

Non-Intrusive Monitoring of VoIP Call Quality

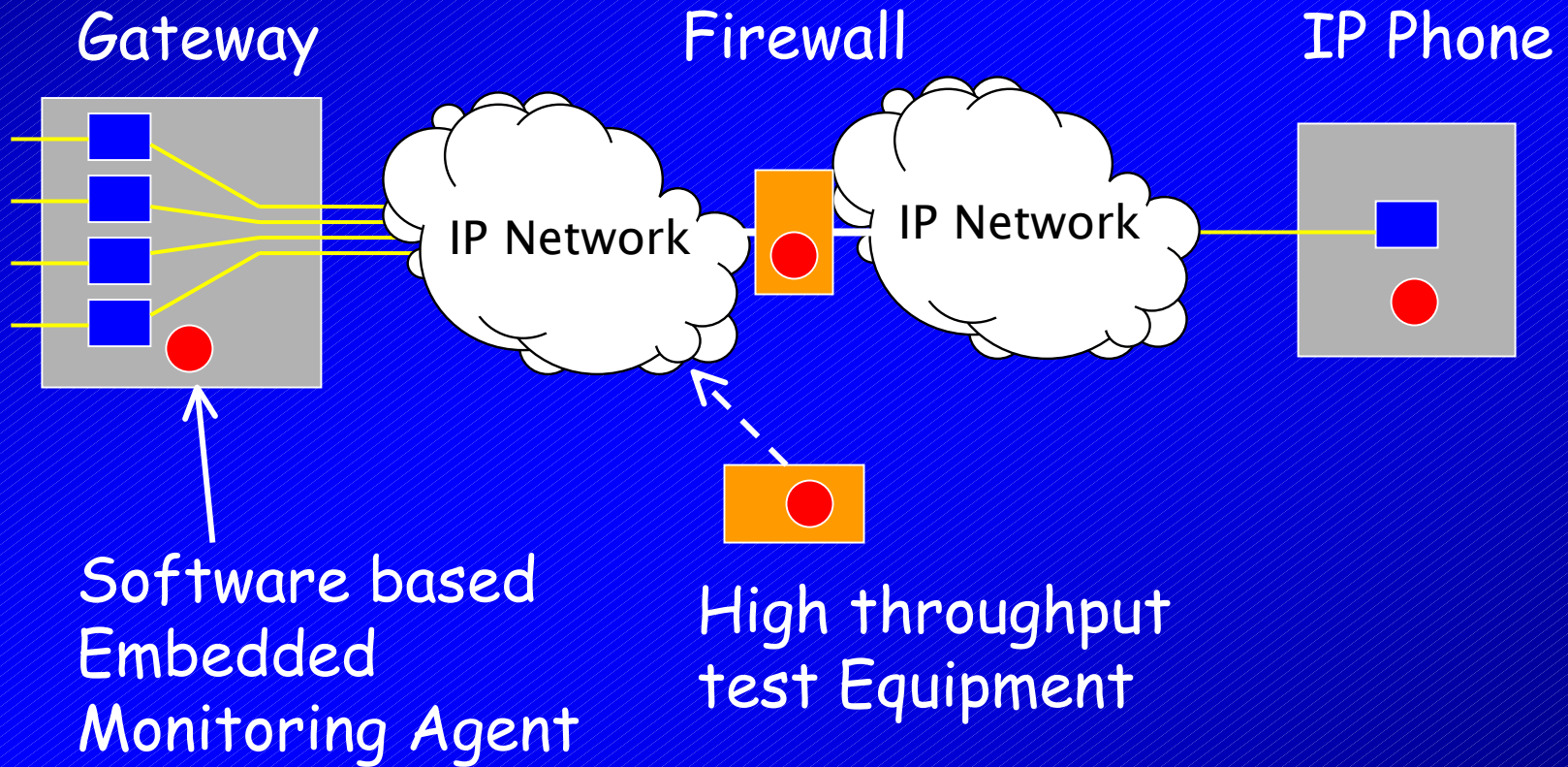
Alan Clark
Telchemy Incorporated

*Web: www.telchemy.com
Email: alan@telchemy.com*

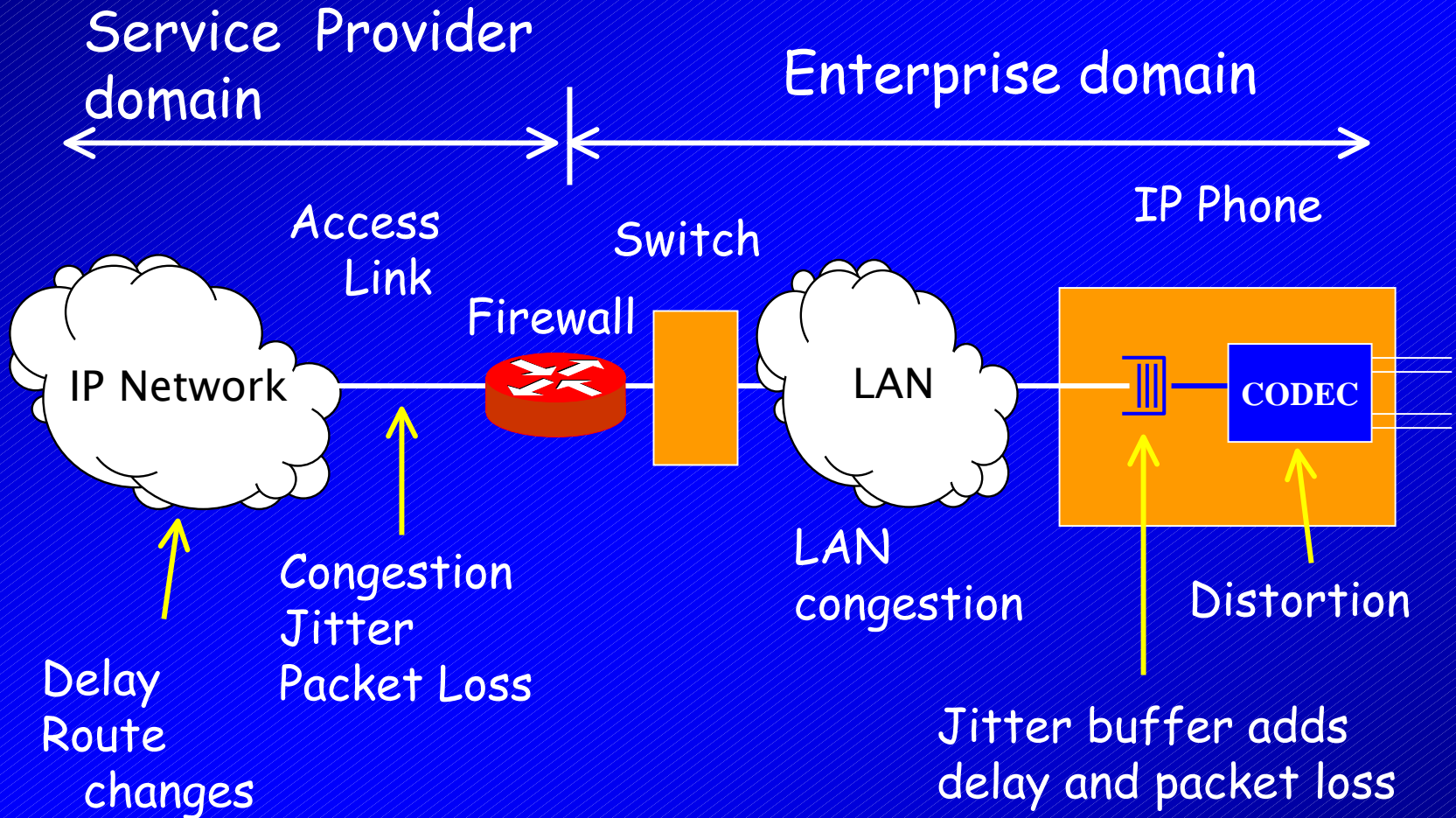
Problem Addressed

- Quality monitoring for
 - Service management and billing
 - Inter-domain SLA monitoring
 - Locating and identifying problems affecting service quality

