

Time and Timing Issues for IP Based Multimedia

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Outline

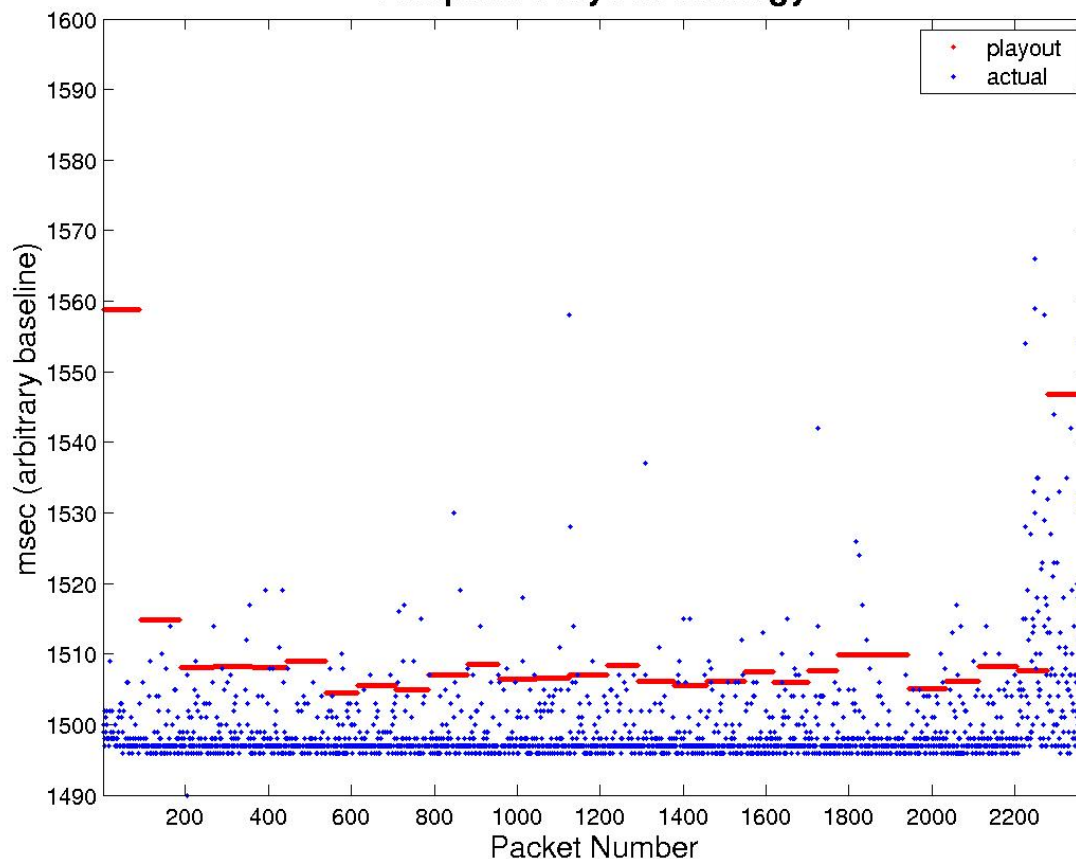
- Time & Timing within IP Multimedia Applications
 - Voice & Video over IP
 - Streaming
 - Media Mixer
 - Gaming
- Challenges of Wireless Ubiquity
 - Wireless speaker scenario
 - Delivering NTP over wireless
- Conclusions

Background Research VoIP

- VoIP : Adaptive Jitter Buffer Algorithms
 - Jitter buffer absorbs packet arrival jitter
 - Adaptive buffer tracks network
 - Implemented via silence period adjustment
 - QoS impact unknown
 - No Time Synch between endpoints
 - No knowledge of actual M2E delay
 - Adjustments often unnecessary in context of acceptable M2E delay

