# Explicit Congestion Notification (ECN) for RTP over UDP

draft-ietf-avtcore-ecn-for-rtp-01

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#### Outline

- Important Changes
- Open Issues
  - 1. Saturation of the packet loss counters
  - 2. Initiation Optimization for Multi-SSRC per host sessions
  - 3. Congestion reporting in bytes or packets?
  - 4. ECN setting capability direction
- Next Step

### Important Changes since AVT-03 version

- Restructured the sections around signalling
  - Attempted to clarify use of "a=ecn-capable-rtp" SDP attribute at media and/or session level – further updates needed; no reason to disallow at session level
  - Added signalling parameters for the Feedback and RTCP XR packets
  - Requiring the "a=ice-options:rtp+ecn" when using ICE initiation
  - Added examples for the signalling using SDP
- Included the proposal to give more flexibility for congestion control algorithms on how to interpret CE marks
  - e.g., to support some authorised emergency responder scenarios
- Added section on Interoperability
- Clarified the roles of sender and receiver in regards to the ECN solution
- > Clarified "all known receivers"
  - i.e., all SSRC's that aren't local to the sending SSRC
  - Also forbid the usage of ECN and Sampling of Group Membership [RFC2762]
- Clarified how to handle non ECT end-points, both aware and non-aware of the ECN signalling

#### Issue: Saturation in Packet loss counter

- > RTCP RR and ECN feedback use packet loss counters that are signed integers
  - RTCP RR/SR report block: 24 bits; ECN Feedback format: 12-bit
  - Positive values indicate loss, negative values duplication
  - Counters clamp if they reach their maximum value, and do not wrap
- How is the 12-bit lost packet counter in ECN feedback packets derived?
  - Copy lower 11-bits of RTCP RR lost packet counter + plus the 24<sup>th</sup> sign bit into the 12-bit ECN feedback lost packet counter?
    - That way one can handle both 0 transitions and wrapping by using the long RTCP RR/SR report block counter as base value for extending it.
    - Works unless cumulative change between previous report is more than 1024 losses or duplication, in which case uncertainty may occur
  - Replace with an unsigned counter of lost packets, that wraps on overflow?
- How to deal with saturation of the lost packet counters?
  - If the saturating format is used, saturation implies that packet loss can no longer be reported – may need to change SSRC to report further loss?
    - Also can issue with standard RFC 3550 RTCP SR/RR packets
  - Might be appropriate to use an unsigned packet loss counter?

## Issue: Initiation of multi-SSRC per host sessions

- > Draft contains an optimisation for unicast sessions:
  - "As an optimisation, if an RTP sender is initiating ECN usage towards a unicast address, then it MAY treat the ECN initiation as provisionally successful if it receives a single RTCP ECN feedback report indicating successful receipt of the ECT-marked packets, with no negative indications, from a single RTP receiver."
- > The restriction on "single RTP receiver" maybe unnecessary strict:
  - Some uses of RTP use multiple SSRCs per host and so may be interpreted as being more than one receiver (e.g., SSRC multiplexed RTP retransmission [RFC4588])
- > Should we relax this restriction?
  - Rephrase as "from a single destination host"?
    - Fails if the destination is a star topology packet-relay translator
  - Allow successful ECN initiation for one SSRC to imply success for other SSRCs with the same CNAME?
    - > Fails for cases where multiple hosts collude to form an RTP endpoint
  - Ignore the issue further optimisation is not worth the complexity
    - > Everything works, but ECN initiation is slow
  - Preference: ignore this for now, can optimise later if it becomes a real problem

## Issue: Report congestion in bytes or packets?

- > RTCP reports congestion in terms of packets lost
- > For ECN feedback, we can report the number of packets marked, or the number of bytes marked
  - The two options can give different behaviour, especially when considering translators that fragment and reassemble packet
  - ECN community would prefer byte marking
- Inconsistency with RTCP report problematic, though
  - Loss reported in packets, ECN marks reported in bytes
- > We propose to report ECN marks in packets

### Issue: ECN setting capability direction

- No point negotiating ECN if no-one can actually set the ECT bits
- > This appears to not be an significant issue
  - In SSM usages, the sender will know of its capability prior to creating any SDP
  - For ASM with centralized O/A signalling with Application server
    - The server can determine if no participant is capable and re-invite without ECN if desired
  - For ASM with declarative SDP
    - > In this case there is no way to determine that no senders supports ECN
- Not a significant issue, remove the issue

#### **Next Steps**

- > Resolve Open Issues
- > Submit an update
  - Open issue resolutions
  - Editorial pass
  - Addressing your comments
- > Aim at WG last call ready with next version