Audio Video Transport Working Group

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Wednesday Agenda

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» Drafts in process; drafts to act upon	
Status of RTP & H.Multimedia MIB	5
RTP multiplexing proposals	45
» Rosenberg, Hoshi, Subbiah	
» Discussion	30
DMIF for RTP/MPEG4	10
» Discussion	15

Thursday Agenda

RTP spec and profile issues	30
» Registering encodings as MIME types	
» Improvement to SSRC scaling	10
Update on RTP redundancy mech.	10
FEC payload format	15
AVT revised charter bashing	15

Status of RTP

- RFC1889, 1890 published as Proposed Standards in January 1996
- □ Internet-Draft revisions for Draft Std.
 - » Spec is draft-ietf-avt-rtp-new-01.ps,txt
 - » Profile is <u>draft-ietf-avt-profile-new-03.ps,txt</u>
- Spec mostly done, Profile needs more
- Plan: Complete revisions, then Last Call for Draft Standard

If you haven't read it yet...

Please see <u>draft-ietf-avt-rtp-new-01.txt</u> sections:

- Sec 0: Resolution of open issues
- □ Sec 6.2: RTCP Transmission Interval
- Sec 6.3: RTCP packet send and receive rules
- Appendix B: Changes from RFC1889
- Extra credit: check code in Appendix A

RTP Drafts in Process

- RFCs recently published:
 - » 2343: Bundled MPEG payload (Exp.)
 - » 2354: Options for Repair (Info.)
- Drafts submitted for publication:
 - » IP/UDP/RTP header compression: at IESG
 - » JPEG payload format revision: IESG ballot
 - » BT656 payload format: in Last Call
 - » H.263+ payload format: in AD's queue

Drafts to act upon

- Guidelines for RTP payload formats
 draft-ietf-avt-rtp-format-guidelines-00.txt,.ps
 - » Part of a framework to facilitate continuing long-term development of payload formats
- PureVoice (QCELP) payload format draft-mckay-qcelp-01.txt
 - » Ready for Last Call?
- Generic payload formats to be merged

Multiplexing: Key Question

Who gets to do the multiplexing?

- » why: common handling, reduce overhead
- » when: separate or bundled?
- » where: what protocol level?
- » how: application specific or general

Points to consider in this discussion:

- Can/should we generalize current proposals?
- Philosophy: keep muxing at one level if possible

RTP Multiplexing

When is mux useful, and how many bits?

- » Is multiplexing format a payload type?
- » Inband length field or fixed length?
- » How many bits of ID per multiplexed user?
- » Payloads time aligned, close, or arbitrary?
- » CSRC list and extension per user?
- » RTCP per user or per multiplexed stream?
- » How many users per multiplexed stream?

Recent Changes to RTP Spec

- Added fudge factor (e 3/2) in timer reconsideration
- Added fix for underestimate with sampled SSRC algorithm when size decreases
- RTCP sender & receiver BW may be parameters
- RTCP min interval may scale smaller for high BW sessions, and zero initial delay for unicast sessions
- Specified P-bit padding for RTCP only on last packet
- Specified "relative" NTP uses "best" platform clock
- Formal ref to IPSEC; spec "codifies existing practice"
- Partial conversion to MUST, SHOULD, MAY
- Last paragraph of introduction deleted

Changes <u>not</u> made

- Ignore problem of group size dropping to zero in "reverse reconsideration"
- No scaling of min RTCP interval larger (could cause timeouts, and not that big a benefit)
- No change to jitter algorithm for multi-packet video frames
- Additional SDES items deferred to IANA registration (e.g.PHOTO URL, Nick-name, Organization)
- No change to definition of RR "loss fraction"
- Nothing about translators adding random offsets

Changes Still Needed

- IANA Considerations section
- □ Collect more constants into Sec. 11
- Complete MUST, SHOULD, MAY
- Make sure code in appendix is correct

Open Issues for RTP Spec

- Does new wording provide right motivation for sending RTCP?
- Should we have both conditional and unconditional reconsideration?
- Is new Section 6.3 clear and correct requirement level?
- Lucent patent filed on "binning" algorithm for SSRC sampling

Recent Changes to RTP Profile

- □ PureVoice (QCELP) added as PT=12
- New policy stated: No additional static payload types will be added.
- RED, MP1S, MP2P added to encoding table as "dyn" payload types (RED is no longer PT=77)
- RFC references updated

Changes Still Needed

- Better explain new policy for static payloads
- Allow default 5% RTCP bandwidth to be overridden, and define SDP BW modifiers to specify explicit RTCP sender and receiver BW (here or RTP?)
- May need changes for generic formats, but can we proceed without them?
- IANA considerations section
- Complete use of MUST, SHOULD, MAY

Encodings as MIME Subtypes

- ☐ MIME major type on m= (audio, video)
- What about audio+video types?
- Encoding (subtype) in a=rtpmap
- Registration via profile doc itself?
- What new info is bound to the name for RTP purposes? Just payload format?
- What to do for conflicts with existing types, such as PCMU=audio/basic?

New AVT Charter

- Old charter is way out of date:
 - » Last existing milestone is 1993
 - » Says only define experimental protocols
- New charter proposed
 - » Reflect current state of RTP
 - » Set milestones for remaining work
 - » Lay out expectations for future work

AVT Work Items [1]

- Revise RTP & profile for advancement to draft std.
 - » WG last call in November and submission to the IESG immediately after the December IETF.
 - » Register encoding names as MIME subtypes
- Complete the MIB -- "finished" by December
- Finish "guidelines for payload format authors" draft
 - » Can probably be ready for last call by December?
- Generic payload format?
 - » Proposals to be merged and submitted as draft
 - » Discussion in December, revised draft in February

AVT Work Items [2]

- Other payload formats
 - » BT656, H.263+, JPEG are done
 - » PureVoice audio: last call September 1998
 - » Generic FEC: split into parity FEC to last call soon and separate R-S different draft to last call after December meeting.
 - » DMIF/MPEG-4: tied in with the generic payload?
 - » X protocol streams: off topic?
- Multiplexing protocol
 - » Decide course of action at this meeting