Adaptive Jitter Buffering

Adaptive Playout Strategy

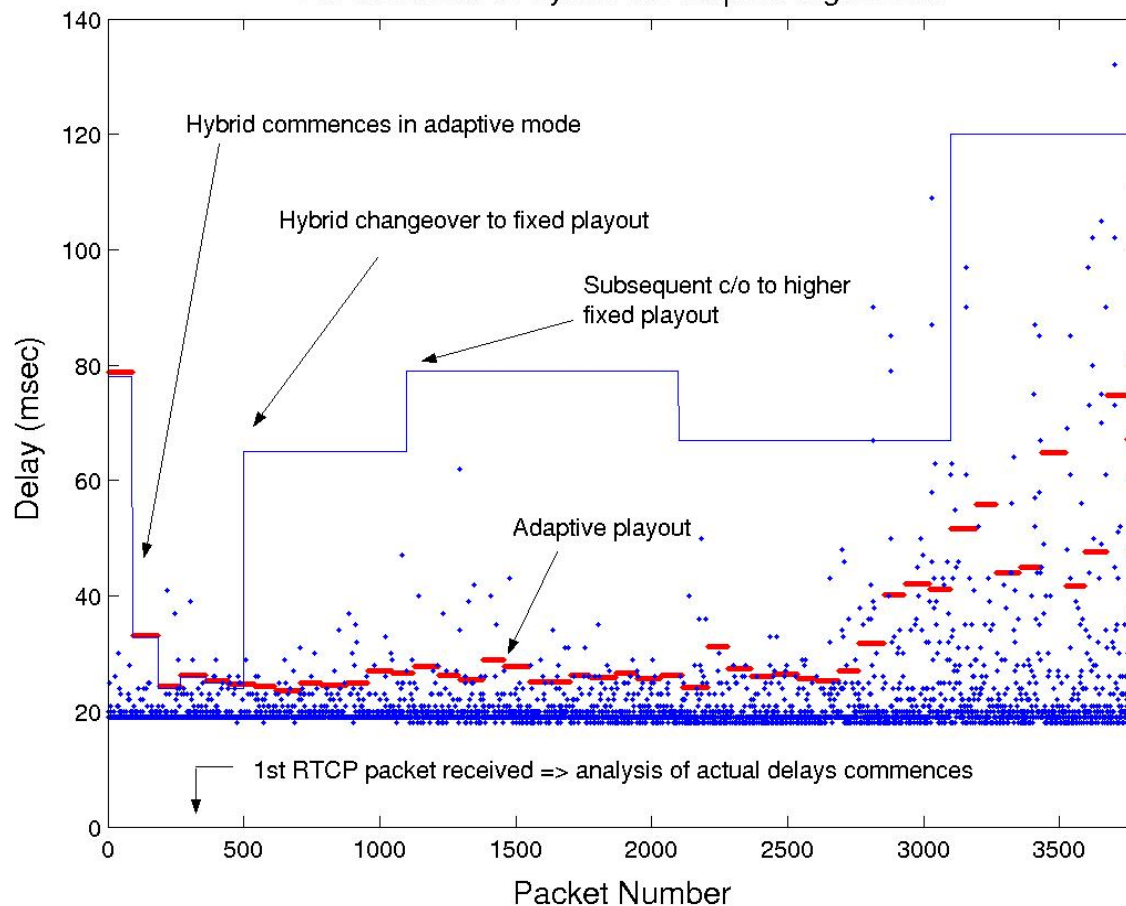


Hybrid Playout Algorithm

- Hybrid Algorithm
 - Based on synchronised time across different hosts
 - Minimise late loss at expense of increased delay
 - ITU-T E-Model
 - Net gain in user-satisfaction R-factor
 - Added (unquantified) benefit of reduced silence period distortion
 - Synchronised time provided by NTP
 - Precise delay info facilitated by RTCP SR packets

