# SIP, SDP, and RTP Interop Issues

**David Hiers** 

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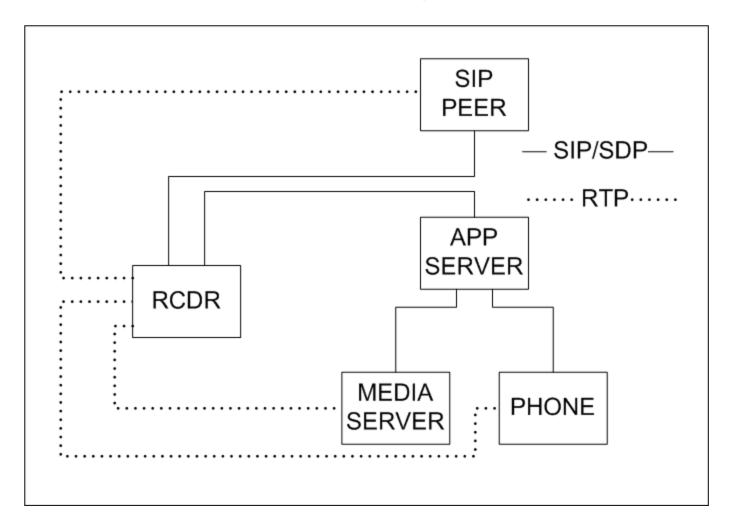
#### www.voiceops.org

- Email list for operators of voice networks
  - TDM suite
  - IP suite
  - Whatever comes next!
- Technical, operational, interop, etc.
- Free, non-commercial, no solicitation
- Started July 2009, over 750 members

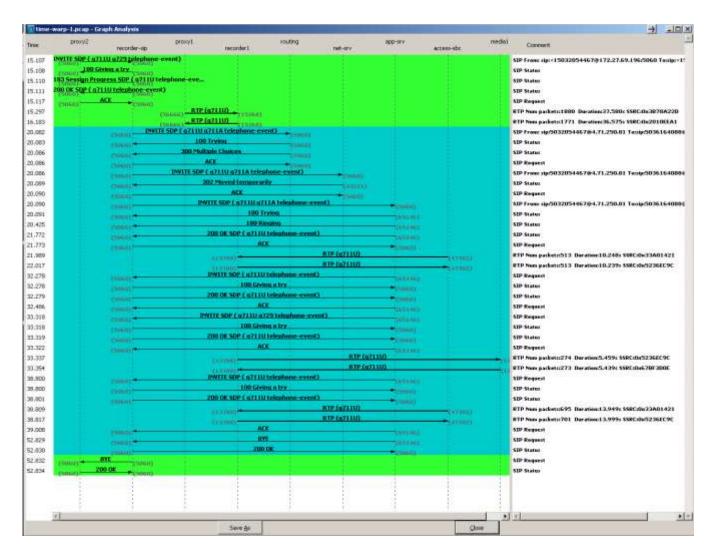
#### Case Studies

- PART 1
  - Time Warp
- PART 2
  - Past is Prologue
  - Conservation Hurts

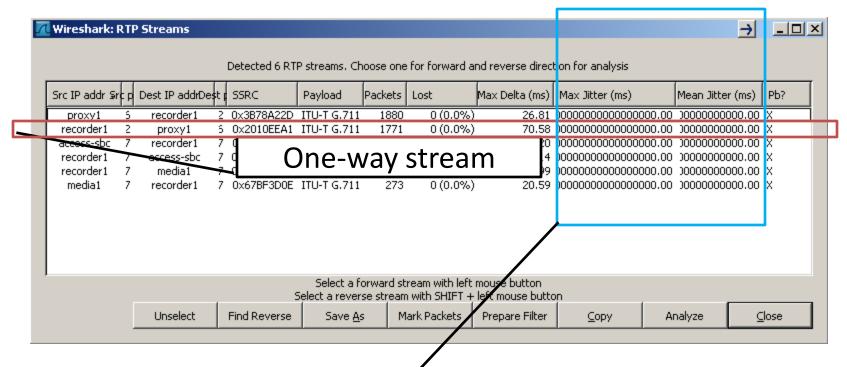
# Recorder Topology/Overview



#### SIP Ladder: OK

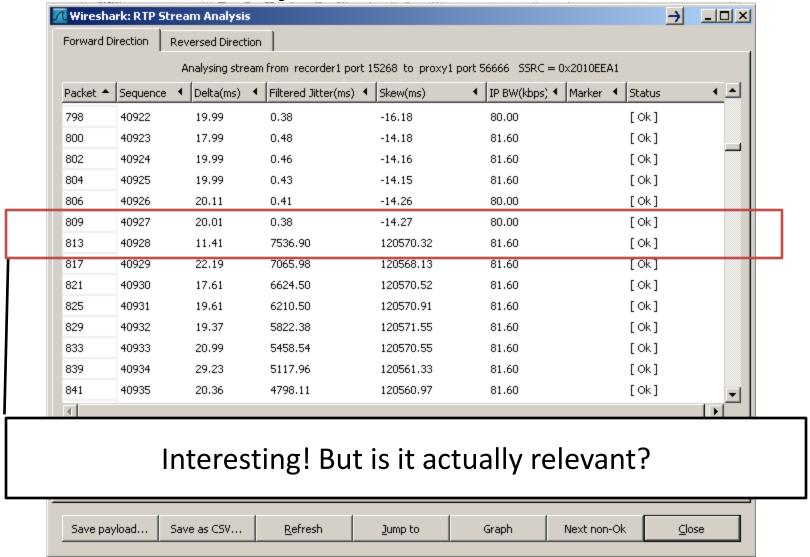


## RTP Streams: One-way stats OK



All zeros for jitter?
Sure, we've got a good network,
but this is not reasonable.
Bug in wireshark?