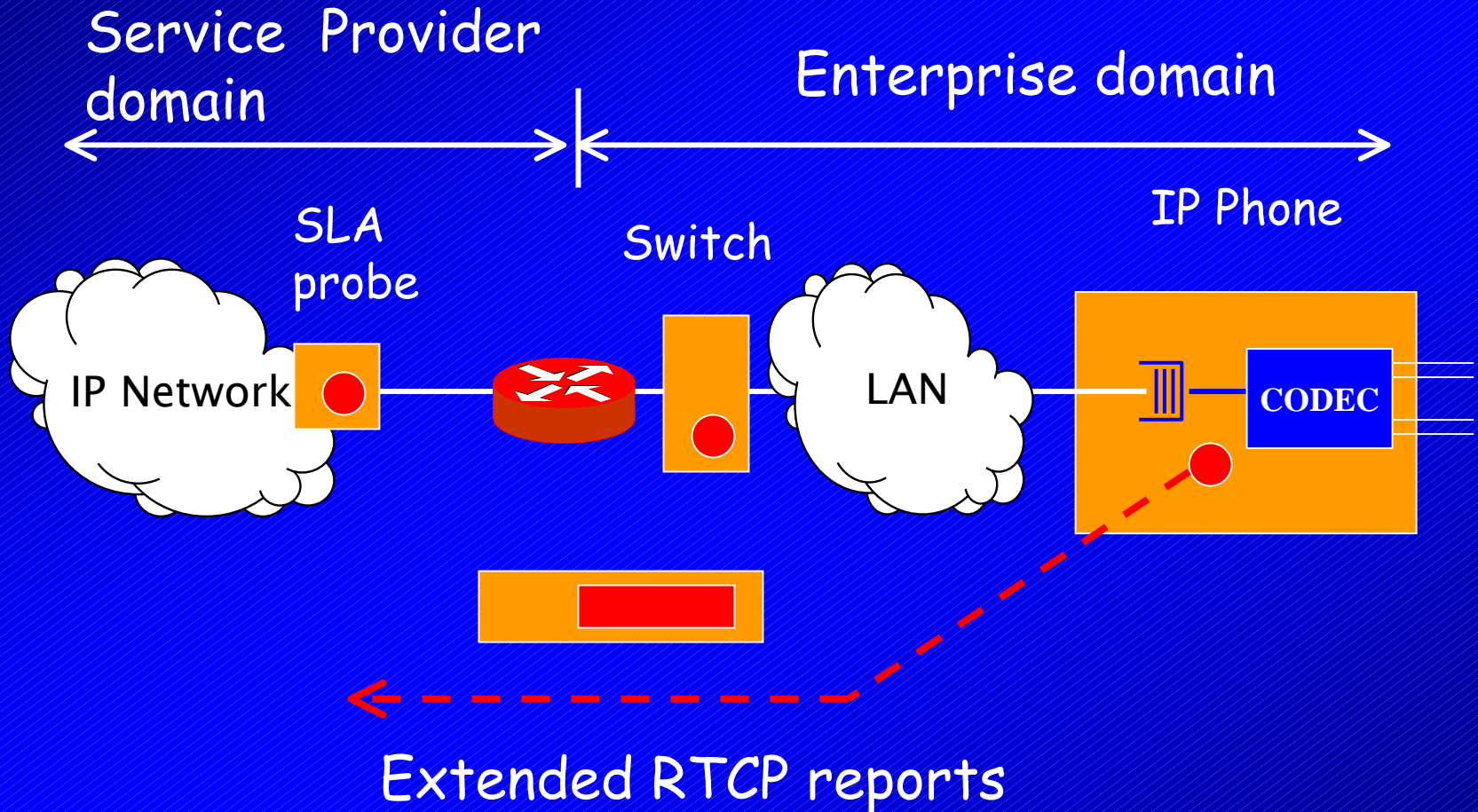
Approach



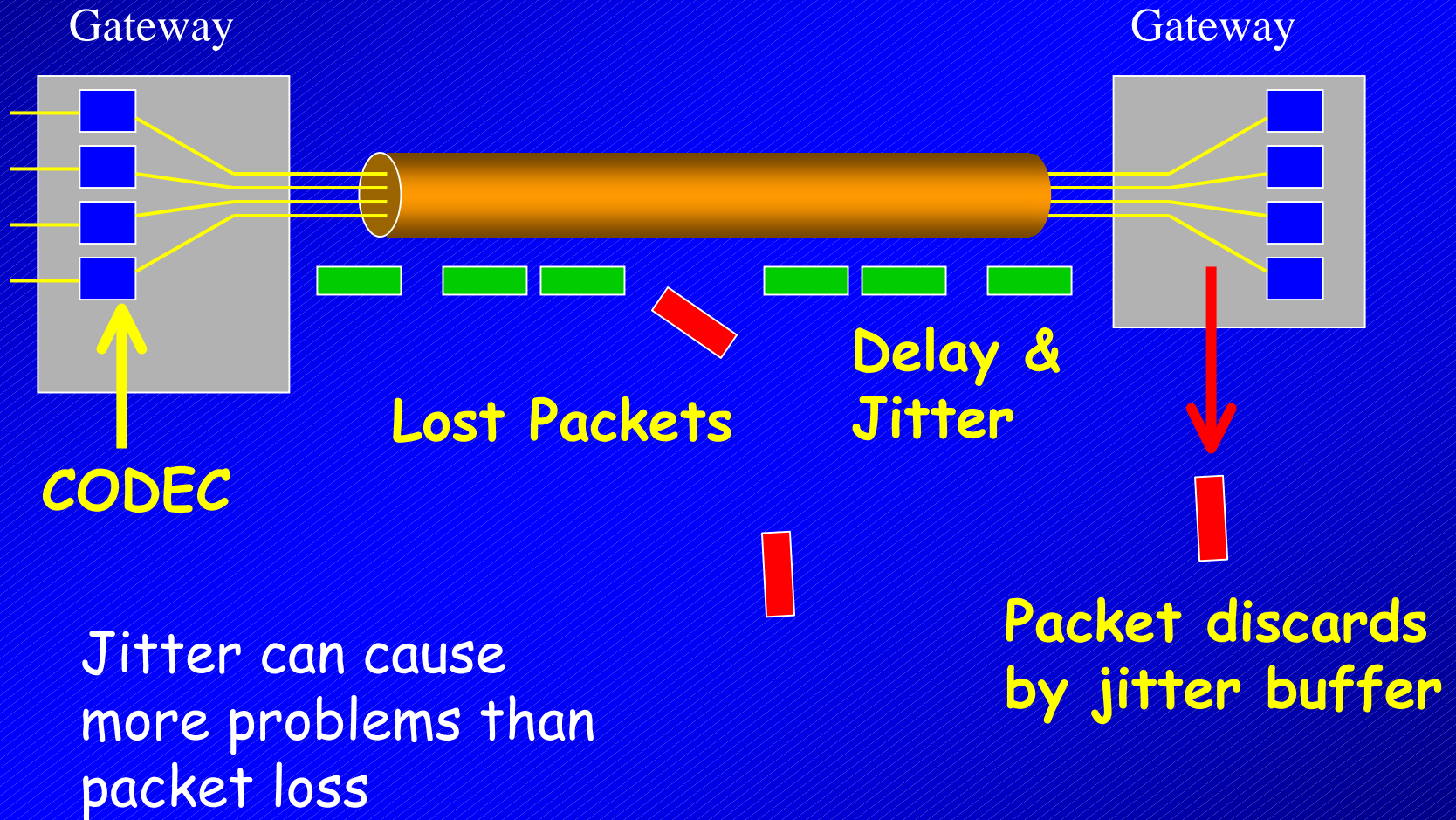
Example - IP Centrex



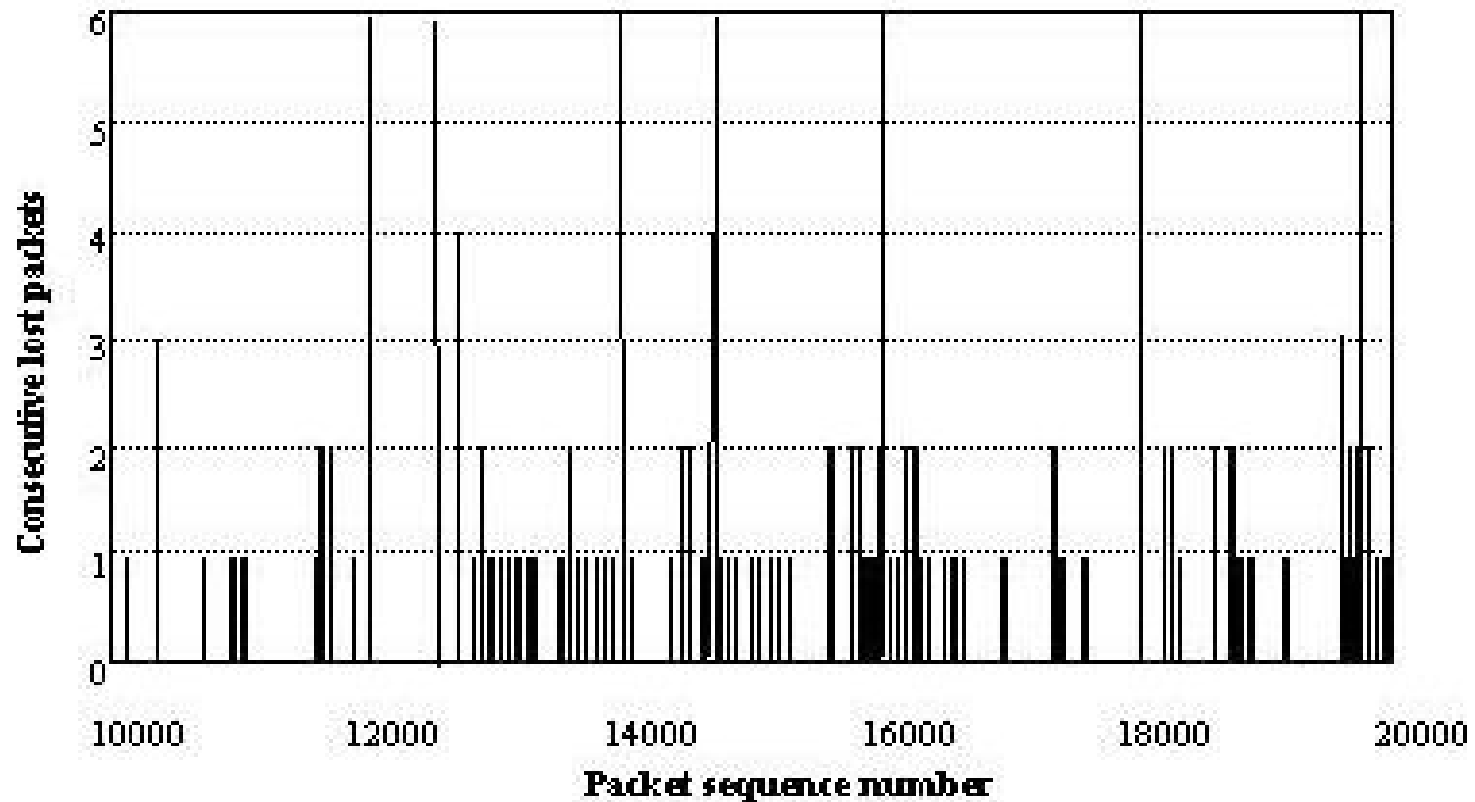
Example - IP Centrex



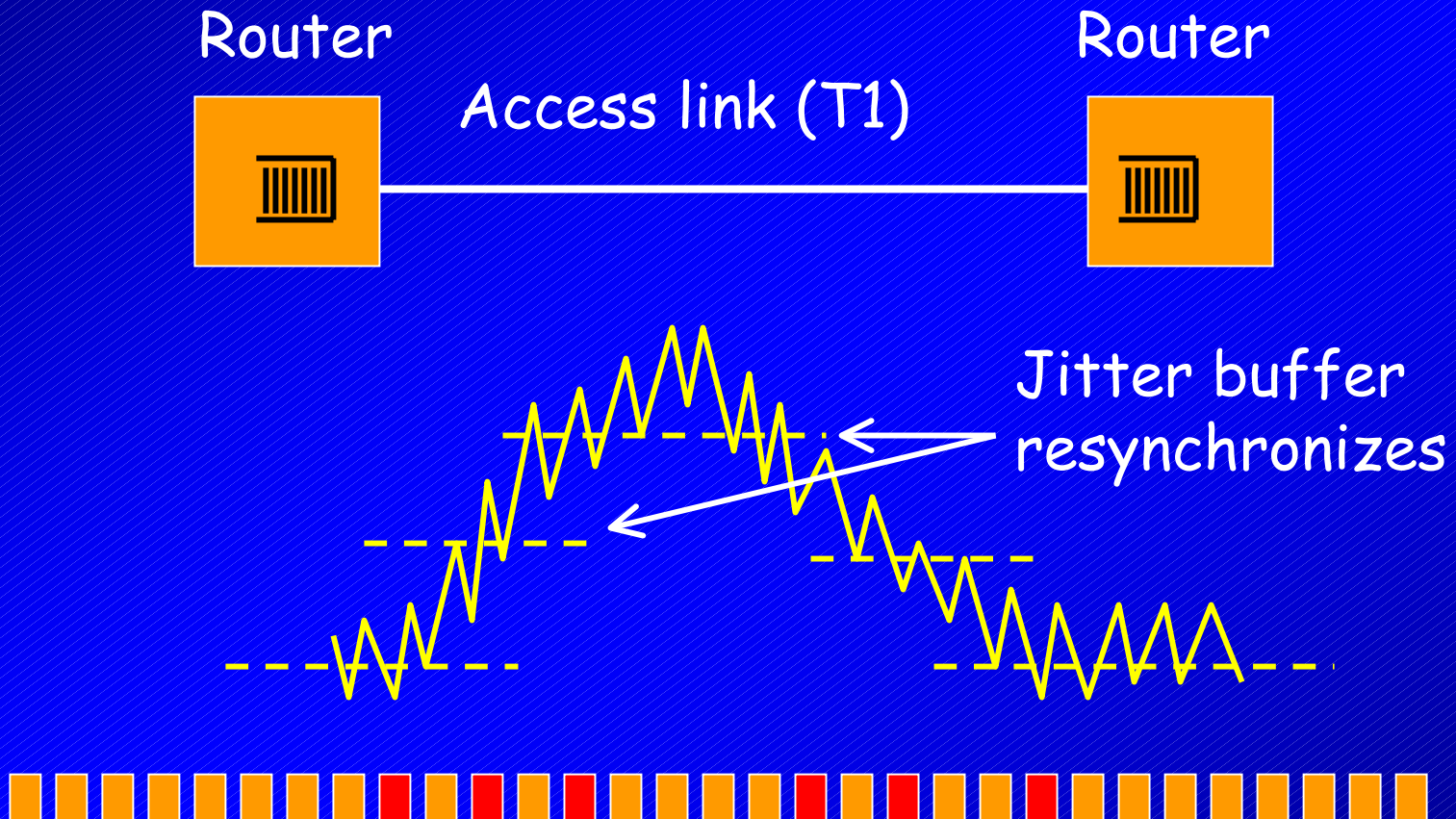
Factors impacting quality