Hybrid Alg.: Synchronised Time

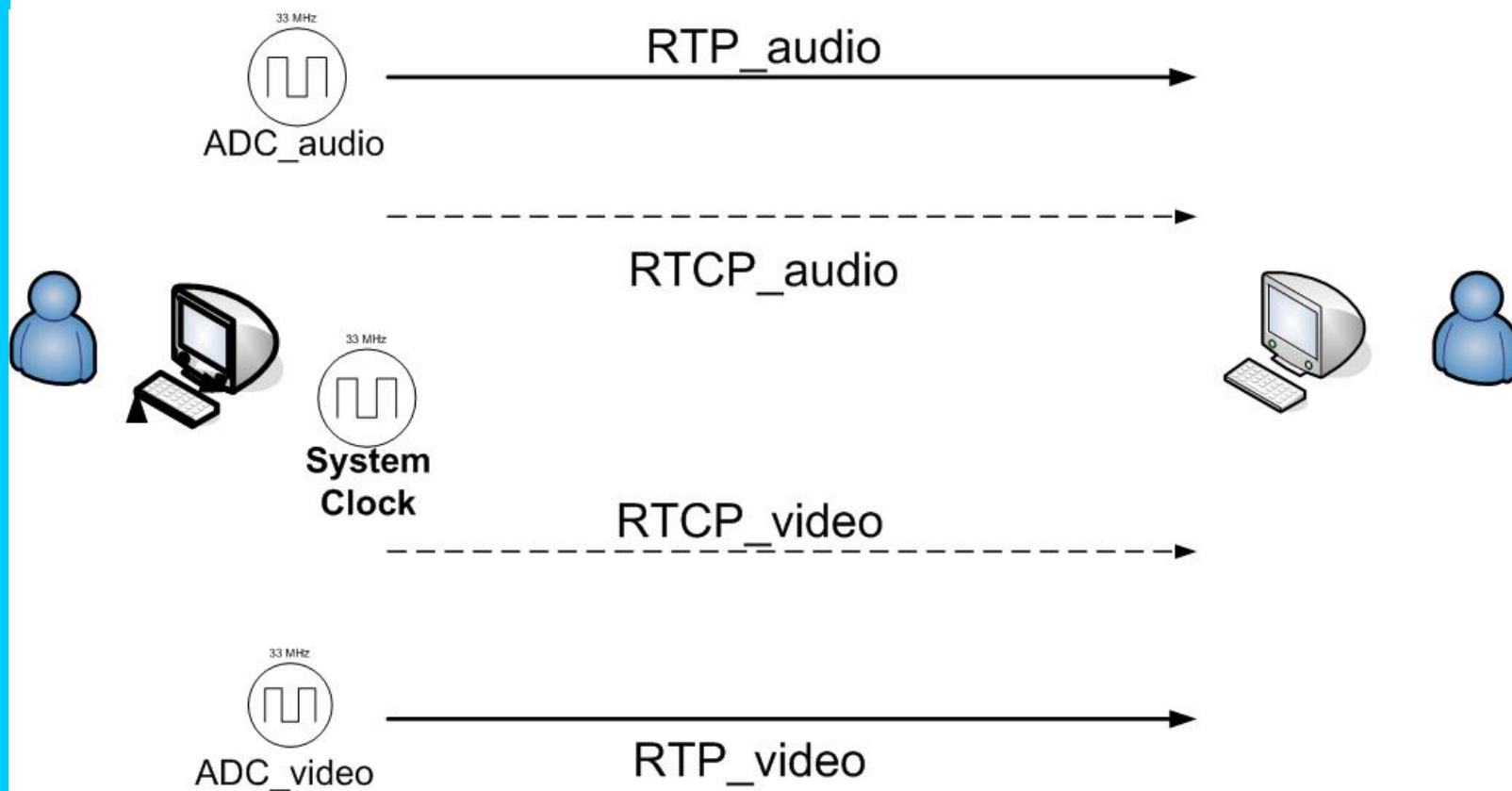
Performance of Hybrid vs Adaptive algorithms



