



# Managing the Performance of Enterprise and Residential VoIP Services

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# Outline

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- Problems affecting Residential VoIP and Teleworkers
- VoIP Performance and Quality Challenges
- VoIP Performance Measurement and Problem Diagnosis
- Passive Monitoring using Embedded Agents
- Active Testing using Downloadable Agents
- Looking forward to IPTV

## Problems affecting Residential VoIP and Teleworkers

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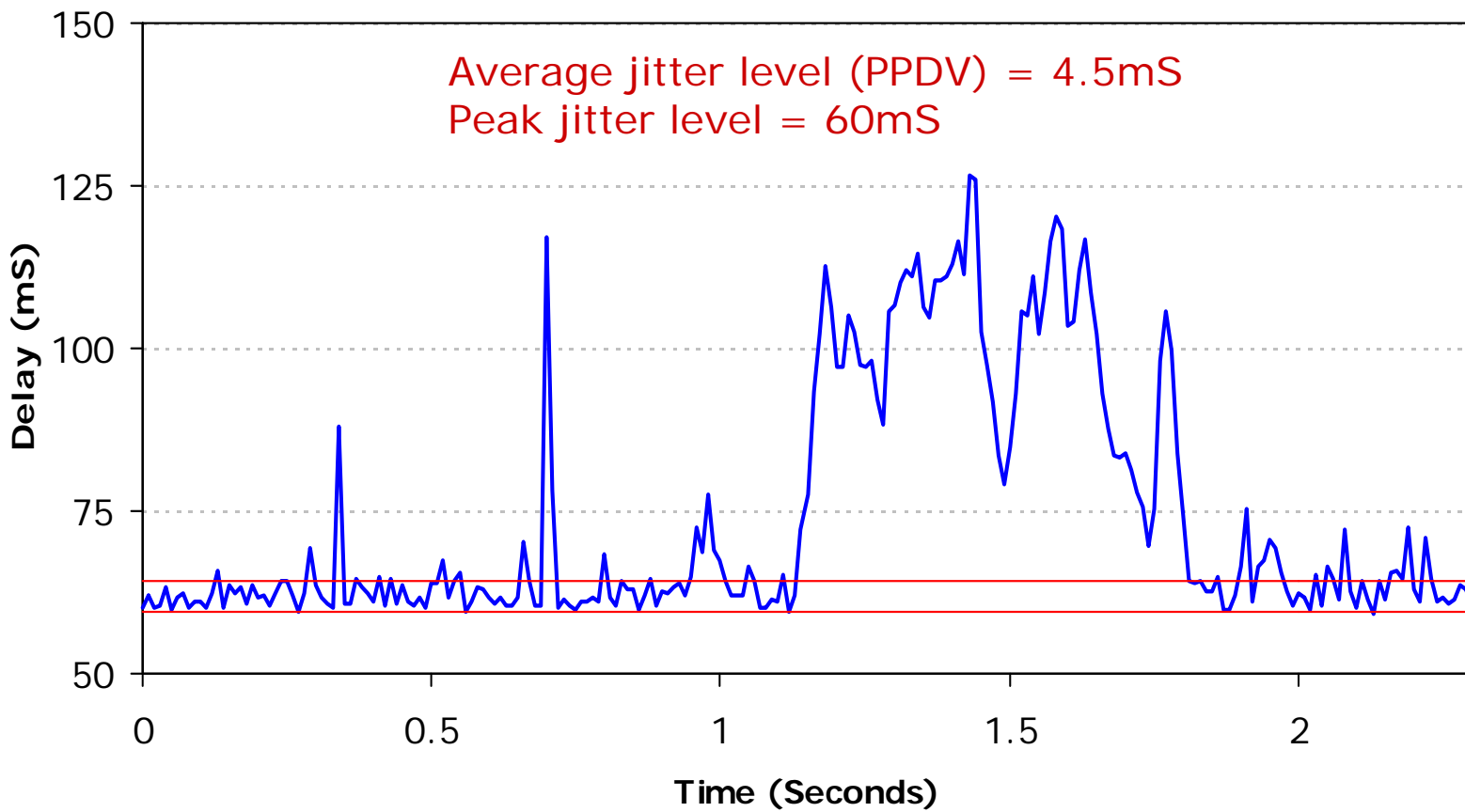
- Net Neutrality issues
  - Deliberate downgrading of priority
  - Artificial congestion
- Residential gateways
  - Performance
  - Difficult to control bandwidth allocation
  - “Bugs” affecting SIP/RTP connectivity
- Home networks
  - Home environment
  - Unexpected sources of congestion

