

A receiver that intentionally generates inaccurate RTCP congestion control feedback reports might be able trick the sender into sending at a greater rate than the path can support, thereby causing congestion on the path. This will negatively impact the quality of experience of that receiver. Since RTP is an unreliable transport, a sender can intentionally leave a gap in the RTP sequence number space without causing harm, to check that the receiver is correctly reporting losses.

An on-path attacker that can modify RTCP congestion control feedback packets can change the reports to trick the sender into sending at either an excessively high or excessively low rate, leading to denial of service. The secure RTCP profile [RFC3711] can be used to authenticate RTCP packets to protect against this attack.

An off-path attacker that can spoof RTCP congestion control feedback packets can similarly trick a sender into sending at an incorrect rate, leading to denial of service. This attack is difficult, since the attacker needs to guess the SSRC and sequence number in addition to the destination transport address. As with on-path attacks, the secure RTCP profile [RFC3711] can be used to authenticate RTCP packets to protect against this attack.
Next Steps

• Is the WG okay with these changes?

• If yes, IESG approves and this goes to RFC Editor
• If no, need update and new review cycle