
Audio/Video Transport Working Group

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Wednesday Agenda

- Introduction and status 10
- RTP spec and profile Last Call 30
- RTCP-based retransmission 10
- Additional payload formats
 - » Pointers 5
 - » DV video, DV audio 20
 - » MPEG 2 AAC 10
 - » Loss-tolerant MP3 10
 - » MPEG 2 Transport Stream 10
 - » HDTV and AC3 5

Thursday Agenda

- Conversational text payload format 5
- Transport of MPEG-4 60
 - » SL-based payload format 5
 - » Generic, error resilient payload format 5
 - » Elementary stream payload format 5
- RTP header compression / muxing 50
 - » CRTP enhancements + PPP mux + tunnel 10
 - » Dynamic Nx64 payload format 10
- Charter bashing 10
 - » Multicast feedback / new Profiles?

RTP Drafts in Process

- Drafts at IESG awaiting publication:
 - » FEC payload format (approved at PS)
 - » Payload format guidelines (approved BCP)
 - » SSRC Sampling (Last Call requested)
 - » RTP MIB (Last Call requested)
- Drafts in WG last call:
 - » DTMF tones payload format
 - » RTP spec and profile package

Status of RTP

- RFC1889, 1890 published as Proposed Standards in January 1996
- WG Last Call issued for Internet-Draft revisions for Draft Standard, ends now!
 - » Spec is [draft-ietf-avt-rtp-new-05.ps,txt](#)
 - » Profile is [draft-ietf-avt-profile-new-07.ps,txt](#)
 - » Package with companion drafts...

Companion Drafts for RTP Spec

- **MIME reg., RTCP bw specs – Proposed**
 - » [draft-ietf-avt-rtp-mime-01.txt](#)
 - » [draft-ietf-avt-rtcp-bw-00.txt](#)
- **SSRC sampling – Experimental**
 - » [draft-ietf-avt-rtpsample-04.txt](#)
- **RTP interoperability statement**
 - » [draft-ietf-avt-rtp-interop-02.txt](#)
 - » [draft-ietf-avt-profile-interop-00.txt](#)
- **RTP/RTCP testing, scalability – Inform.**
 - » [draft-ietf-avt-rtptest-02.txt](#)

Recent Changes to RTP Spec

- Collision/loop detection algorithm nesting syntax changed to pseudo-C
- Added SHOULD keep packets from new address on third-party collision for apps that may change addresses (e.g., where some endpoints are mobile)
- Reverse reconsideration algorithm SHOULD be performed to possibly reduce the delay before sending first SR packet after starting to send data
- Timing out a participant is based on inactivity for a number of RTC report intervals calculated using the receiver RTCP bw fraction even for active senders

Questions on RTP Spec

- Says sources that change their source transport address **MUST** change SSRC — inconsistent with “SHOULD follow new address”
- Should length of RTCP with info about N sources be divided by N for average?
 - » Clarify that a mixer mixing N sources may use N times the single-source RTCP bw

Recent Changes to RTP Profile

- CN (comfort noise) static payload type changed back from 19 to 13 for consistency
- RTP clock rate G722 changed back to 8000 even though sampling rate is 16000 (a note of explanation was added)
- Note that MPEG audio RTP clock rate is always 90000 regardless of sampling rate

Questions on RTP Profile

- What should we say about recommending H263-1998 over H263?
- Did AIFC list of channels change from what is in RFC 1890?
- Change title of Section 3 to “IANA Considerations”
- Bungled edit of CN from 13 to 19 in Table 5
- Proposed enhancements to CN to include spectral shape parameters

MIME Registration

- New draft-ietf-avt-rtp-mime-01.txt
 - » clarified mapping from MIME parms to fmtp
 - » added RED (redundant audio)
 - » added “type” parameter for MPEG
- FEC payload format does not contain a MIME registration for “parityfec”
 - » make 1-page registration draft for this, or
 - » add it to the rtp-mime draft (but it isn't a payload format in the Profile)

Need Interoperability Statements

- Two drafts list the interoperability points:
 - » draft-ietf-avt-rtp-interop-02.txt
 - » draft-ietf-avt-profile-interop-00.txt (not submitted)
- Please submit for your implementation
— need not be complete
- If there are political problems, please let us know