# VoIP Performance and Quality Challenges

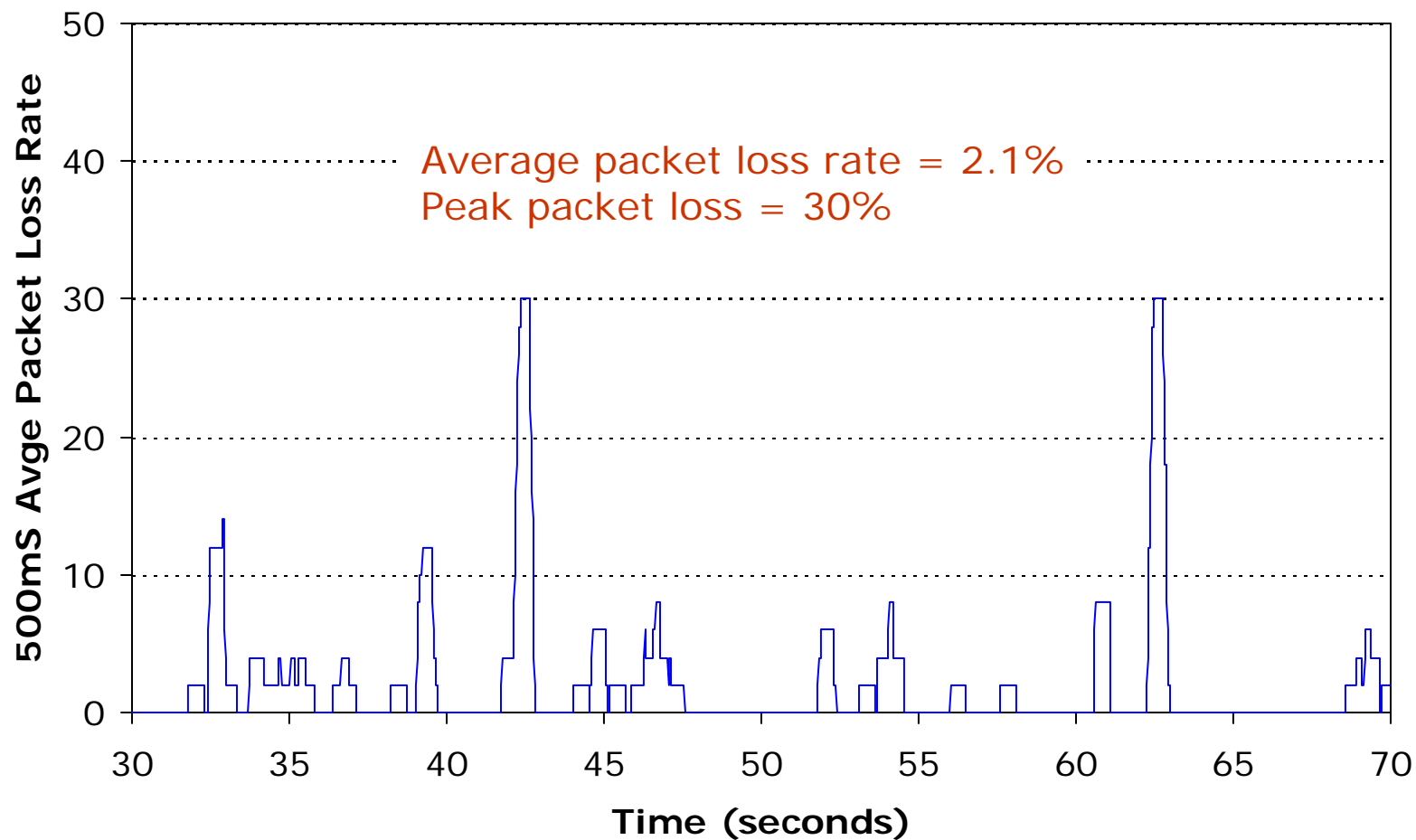
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- Lost packets and jitter due to network congestion
  - Typically time varying/ transient
- Packet loss due to layer 1-2 problems
  - Duplex mismatch, bit errors.....
- Delay
  - High delays lead to conversational problems
  - Even small delays make echo more obvious
- Echo
  - Many phone services suffer from echo
  - Not obvious if delay is very short
- Signal levels
  - High – clipping, Low – noisy speech, gaps
- Noise levels
  - Obvious but – should this be a factor in QoE metrics?