Packet Loss is *Bursty*



Jitter causes bursty "discards"



Effects of Burstiness



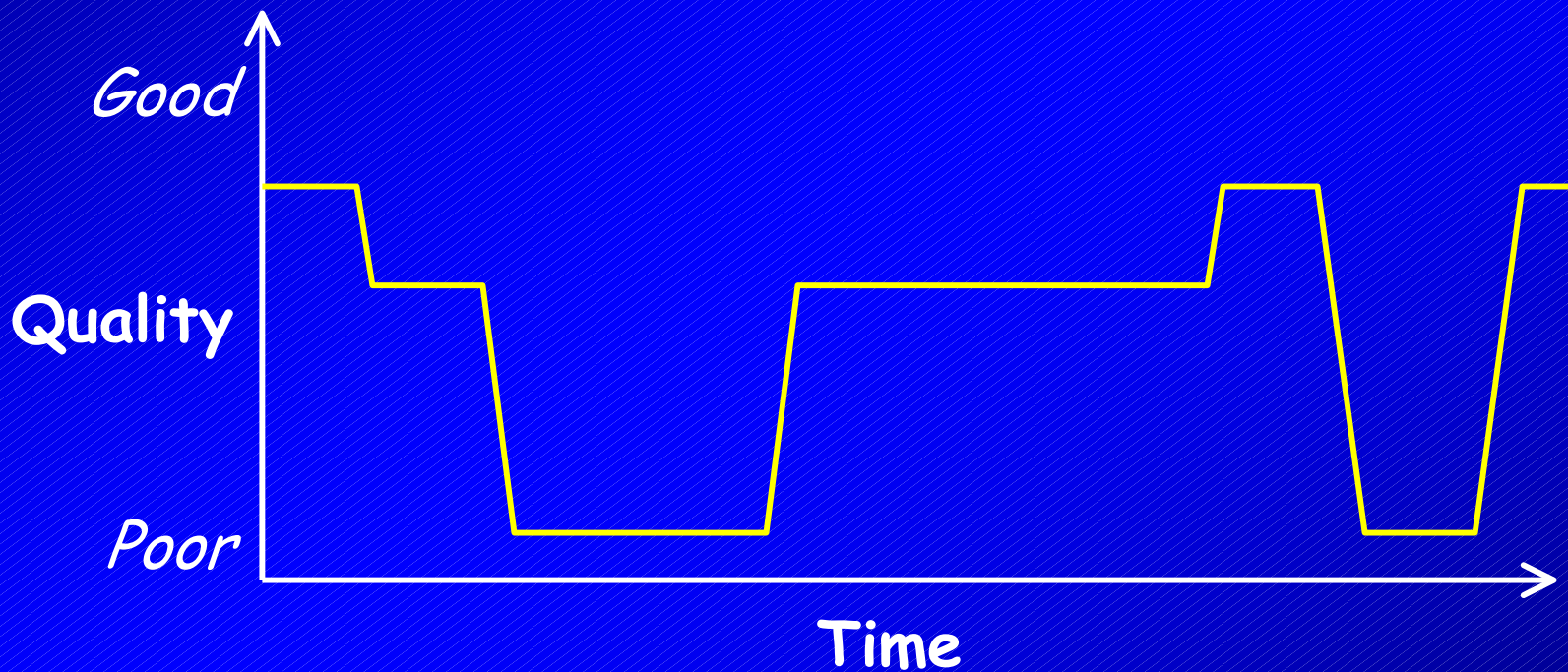
Packet loss concealment is effective for isolated lost packets but can't hide periods of high packet loss