# Stream Analysis: Potential trouble?



Case 1: Time Warp

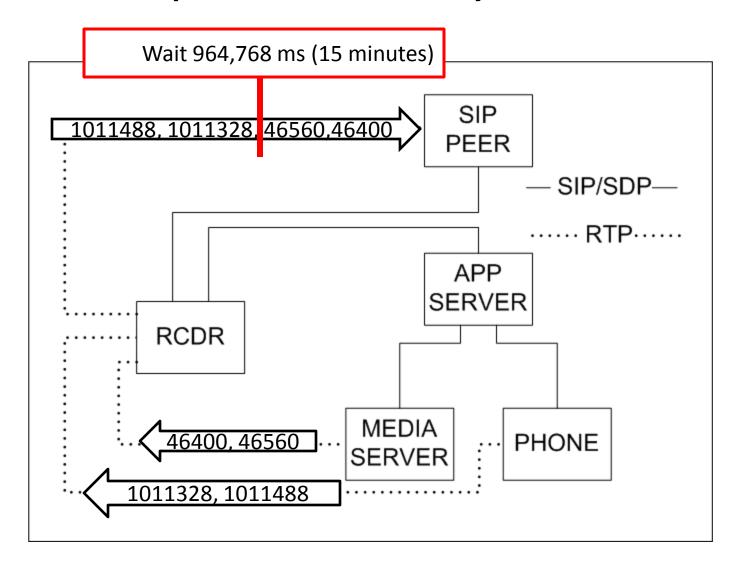
## Packets show potential problem

```
Observed, not calculated: trustworthy
                                                     DEEA1, Seq=40922, Time=45760
                                                            Seg=40923, Time=45920
                      PT=ITU-T G./II PCMU, SSKC=UXZUIDEEA1.
802 recorder1
                      PT=ITU-T G.711 PCMU, SSRC=0x2010EEA1
                                                            Seg=40924, Time=46080
              proxy1
                      PT=ITU-T G.711 PCMU, SSRC=0x2010EEA1, Seq=40925, Time=46240
804 recorder1
              proxy1
                                                            Seg=40926 Time=46400
806 recorder1
                                           SSRC=0x2010FFA1
809 recorder1
                                                            Seg=40927, Time=46560
                      PT=ITU-T G.711 PCMU, SSRC=0x2010EEA1,
              proxyl
813 recorder1
                                    PCMU, SSRC=0x2010EEA1
                                                            Seg=40928, Time=1011328
              proxy1
                                                            Seg=40929, Time=1011488
817 recorder1
              proxy1
                      PT=ITU-T G.711 PCMU, SSRC=0x2010EEA1,
\821 recorder1
                      PT=ITU-T G.711 PCMU, SSRC=0x2010EEA1,
                                                            Seq=40930, Time=1011648
              proxy1
825 recorder1
                                                            Seg=40931, Time=1011808
                      PT=ITU-T G.711 PCMU, SSRC=0x2010EEA1,
              proxy1
829 recorder1
                      PT=ITU-T G.711 PCMU, SSRC=0x2010EEA1,
                                                            Seq=40932, Time=1011968
              proxy1
                                                            Seq=40933, Time=1012128
833\recorder1
                      PT=ITU-T G.711 PCMU, SSRC=0x2010EEA1
              proxy1
                                                                  34, Time-1012288
839
         Start of one-way audio: what caused it?
            Strong association with TIMESTAMP
                                                                     Case 1: Time Warp
```

## **Timestamp Definition**

- RFC 3550 RTP
- 5.1 RTP Fixed Header Fields
  - The timestamp reflects the sampling instant of the first octet in the RTP data packet. The sampling instant MUST be derived from a clock that increments monotonically and linearly in time...
  - The initial value of the timestamp SHOULD be random...

#### Timestamp Discontinuity Breaks Audio



## **Analysis and Resolution**

- Recorder failed to implement RFC
- Wireshark player failed to enforce RFC
- Peer's old GWs failed to enforce RFC
- Peer started rolling upgrade of GWs
- Peer's new GWs enforced RFC, esp TIMESTAMP
- Calls that connected to new GWs failed while old GWs continued to work
- Calls with small jumps continued to work
- Resolution: Recorder normalized TIMESTAMP

#### Summary

- Mutually supporting defects exist
- Behavioral tests are not completely trustworthy
- Must ensure that things work for the right reasons