# Jitter measurements can be misleading!!!



# Packet Loss is time varying



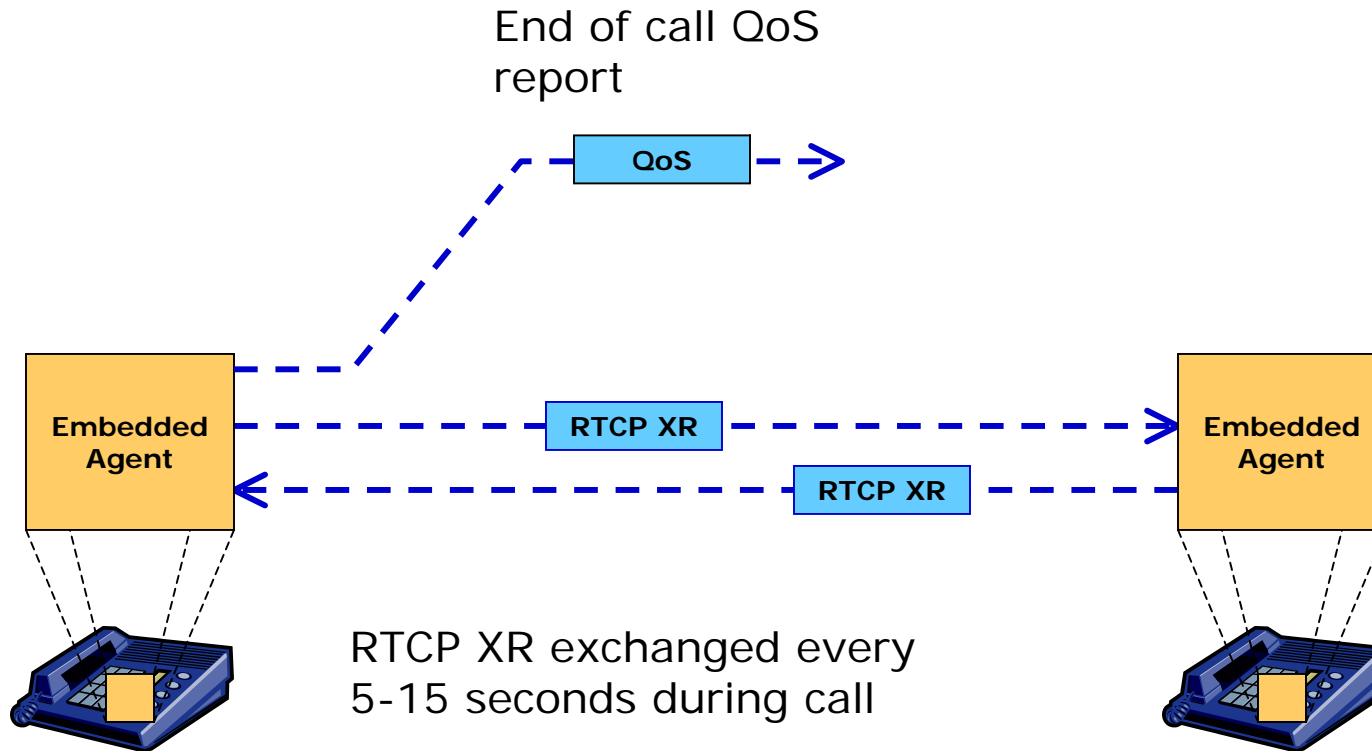
# Passive Monitoring using Embedded Agents

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- Basic premise
  - Measure performance close to the user
    - *Integrate a performance monitoring agent into the IP endpoint (IP phone)*
  - Send fewer but more meaningful reports
    - *Capture information and report info on transient problems*
  - Incorporate signal, noise, echo information but avoid the need to decode the signal
    - *Leverage data already available from the endpoint's codec/ echo canceller*
  - Conversational quality calculation requires some information from the remote endpoint
    - *Need combined information exchange/ reporting protocol*

# VoIP Performance Reporting Architecture

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# How to measure performance?

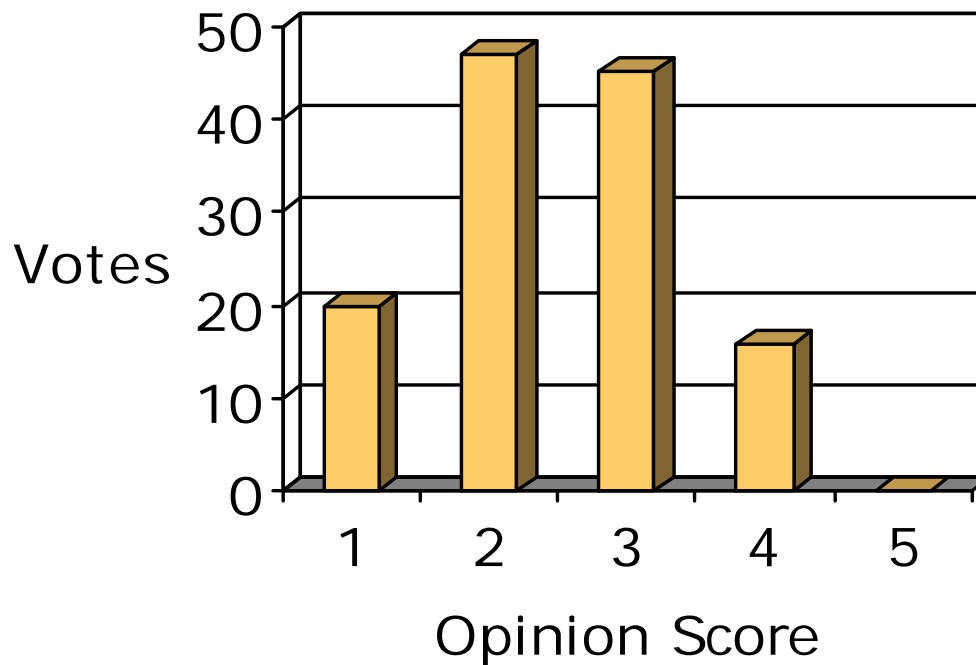
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- Subjective quality – “MOS”
  - Listening quality
  - Conversational quality
- Objective measurement

