Audio/Video Transport Extensions (AVTEXT)
Adminstrivia

- Notetakers?
- Jabber scribe?
- Jabber ChatRoom
  - Address: `xmpp:avtext@jabber.ietf.org`
  - Logs: `http://jabber.ietf.org/logs/avtext/`
Note well

• Any submission to the IETF intended by the Contributor for publication as all or part of an IETF Internet-Draft or RFC and any statement made within the context of an IETF activity is considered an "IETF Contribution". Such statements include oral statements in IETF sessions, as well as written and electronic communications made at any time or place, which are addressed to:
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  – Any IETF mailing list, including the IETF list itself, any working group or design team list, or any other list functioning under IETF auspices
  – Any IETF working group or portion thereof
  – Any Birds of a Feather (BOF) session
  – The IAB or any member thereof on behalf of the IAB
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Agenda

- Wednesday 13:00 - 14:30
- Chairs: Keith Drage Magnus Westerlund
- 13:00 Agenda bash and status check (15 min) Chairs
  - status of Duplicating RTP Streams
    - http://www.ietf.org/id/draft-ietf-avtext-rtp-duplication-00.txt
- 13:15 Content Splicing for RTP Sessions (15 min) Jinwei Xia
  - Update and IESG issues
- 13:30 RTP Media Stream Pause and Resume (10 min) Bo Burman
- 13:40 RTCP SDES Item SRCNAME to Label Individual Sources (15 min) Bo Burman
- 13:55 Support for Multiple Clock Rates in an RTP Session (10 min) Glen Zorn
  - Where are we?
- 14:05 (Wrap up)
Status

- **Done**: Submit Considerations for RAMS Scenarios for Informational
- **Done**: Submit RTP Header extension for mixer to client audio level indication as Proposed Standard
- **Done**: Submit RTP Header extension for client to mixer audio level indication as proposed standard
- **Dec 2011**: Submit Support for multiple clock rates in an RTP session for Proposed Standard
  - draft-ietf-avtext-multiple-clock-rates-06
- **Done**: Content splicing for RTP sessions as Informational
  - draft-ietf-avtext-splicing-for-rtp-11
- **Mar 2013**: Request Publication of specification for duplicating RTP Streams with intended status as Proposed Standard
  - draft-ietf-avtext-rtp-duplication-00
This document explains how RTP streams are to be duplicated without breaking RTP and RTCP rules

--Editors: Ali Begen and Colin Perkins

There are two drafts adopted in MMUSIC we needed

--Duplication Grouping Semantics in the Session Description Protocol
   --(draft-ietf-mmusic-duplication-grouping-00)
--Delayed Duplication Attribute in the Session Description Protocol
   --(draft-ietf-mmusic-delayed-duplication-00)

We also use the srcname from

--RTCP SDES Item SRCNAME to Label Individual Sources
   --(draft-westerlund-avtext-rtcp-sdes-srcname-02)

-- We need it to be adopted by AVTEXT, then we can complete our draft