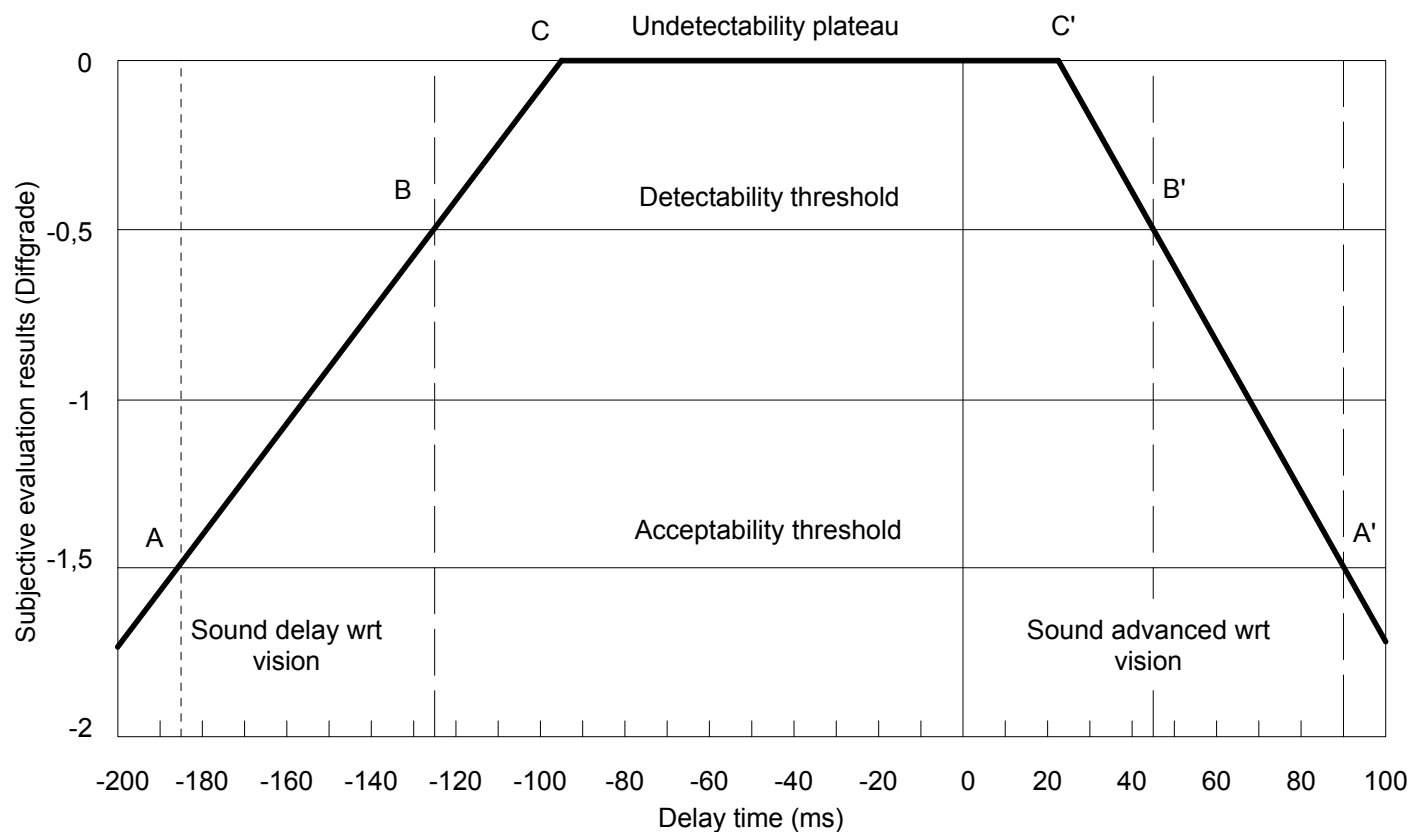
RTP & RTCP

- RTP useful for intra-stream synchronisation (reconstruction)
- RTCP SR useful for inter-stream synch for different streams eg lip-synch *from same host*
 - *Synch Time across hosts not reqd*
- *What if we want to synch different media streams from different hosts ?*
 - *Eg Media Mixer scenario*
 - *Require Synchronised Time **across** hosts*