# What is MOS?

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- Extract from an ITU subjective test
- Mean Opinion Score (MOS) was 2.4
- 1=Unacceptable
- 2=Poor
- 3=Fair
- 4=Good
- 5=Excellent



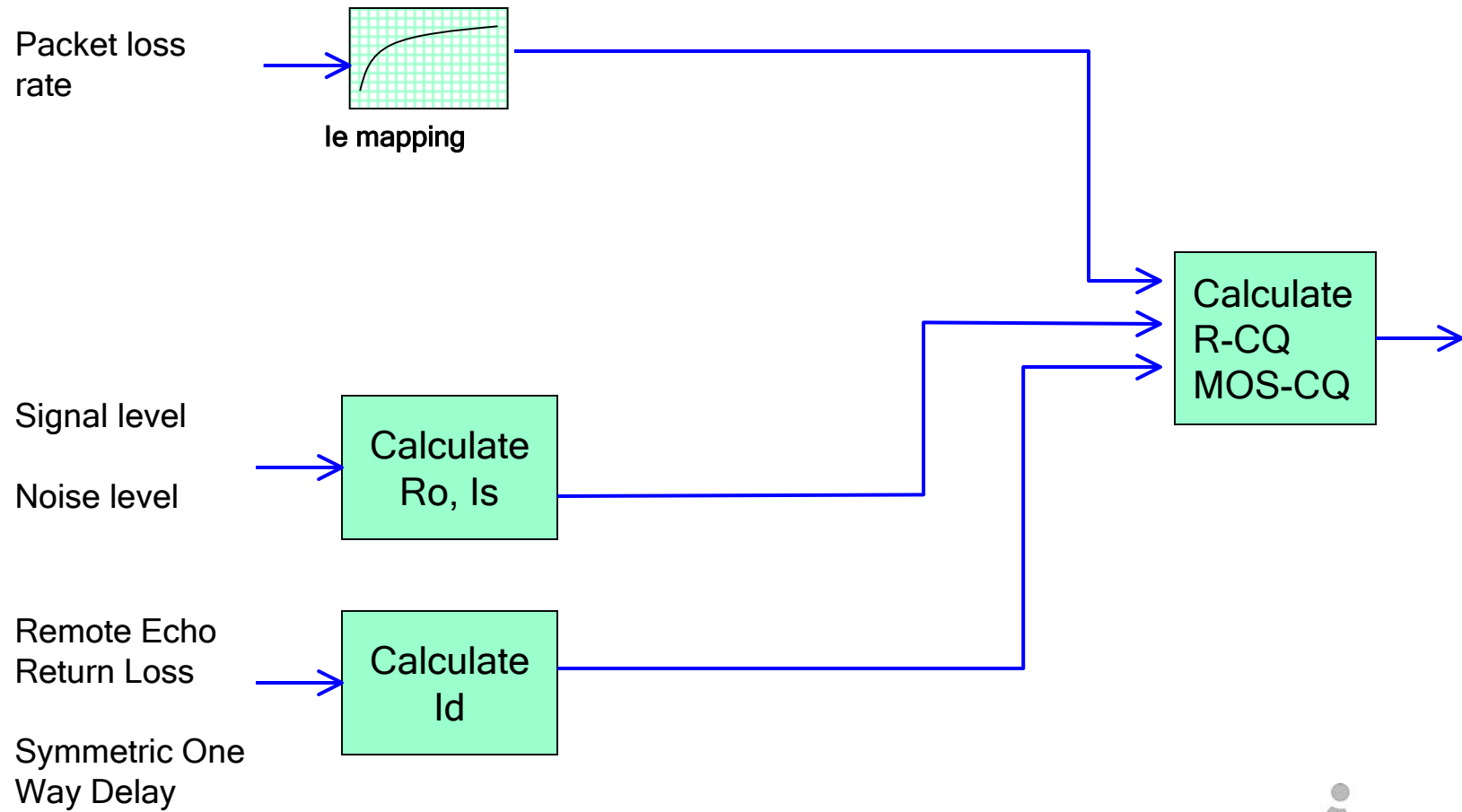
## Objectively estimating MOS

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- VQmon
  - Passive measurement algorithm, widely used in VoIP test tools and IP phones/ gateways (tested vs P.562)
- ITU G.107
  - Transmission planning model
  - Used for monitoring but does not comply with P.562
- ITU P.562
  - Standard for testing IP based passive measurement algorithms
- ITU P.563
  - Voice sample based passive measurement algorithm, very compute intensive
  - Only accurate when averaged over many calls
- ITU P.862 – PESQ
  - Full reference