1% random loss may be ok but

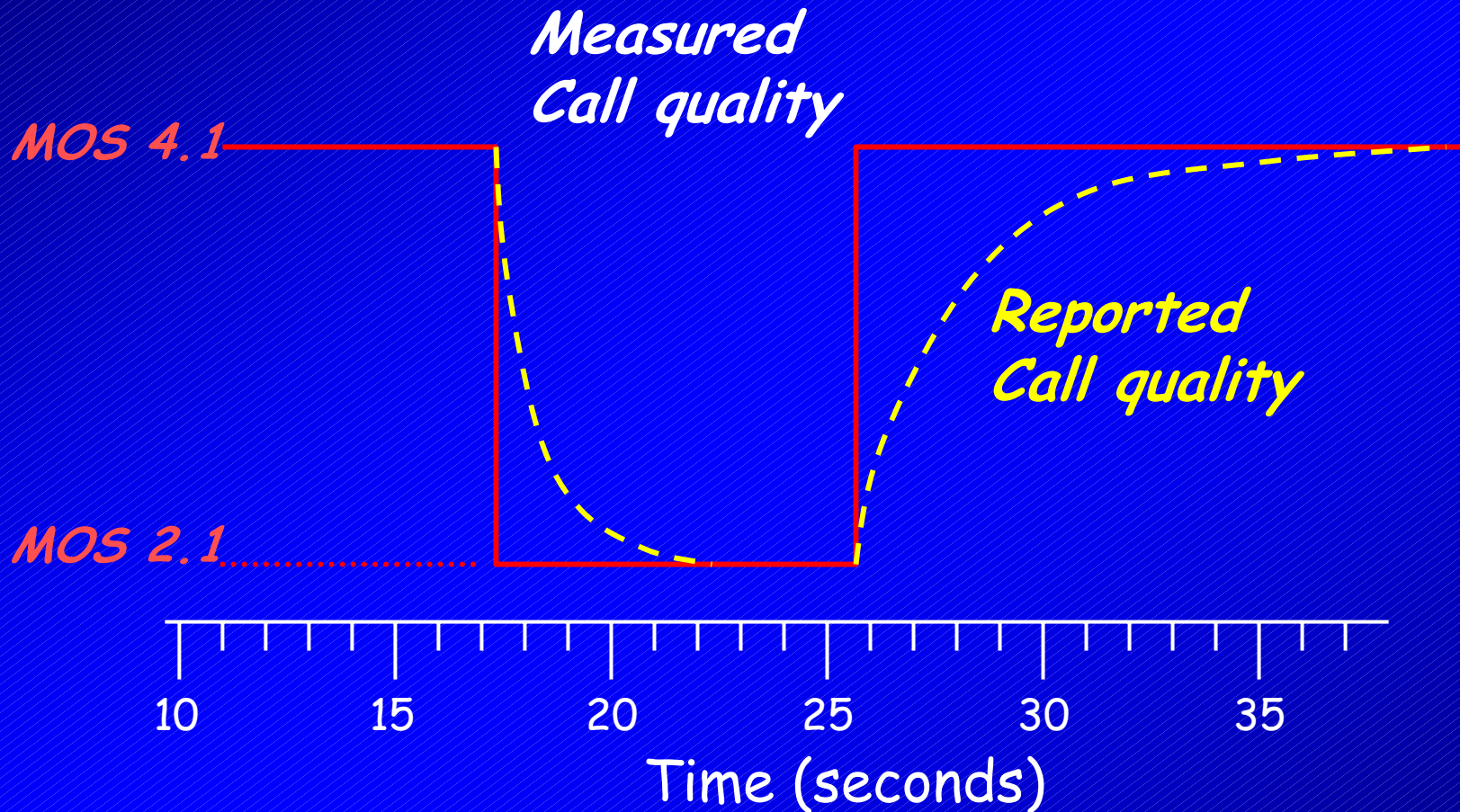
1% bursty loss may not

Result - time varying quality

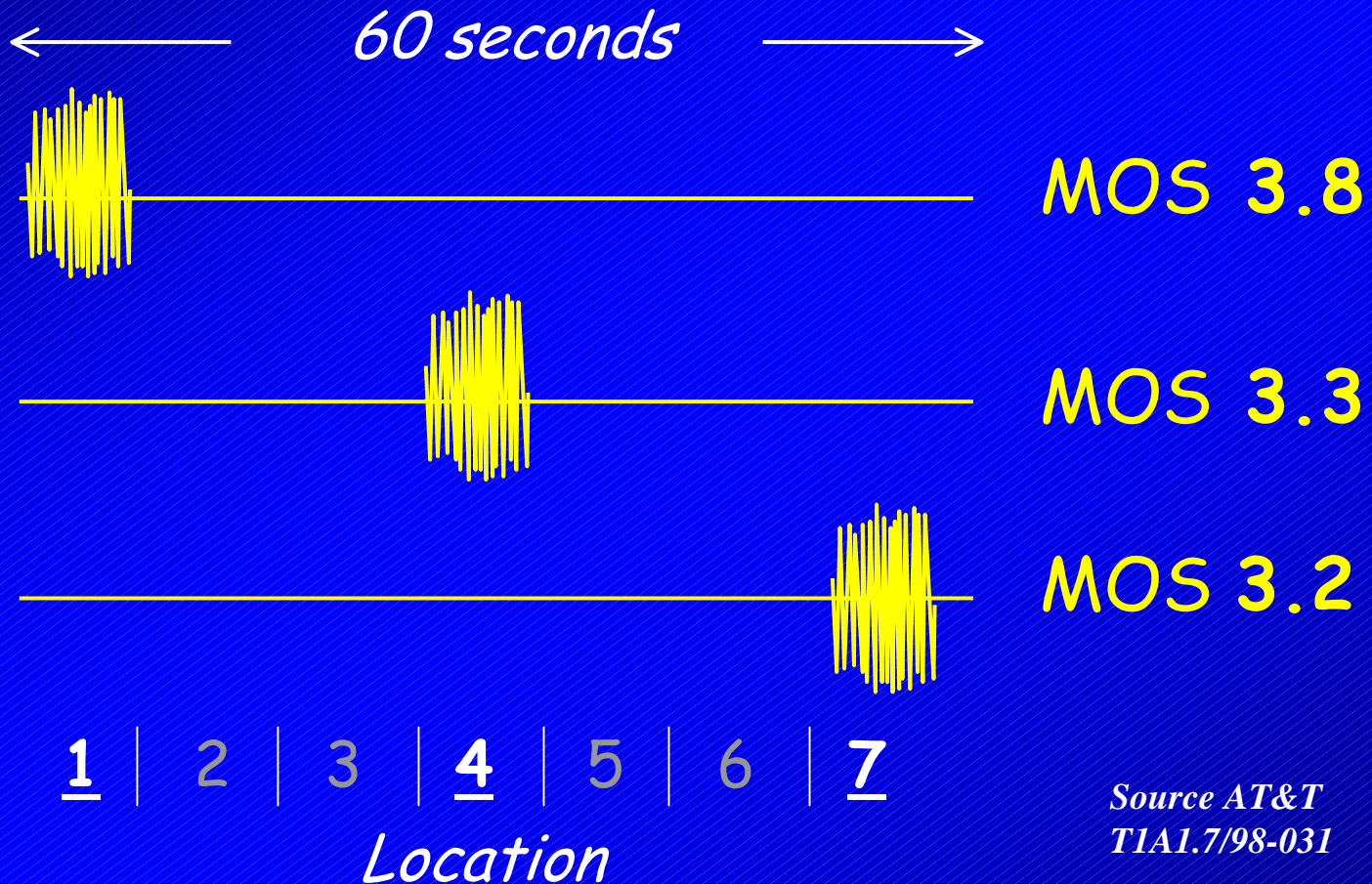
Problem - how to rate the call?