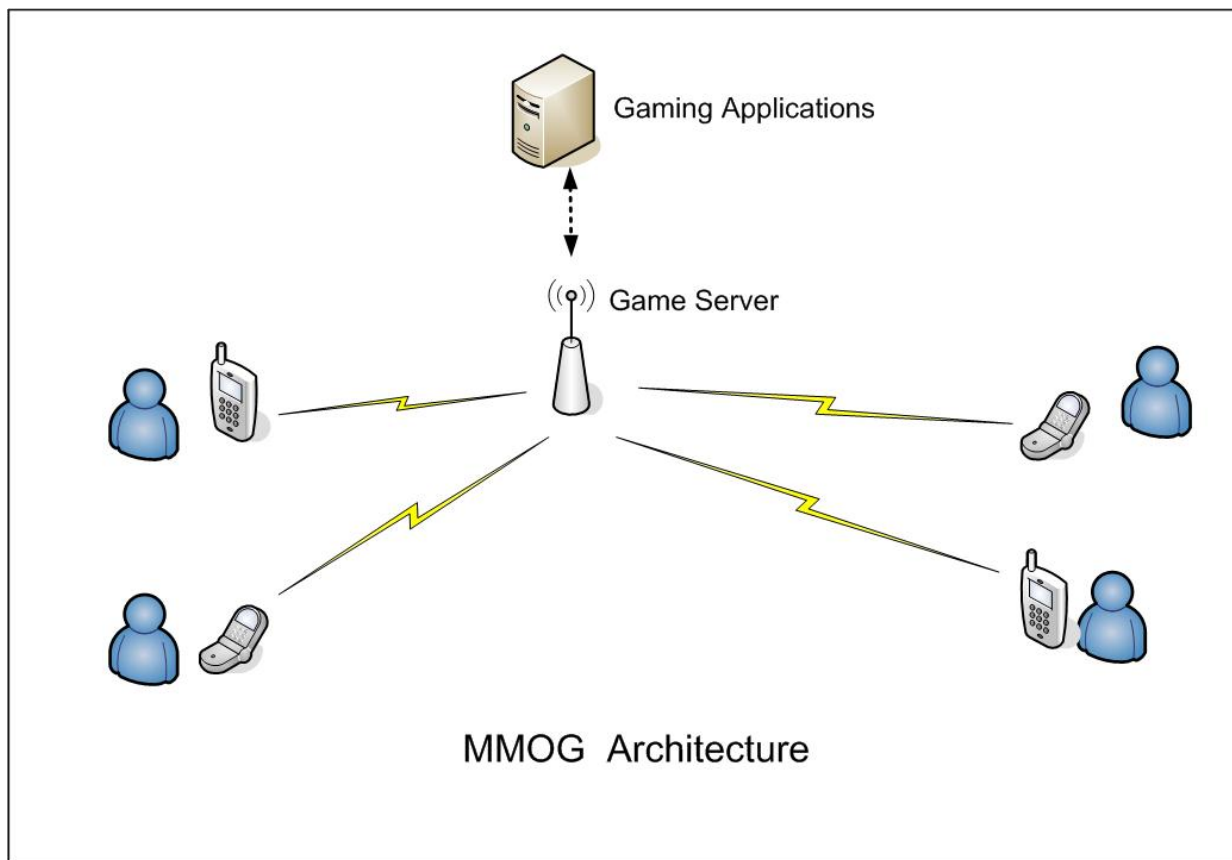
Lip Synch via RTCP SR



Detectability and Acceptability Thresholds for lip synch ETSI STQ



Synch Time for MMOG



Benefits of Synchronised Time

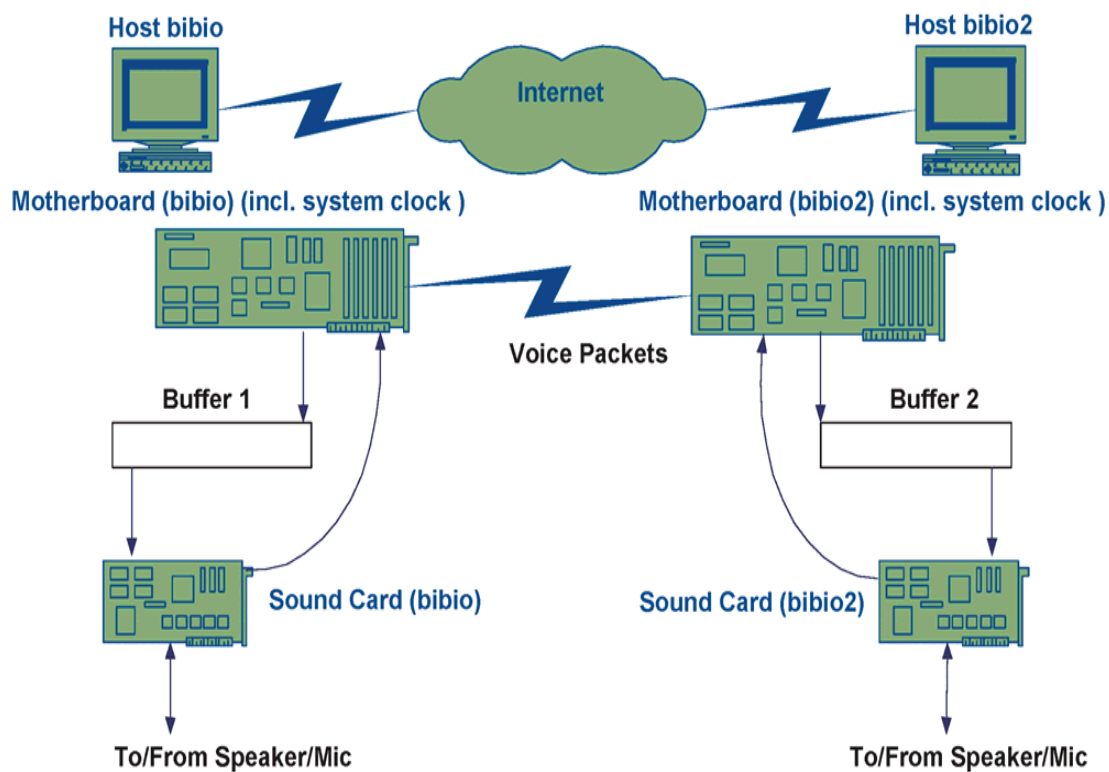
- VoIP / Video over IP
 - Precise Delay Information
 - Improved QoS (E-Model analysis)
- Media Mixer
 - Alignment of media across different hosts
- Gaming
 - MMOG market growth
 - Facilitates equalisation of delays where useful
 - *Levels the playing pitch for all participants*