# E model

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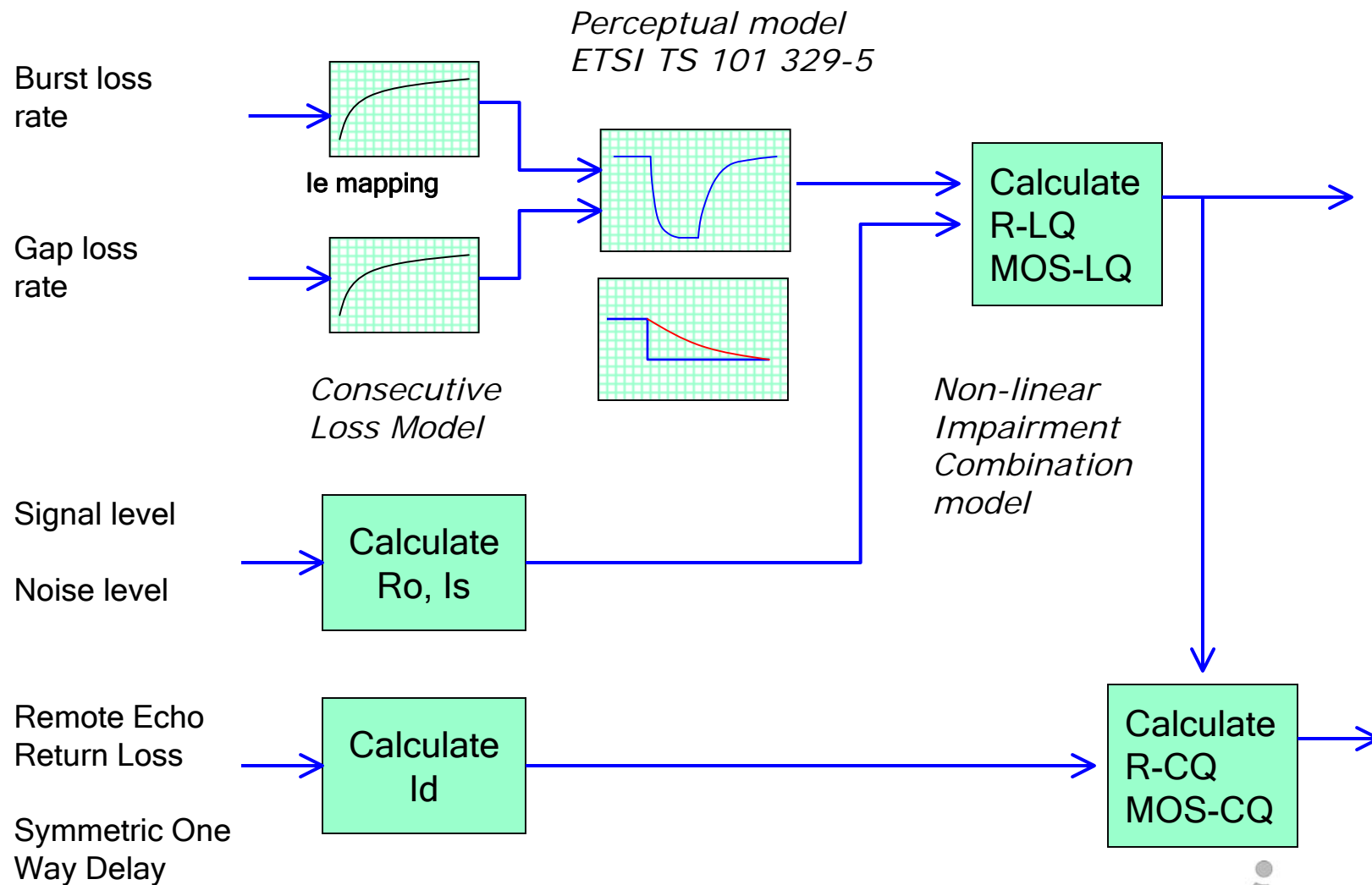