Measured vs Reported Quality

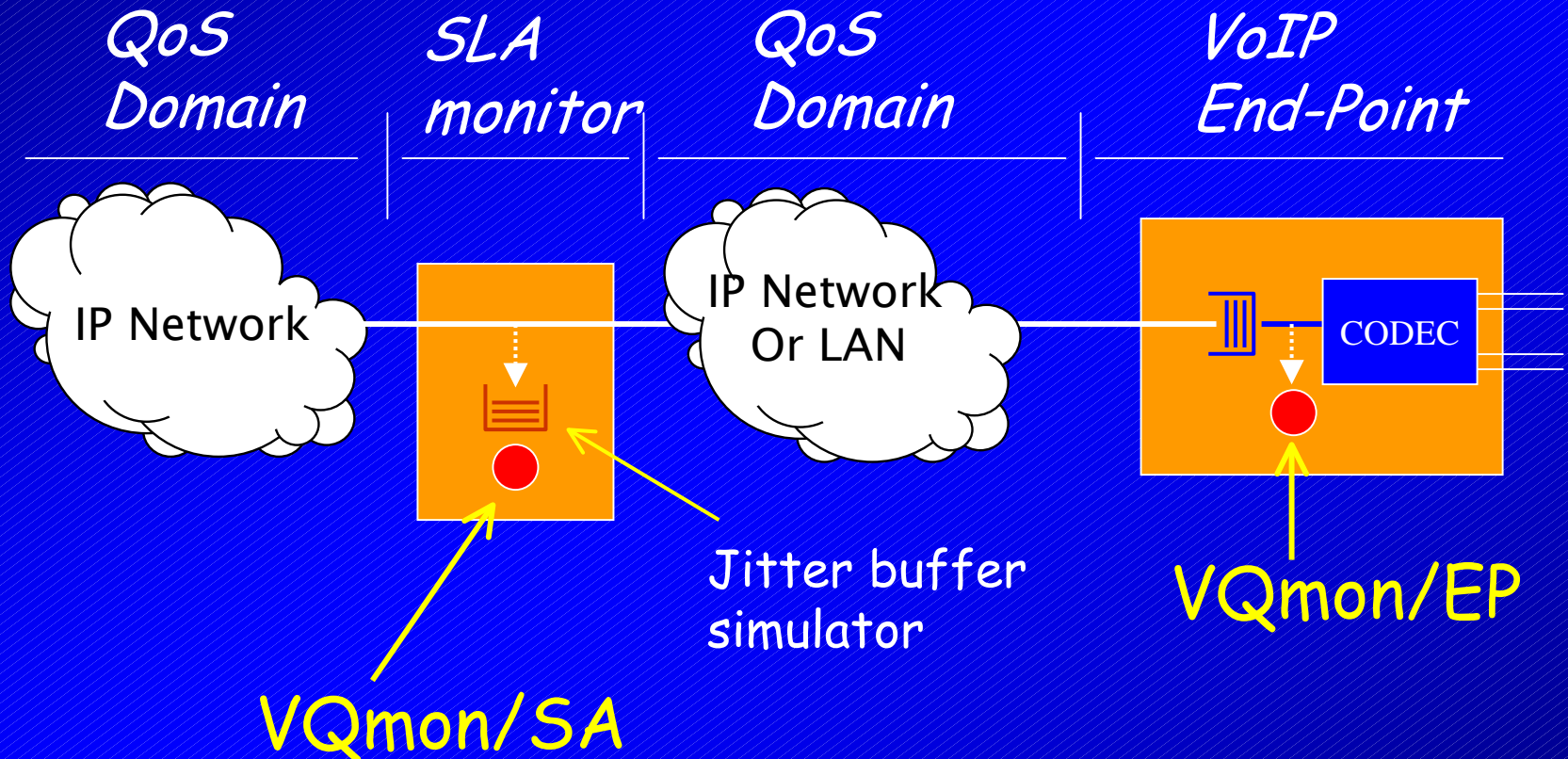


"Recency" effect

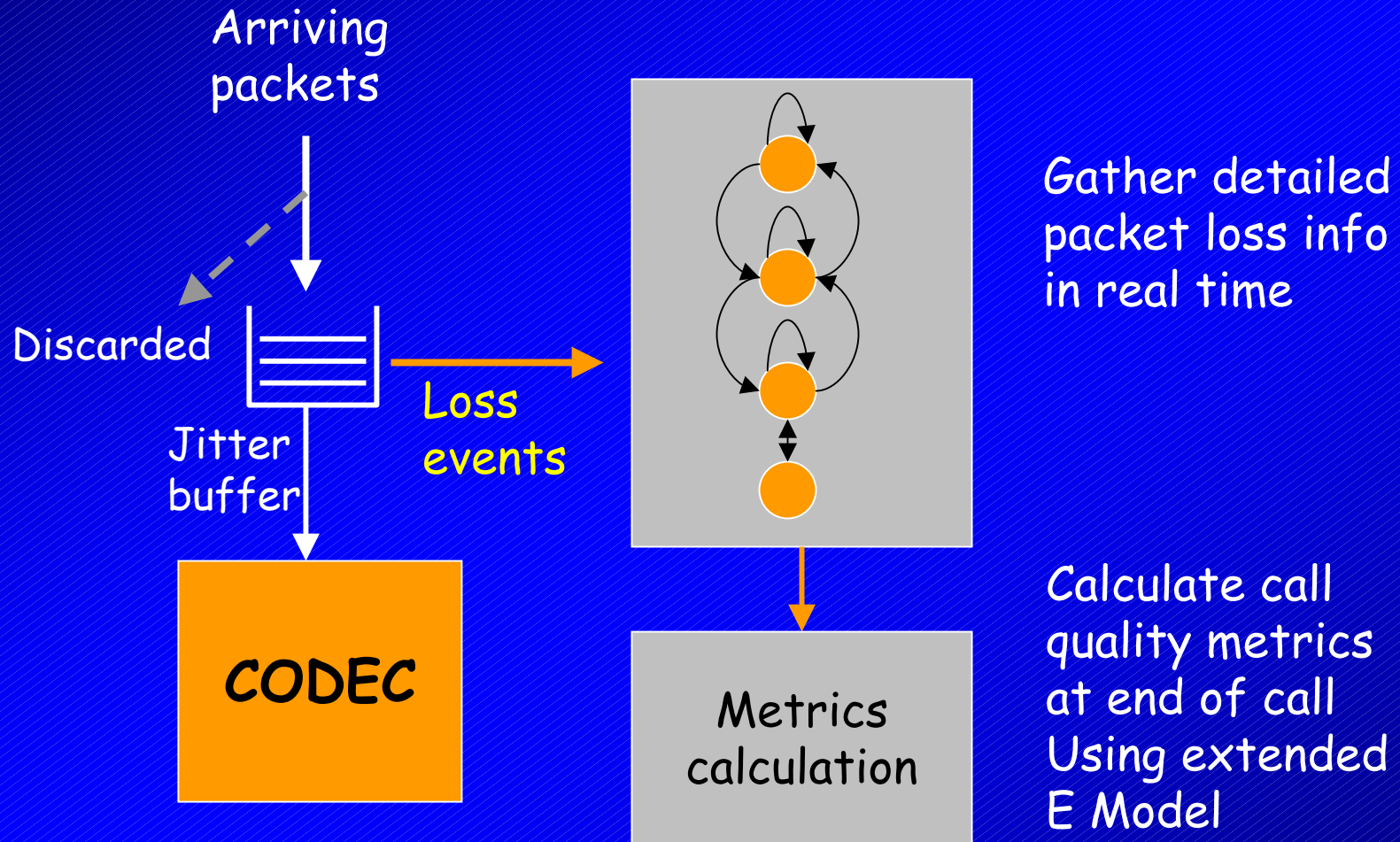


Source AT&T
T1A1.7/98-031

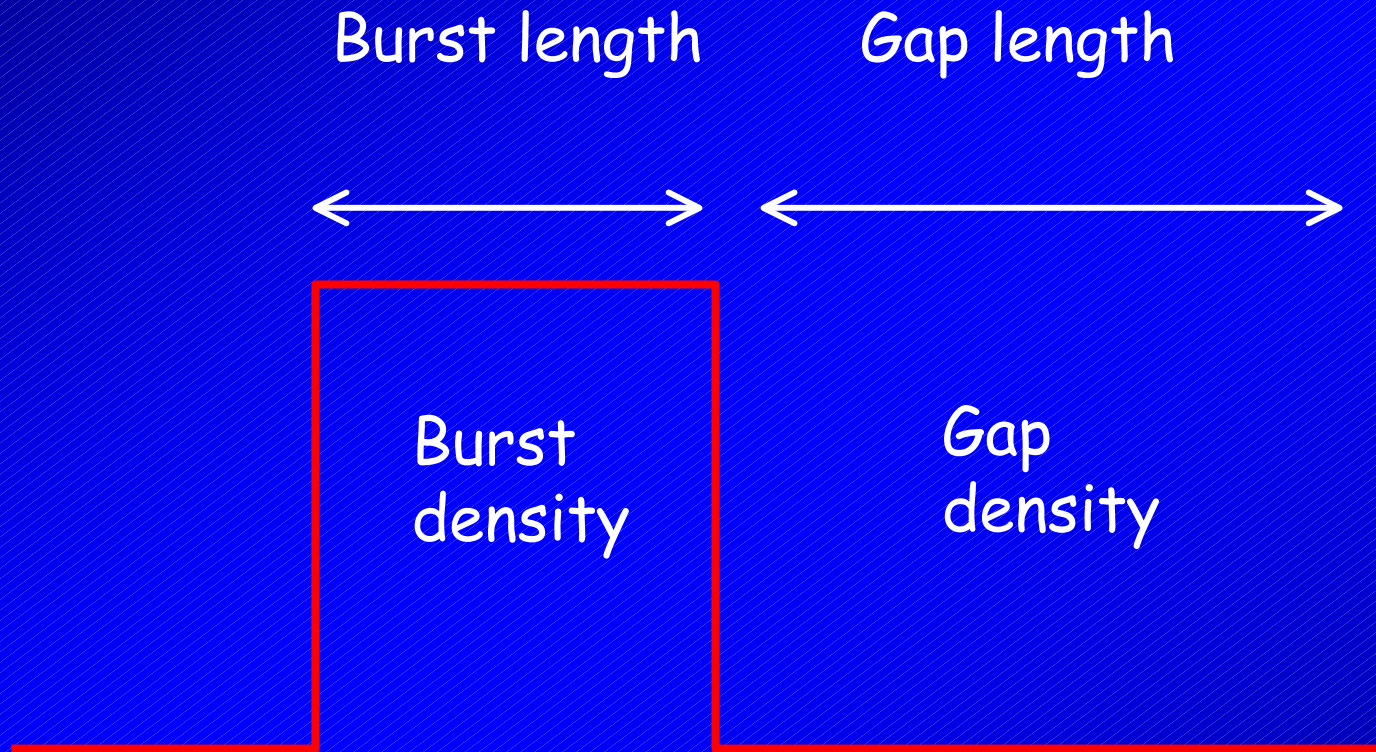
Approach



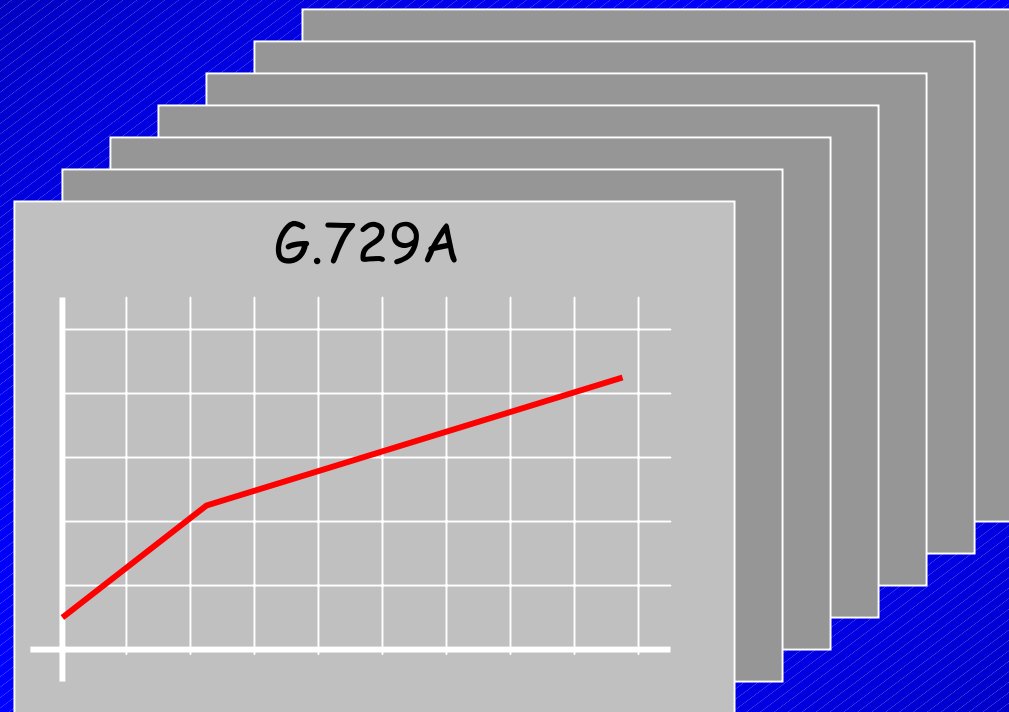
Efficient calculation



Measured packet loss distribution

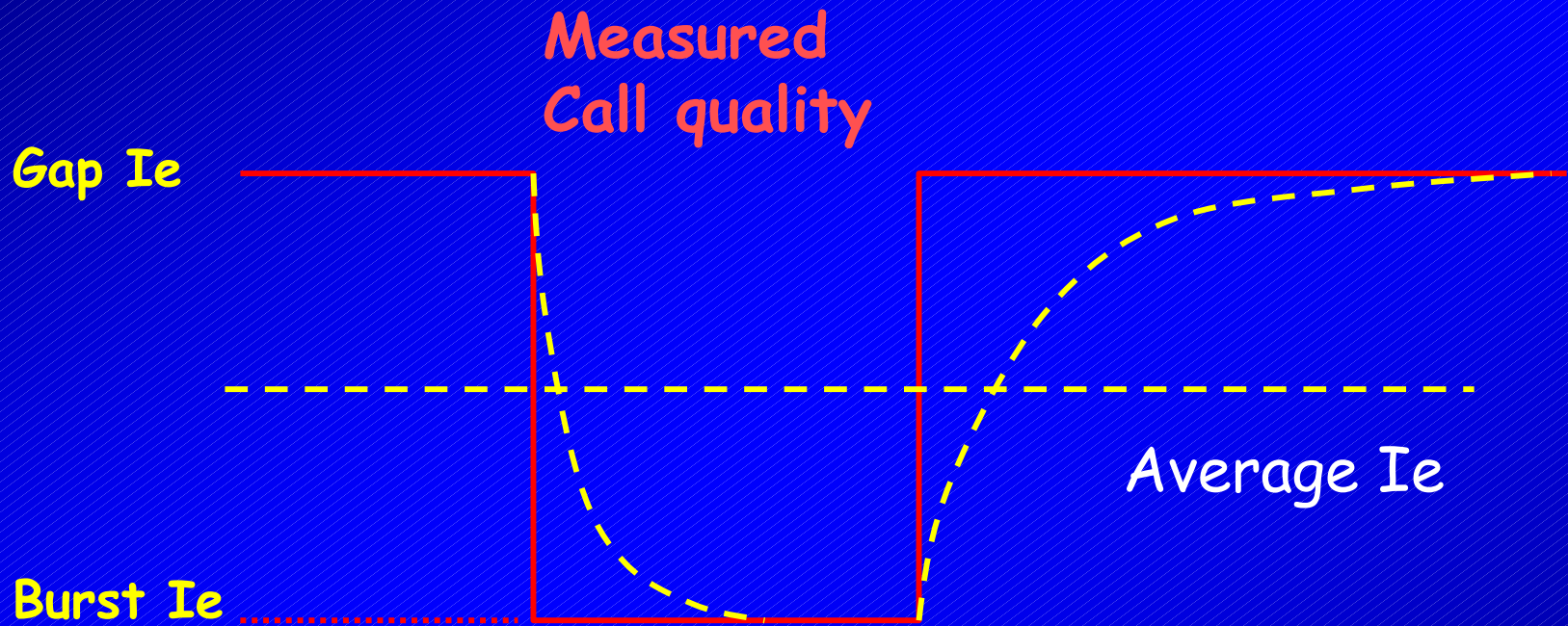


Packet Loss Model for each CODEC



Packet Loss Rate ->
Equipment Impairment factor (I_e)