Synchronised Timing

- NTP synchronises *system* clocks
- *Media* clocks are often separate subsystem
- Multiplicity of clocks introduces complexity
- VoIP Terminals / Gateways
 - Delay and Packet Loss
- Gaming
 - Delay & Packet Loss
- Media Mixer
 - Cumulative misalignment (if initially aligned)
- Streaming
 - Pkt loss rather than delay key issue

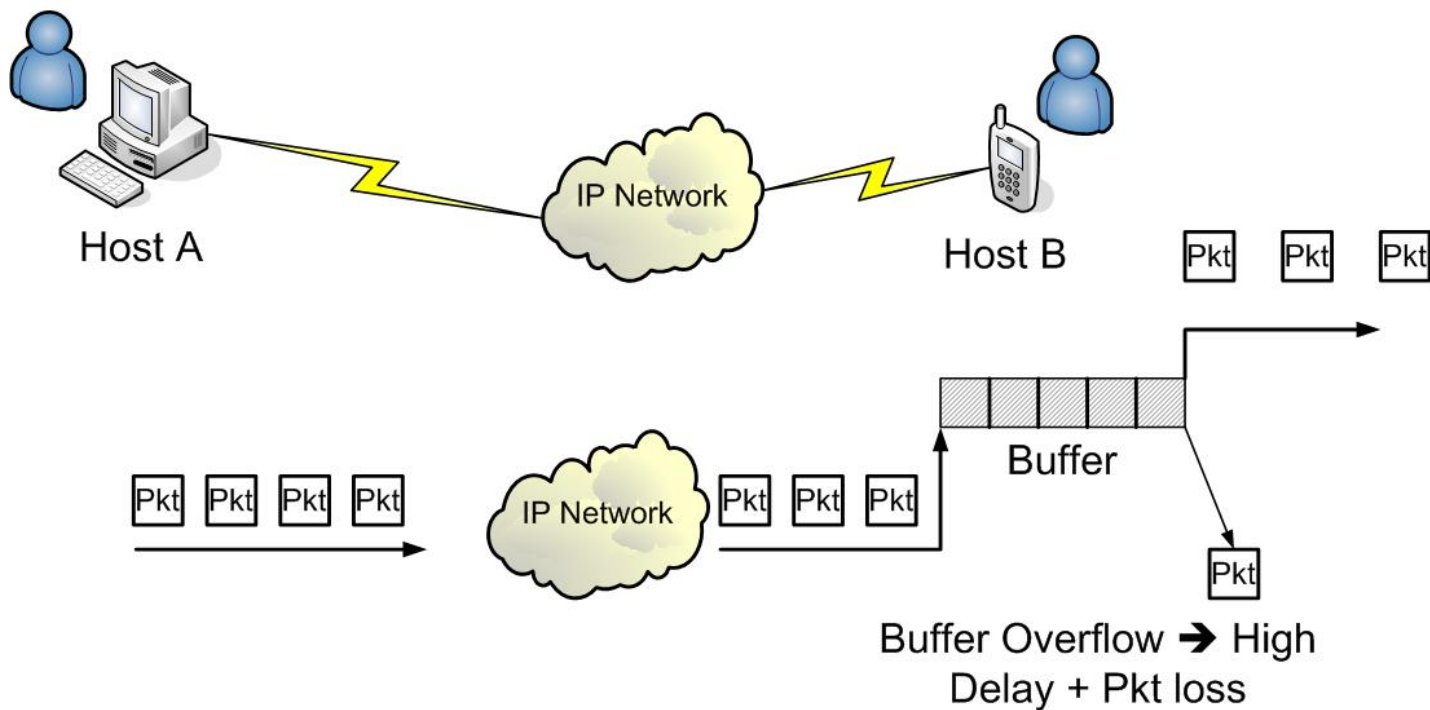
System v Media Clocks

System vs Audio Card Clock Issues



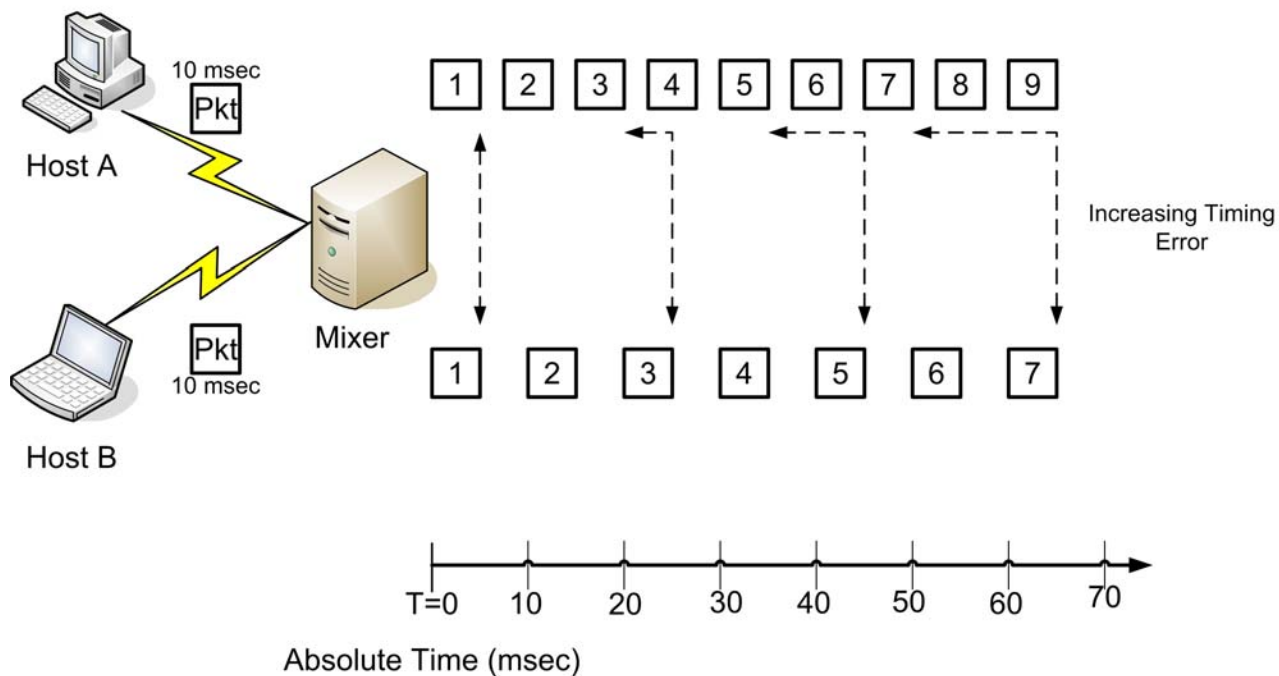
Timing 'Skew' for IP Multimedia

IP-IP Session



Media Clock Skew : Mixer

Mixer Timing Error due to Audio Clock Skew





Skew Solutions in wired world

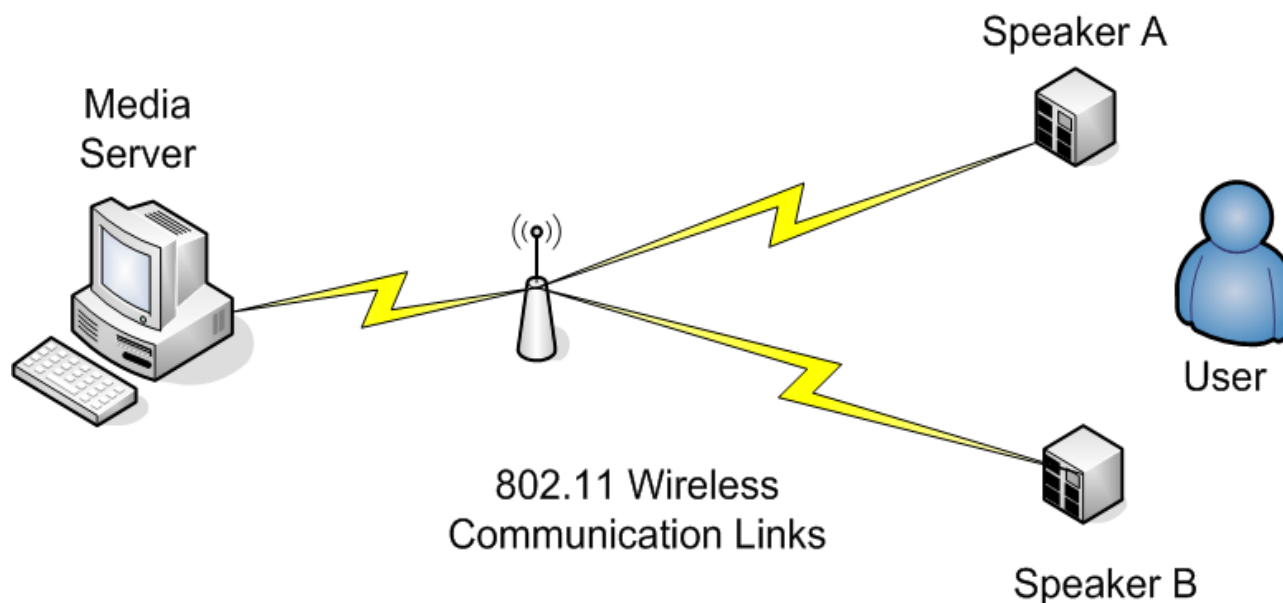
- Various skew detection and compensation mechanisms
 - Low level buffer monitoring to determine skew
- Patented NTP/RTCP approach
 - Based on Synchronised time
 - Independent benefits

Wireless Challenges

- Increased delay/jitter due to
 - Contention level & Signal degradation/interference
- Additional problems for
 - VoIP .. buffer/delay management
 - Gaming .. buffer/delay management /delay equalisation