## Problems with the E Model

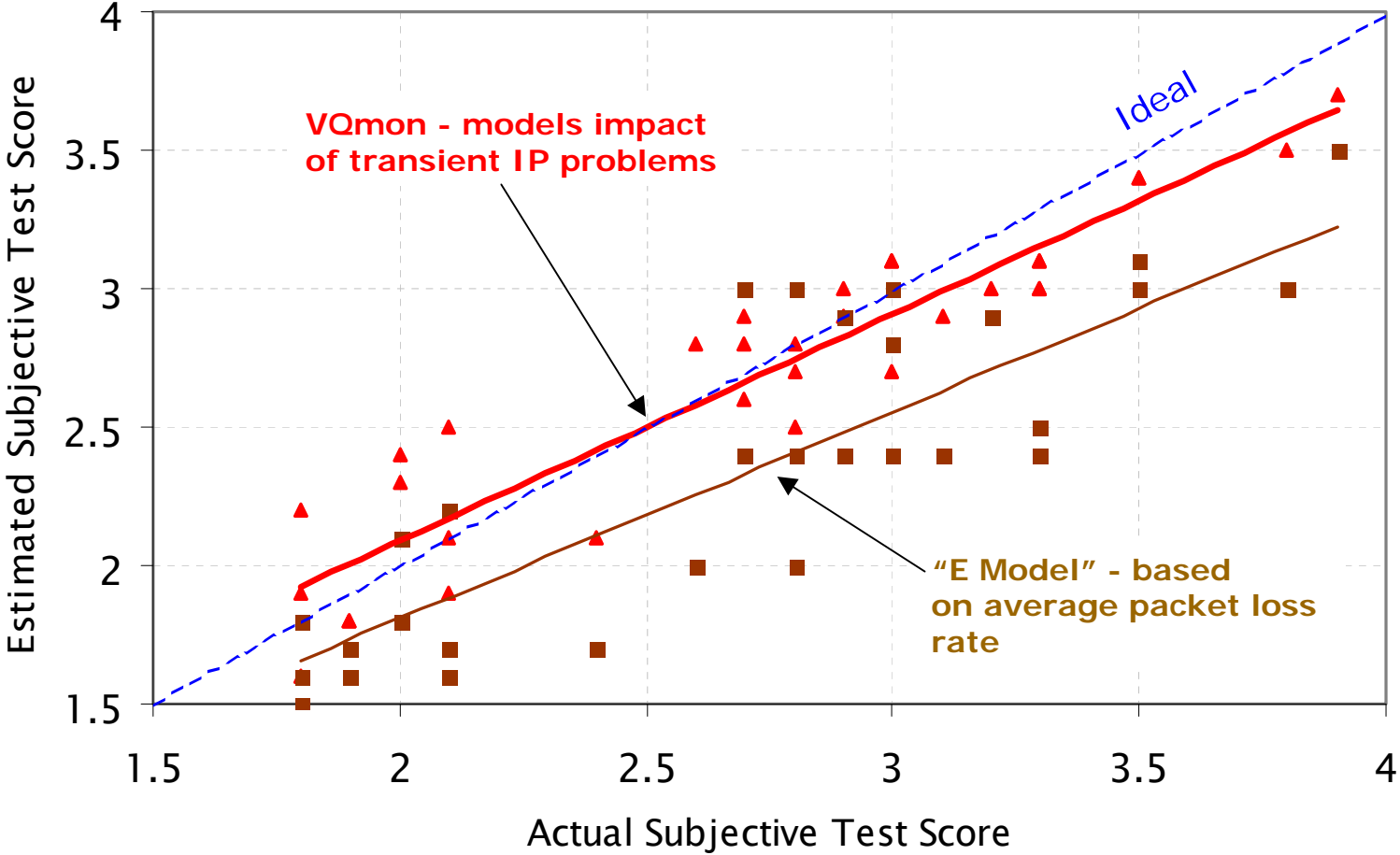
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- Additivity problem
  - Impairments aren't additive
- Inaccurate for time varying loss
- Inaccurate for consecutive loss
- Conversational quality only