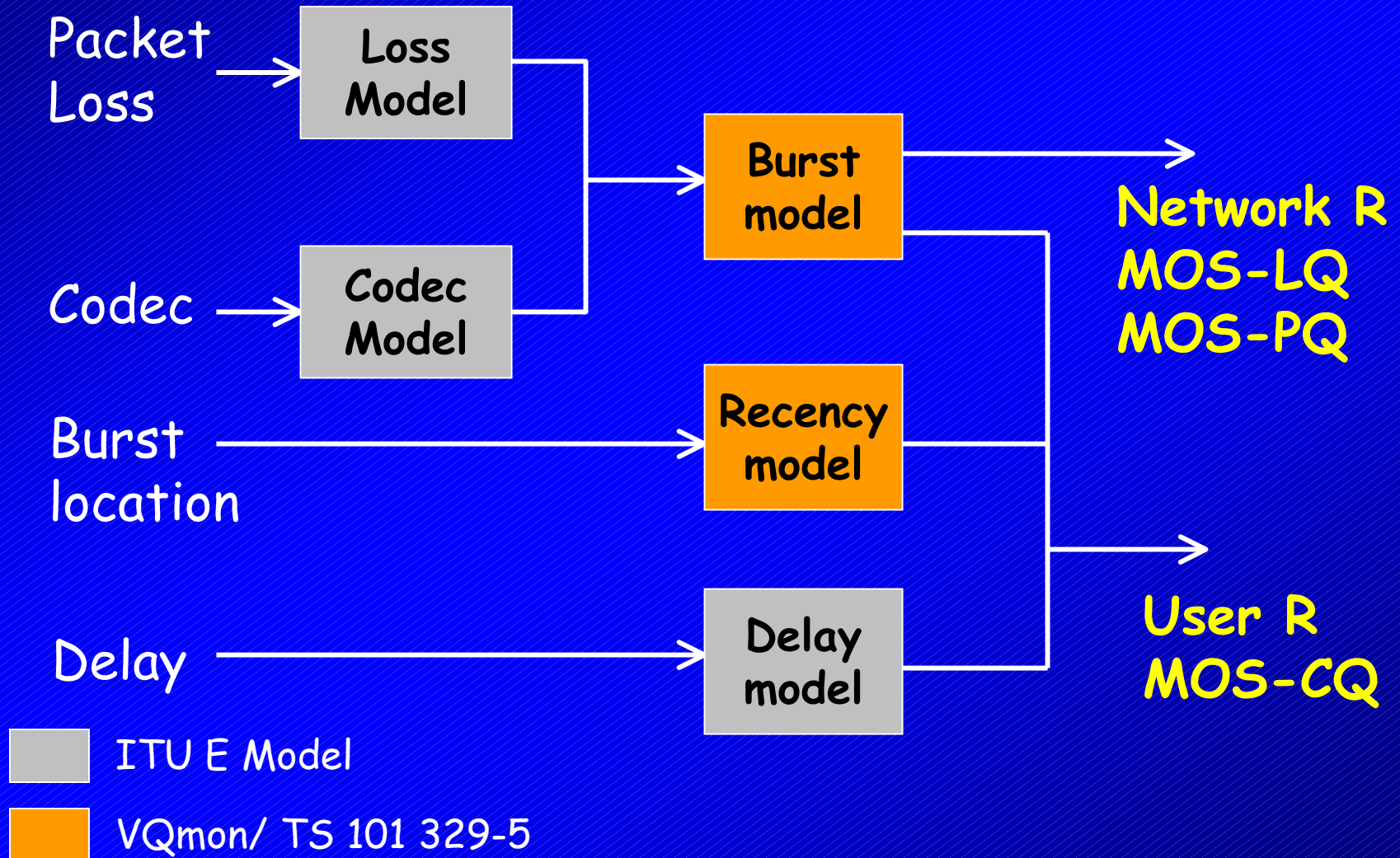
Applying Burst Model



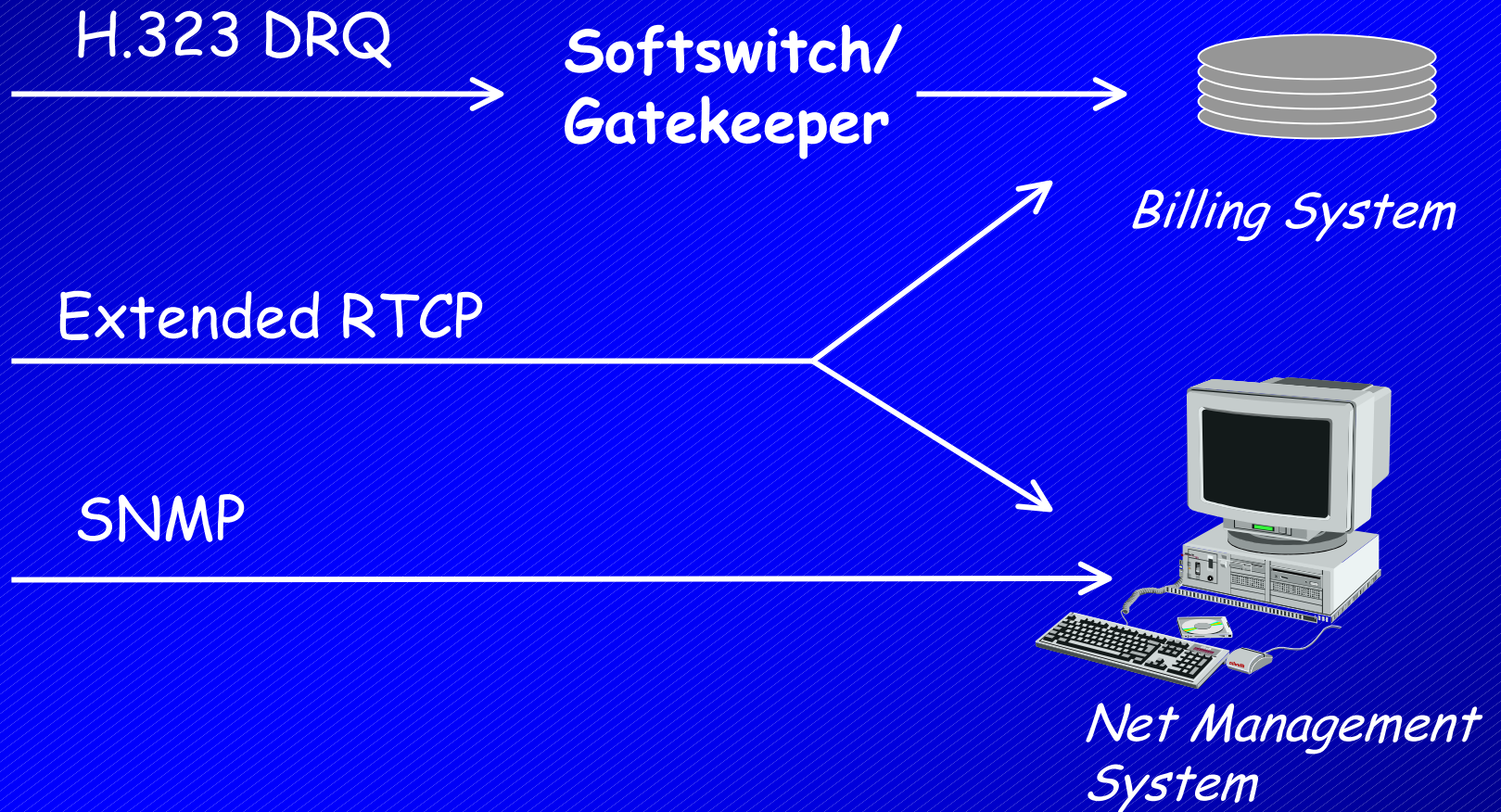
$$R = 94 - I_e - I_d$$

$$MOS = f(R)$$

Extended E Model



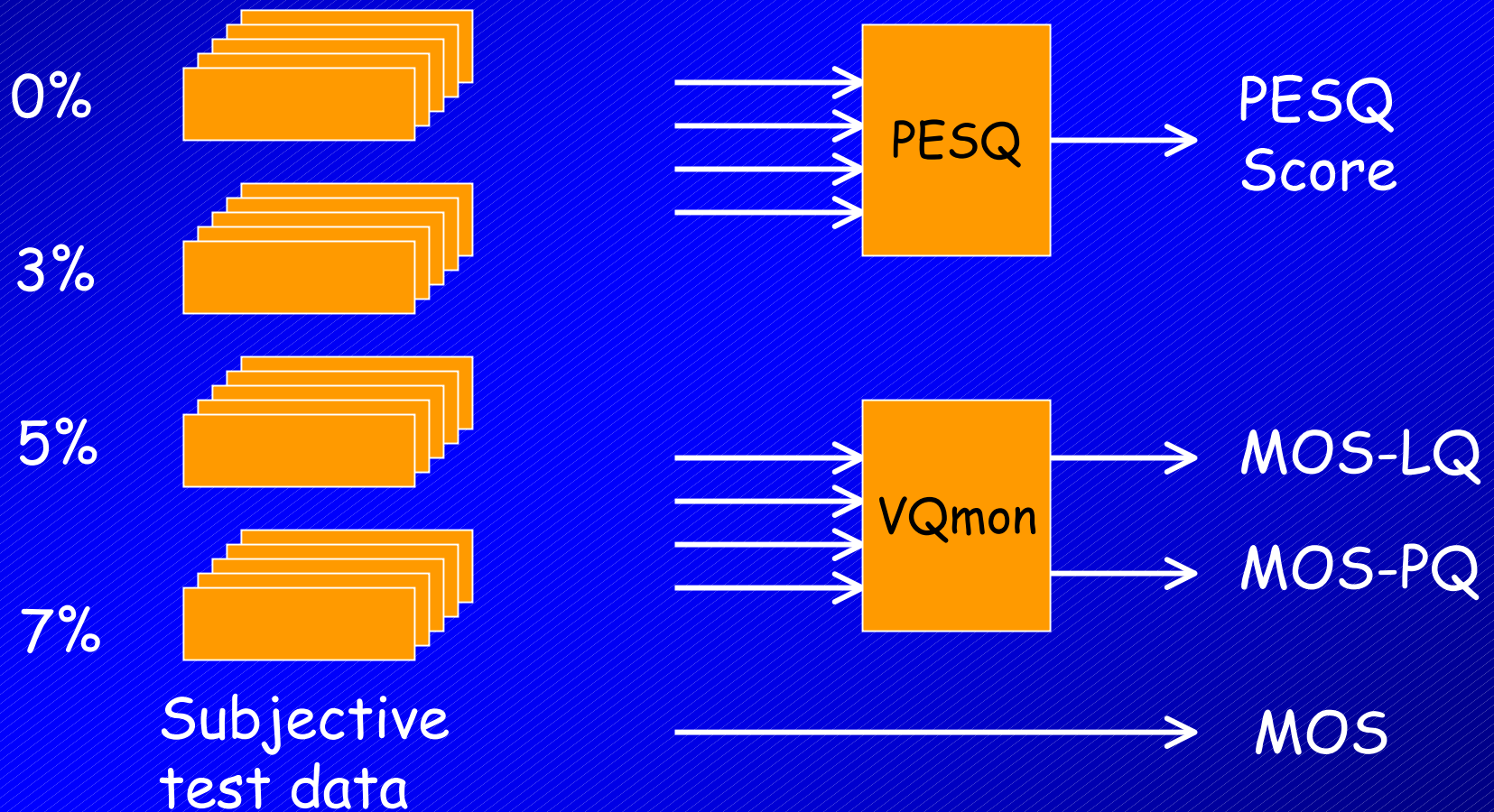
Extracting Metrics



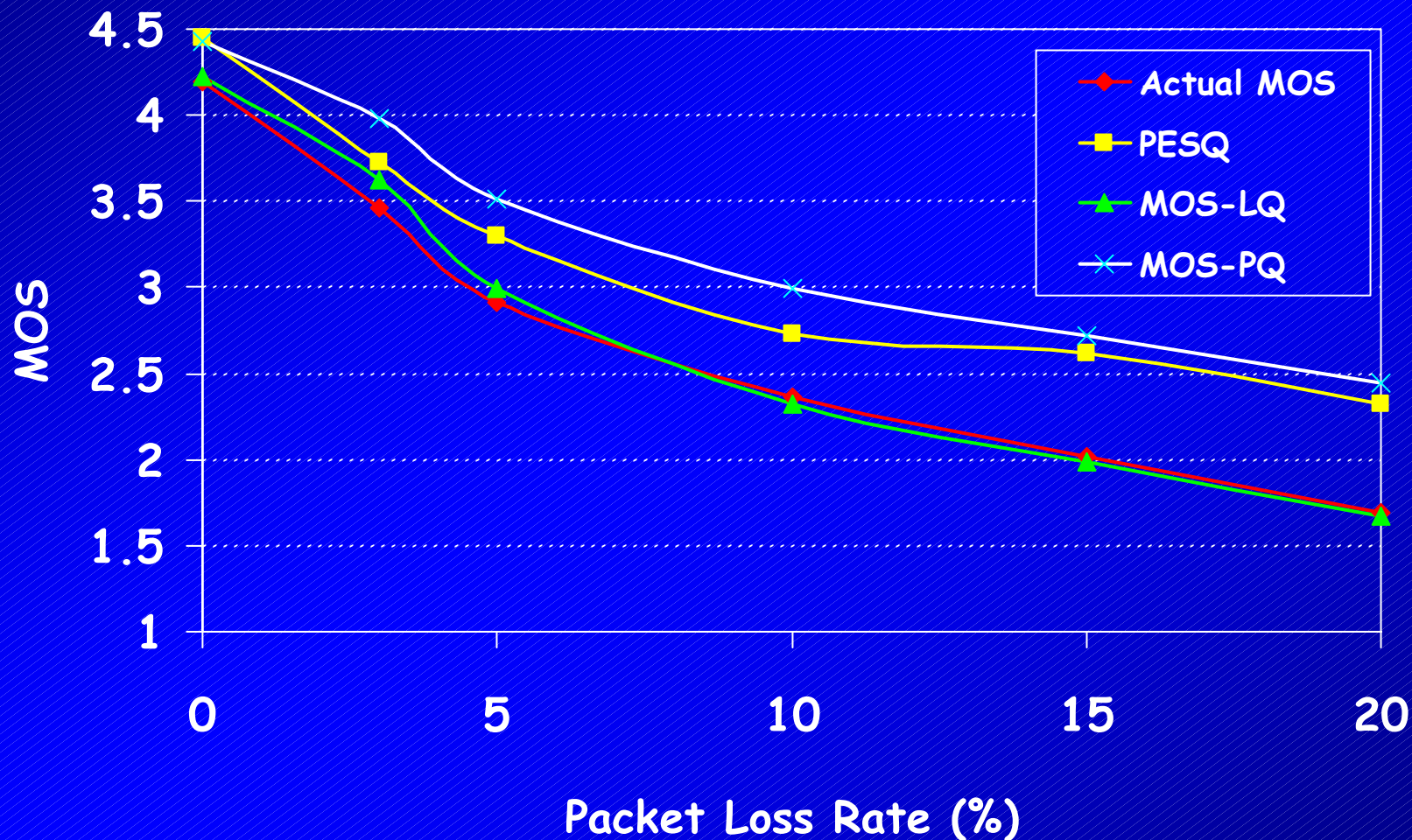
Implementation

- Embedded agent
 - Software that can be integrated directly into IP Gateway/ IP Phone