 - Mixers ... buffer/delay mgt & stream alignment
 - Streaming
 - Buffer management
 - Stream Alignment for multiple parallel streams
 - Wireless Speakers

Wireless Speakers

Synchronisation of Media Streams



Wireless Streaming

- Playout from speakers A & B needs to be very tightly synchronised
- Approx impact of delay difference (Haas Effect)
 - $\Delta T < 30$ msec
 - Sensory inhibition
 - Only hear the 1st
 - Brain processes ΔT to determine sound source
 - $\Delta T > 30$ msec
 - Two distinct sounds heard .. Echo
 - Both undesirable!

Cause of ΔT

- Delays to speaker A and B may be very different due to 802.11 characteristics
 - Need to synchronise streams at start
 - → Synch Time reqd
- Skew between media clocks A & B will cause cumulative misalignment over time
 - 100 ppm = 60 msec over 10 minutes
 - → Synch Timing reqd

NTP in Wireless Environment

- Presents greater challenges
 - NTP operation is based on symmetric networks
 - Wireless networks can be very asymmetric
 - 802.11 'Reliability' will lead to NTP offset errors
 - Server & Path diversity in NTP design
 - Helps identify/eliminate servers on asymm links
 - 802.11 provides common weak link
 - What about 'new' 802.11 e
 - QoS over 802.11

Wireless QoS via 802.11e

- Default Best Effort ping
 - 13 packets transmitted, 13 packets received, 0% packet loss round-trip min/avg/max/stddev = 2.485/16.492/31.758/11.551 ms
- QoS Enabled ping
 - 12 packets transmitted, 12 received, 0% packet loss, time rtt min/avg/max/mdev = 2.458/3.705/6.478/1.119 ms
 - Cf <http://forums.star-os.com/showthread.php?t=6974>
- Much lower delay and jitter..but
- What happens when QoS channel is abused?

Conclusion

- Strong awareness of benefits of synch timing
- Less awareness of benefits of synch time
- Does the user currently care ?
 - Wired IP world :
 - QoS seen as acceptable ?
 - Other more pressing problems in network and terminals
 - Wireless IP world
 - Benefits of synch time more significant
 - Wireless speaker: time & timing critical
 - May raise awareness of benefits?



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Thanks to NPL...

Questions?