 - Approx 0.000075 MIPS per VoIP call
- Stream monitor
 - Software that can be integrated into network analyzers, SLA monitors.....
 - Approx 0.005 MIPS per VoIP call
 - Potential to monitor a Gigabit Ethernet using a 2GHz Pentium PC

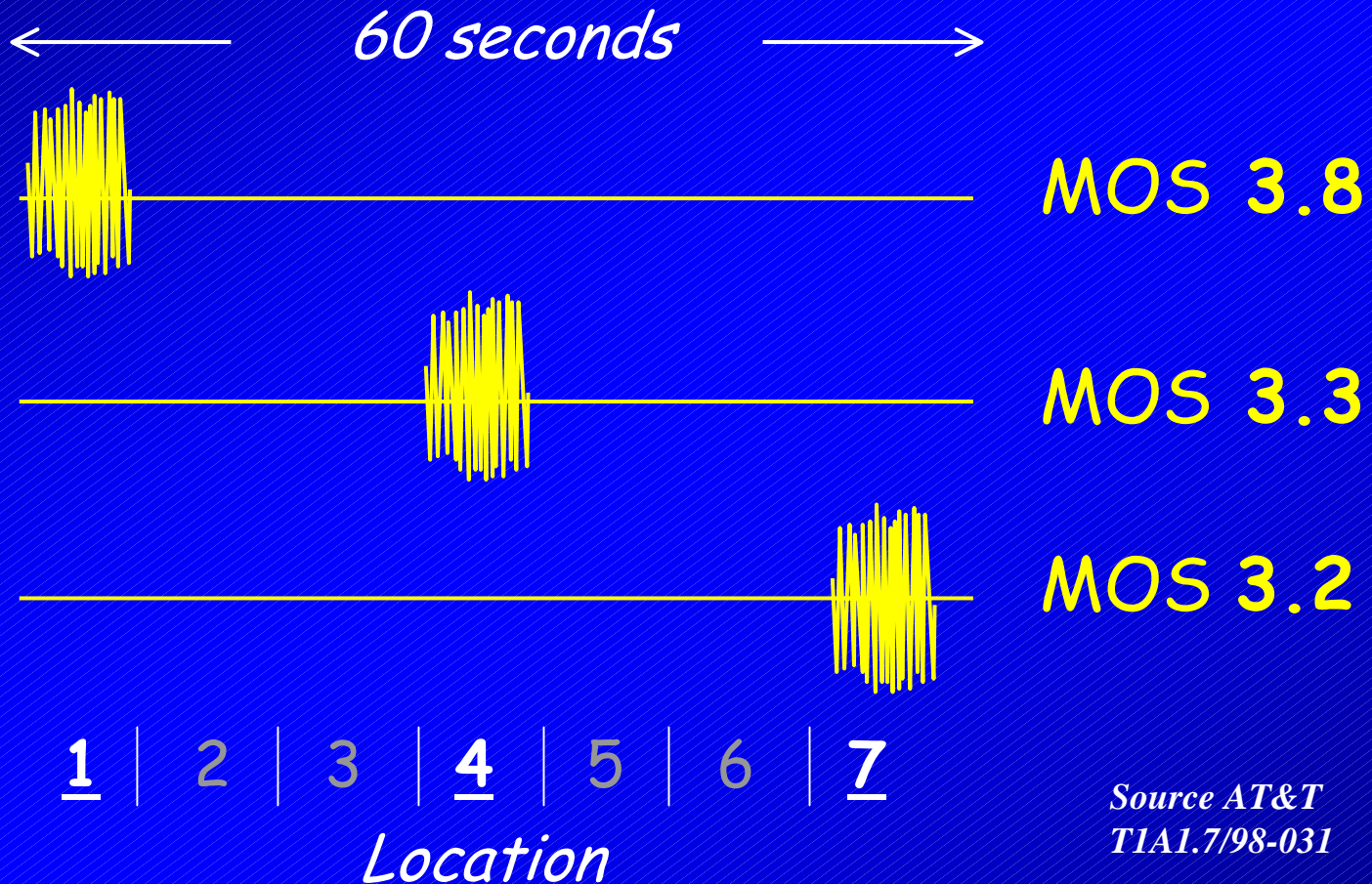
Comparison with PESQ and Subjective Score



VQmon MOS-LQ and MOS-PQ

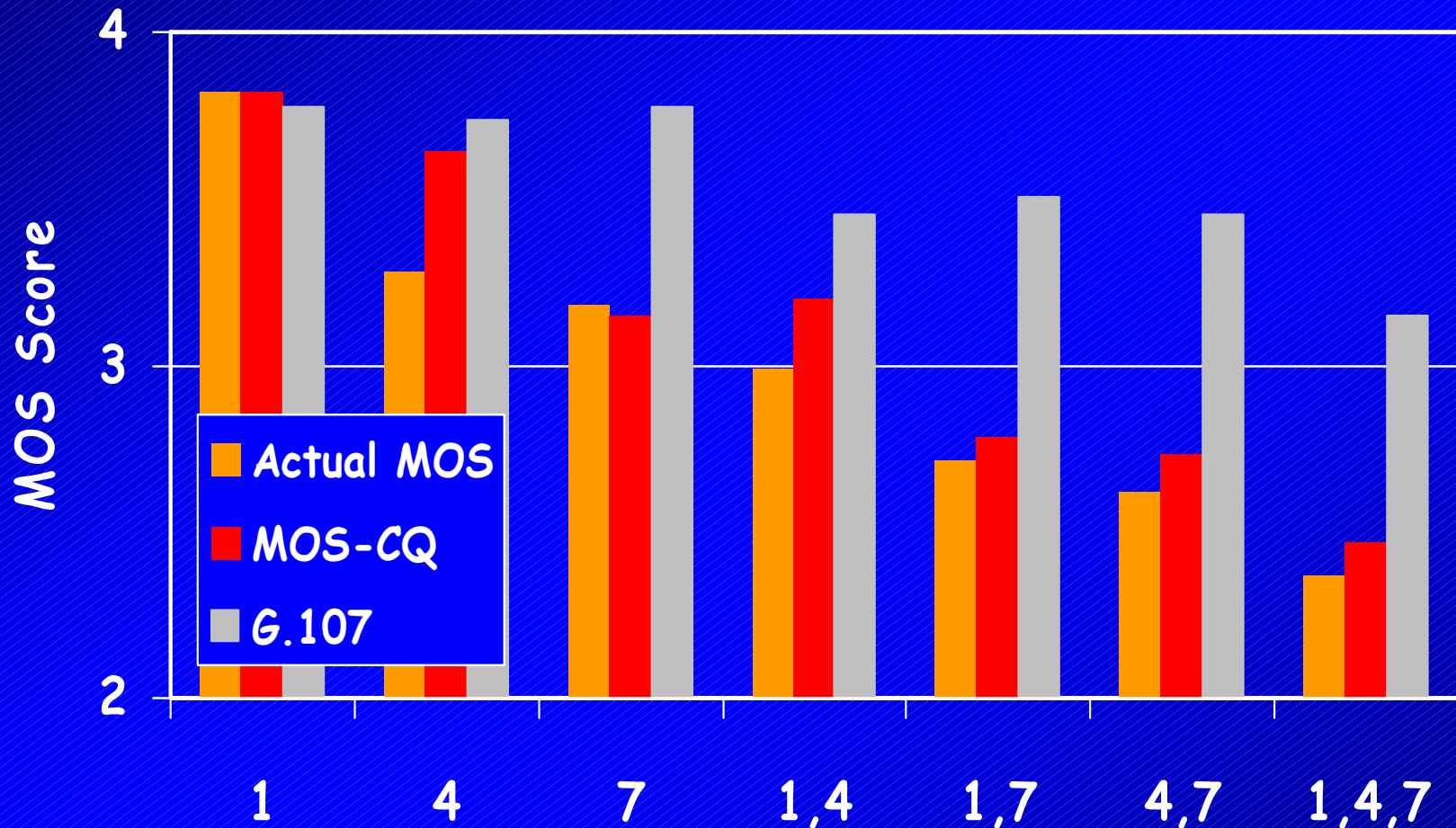


VQmon MOS-CQ vs G.107



Source AT&T
T1A1.7/98-031

VQmon MOS-CQ vs G.107



Impairment location

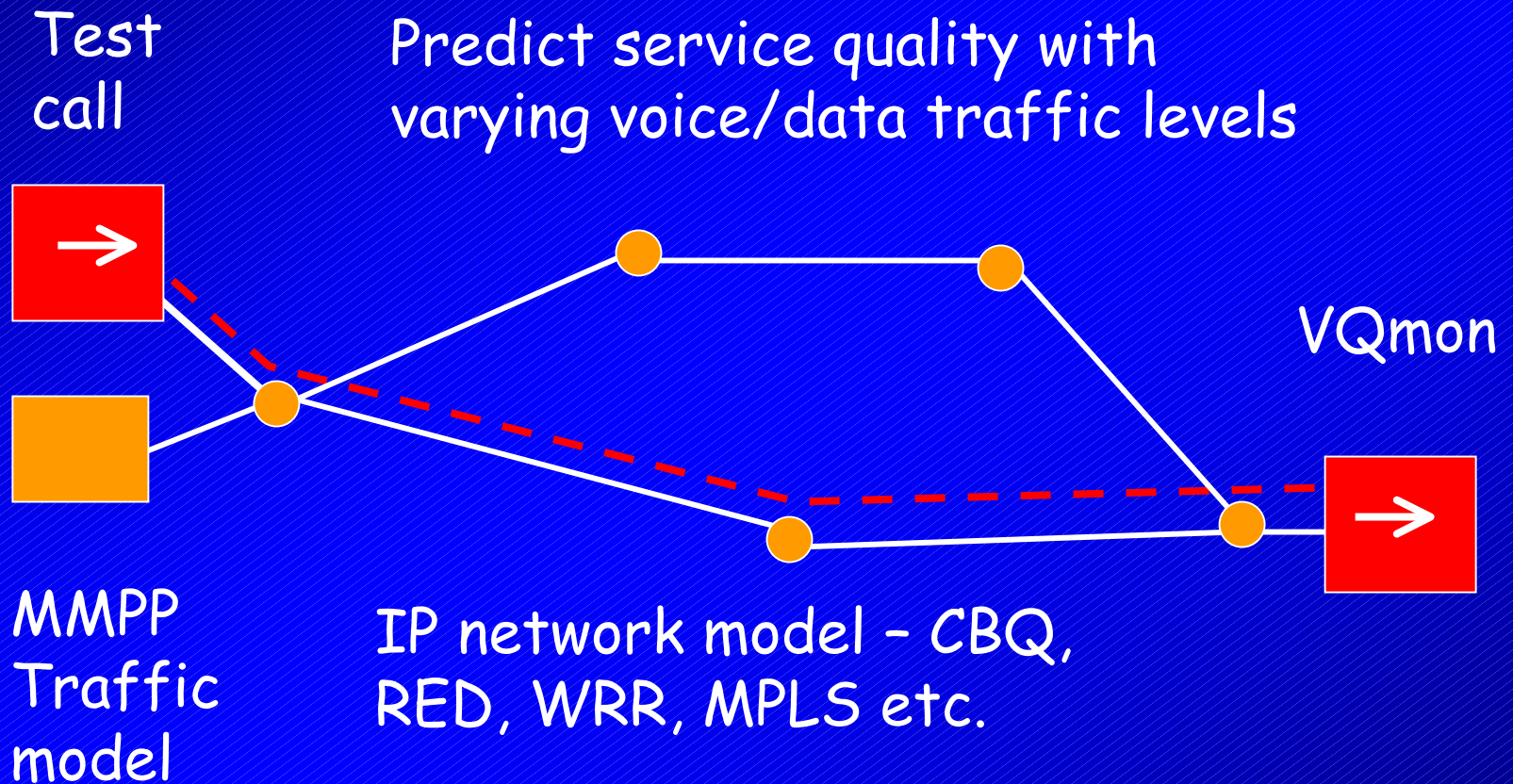
QSDG 2002

Telchemy

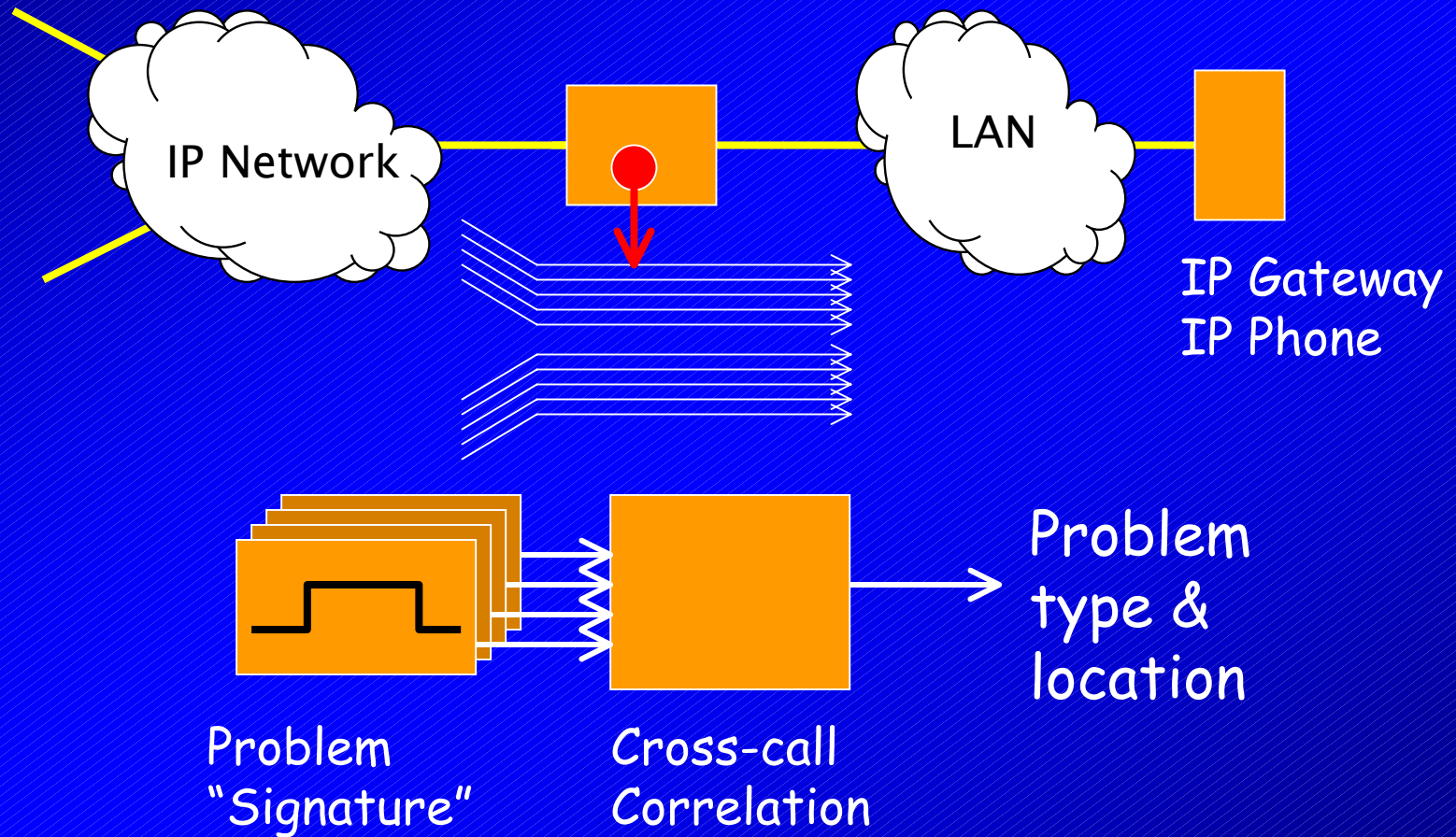
Other applications

- Modeling VoIP Service Quality
- Identifying network problem type and location

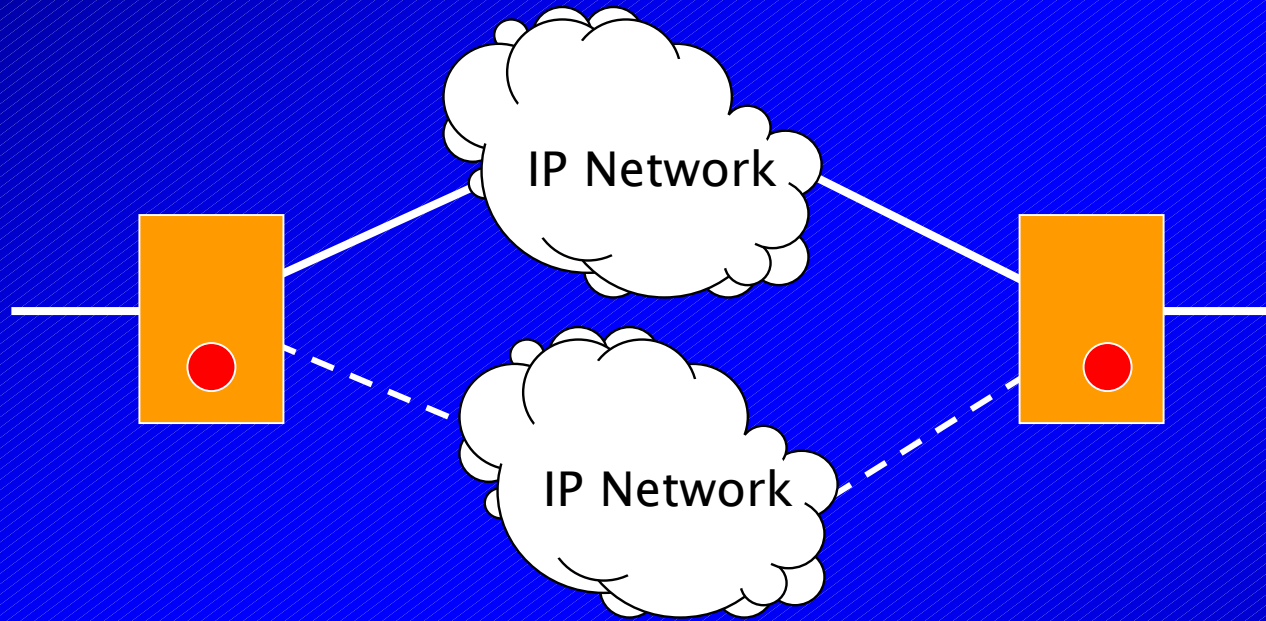
VoIP Service Modeling



Locating IP Network Problems



VoIP Route Selection



Select route based on estimate of subjective
quality metric - not components
Monitor live calls on active route
Generate test calls on inactive route

Key Points

- Packet loss is Bursty
- Call Quality is Time Varying
- For accurate results on live calls must consider time varying impairments
- Need to use the same algorithm at different locations in the network to monitor SLAs, isolate problems