# VQmon computational model



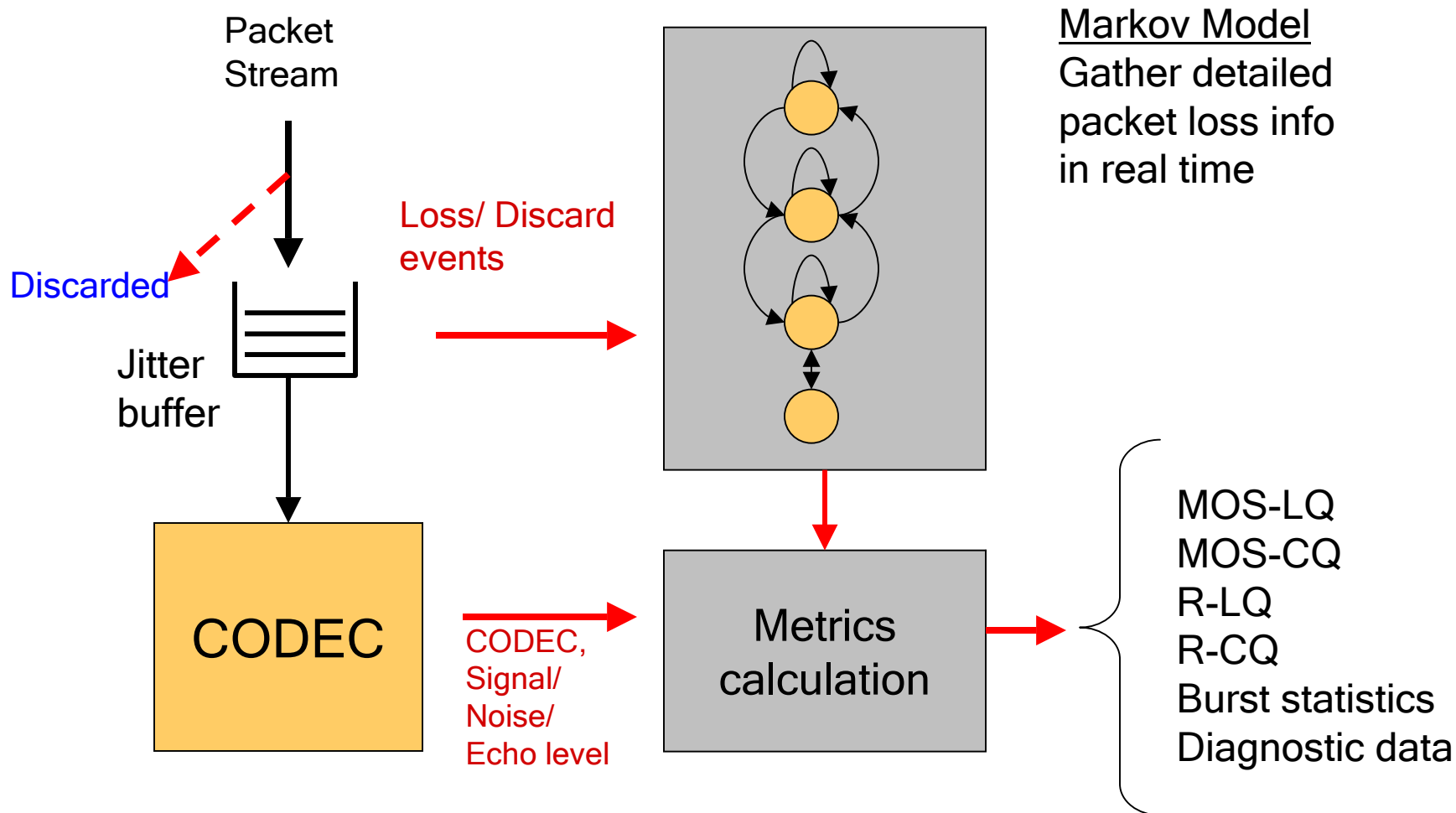
# Impact of TS 101 329-5 model



Independent test by France Telecom and University of Bochum

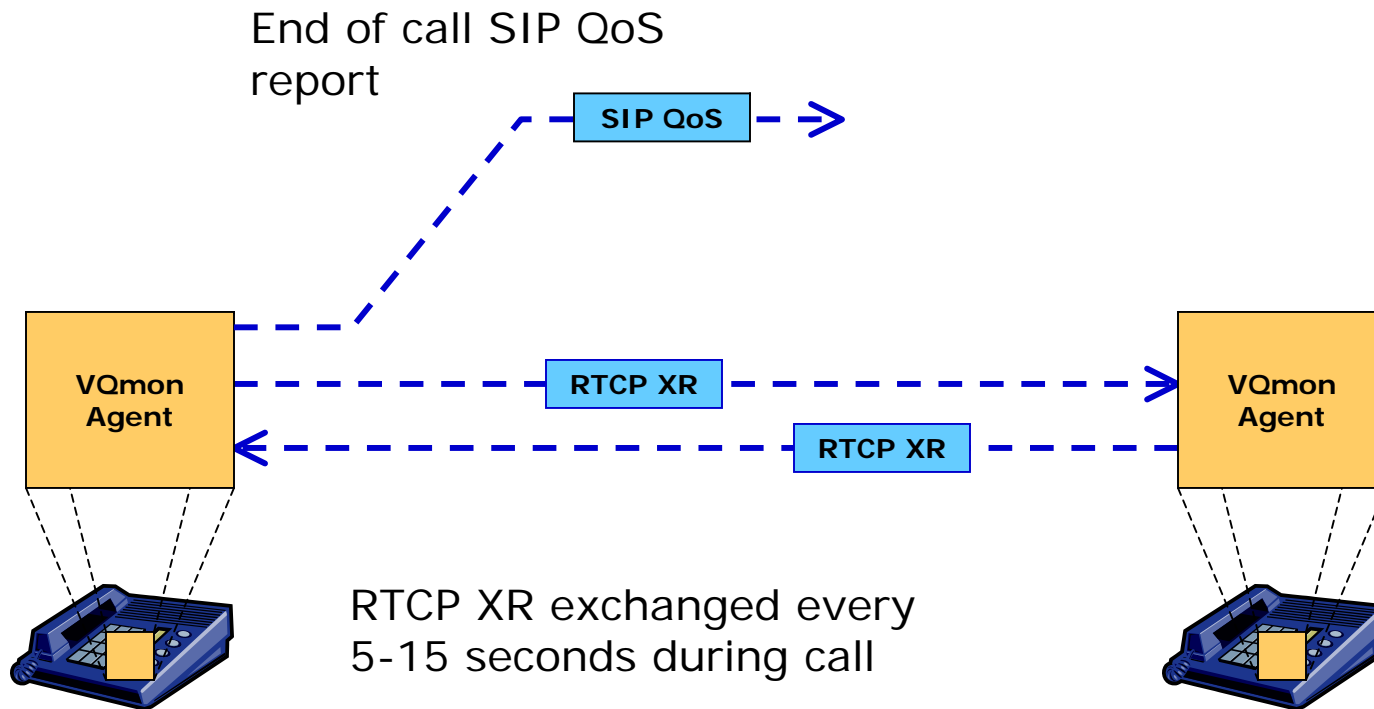


# VQmon



# VoIP Performance Reporting Architecture

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# RTCP XR Reports

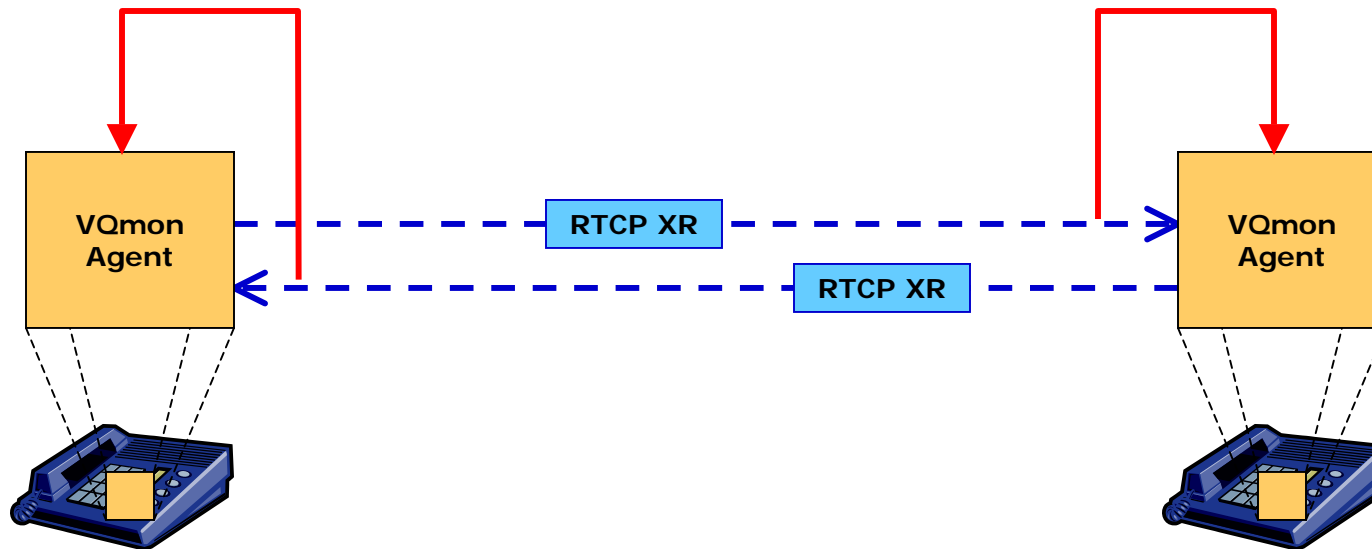
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Loss Rate	Discard Rate	Burst Density	Gap Density
Burst Duration (mS)		Gap Duration (mS)	
Round Trip Delay (mS)		End System Delay (mS)	
Signal level	RERL	Noise Level	Gmin
R Factor	Ext R	MOS-LQ	MOS-CQ
Rx Config	-	Jitter Buffer Nominal	
Jitter Buffer Max		Jitter Buffer Abs Max	

# Information exchange using RTCP XR

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Remote Echo Return Loss  
Remote End System Delay  
Remote Signal Level

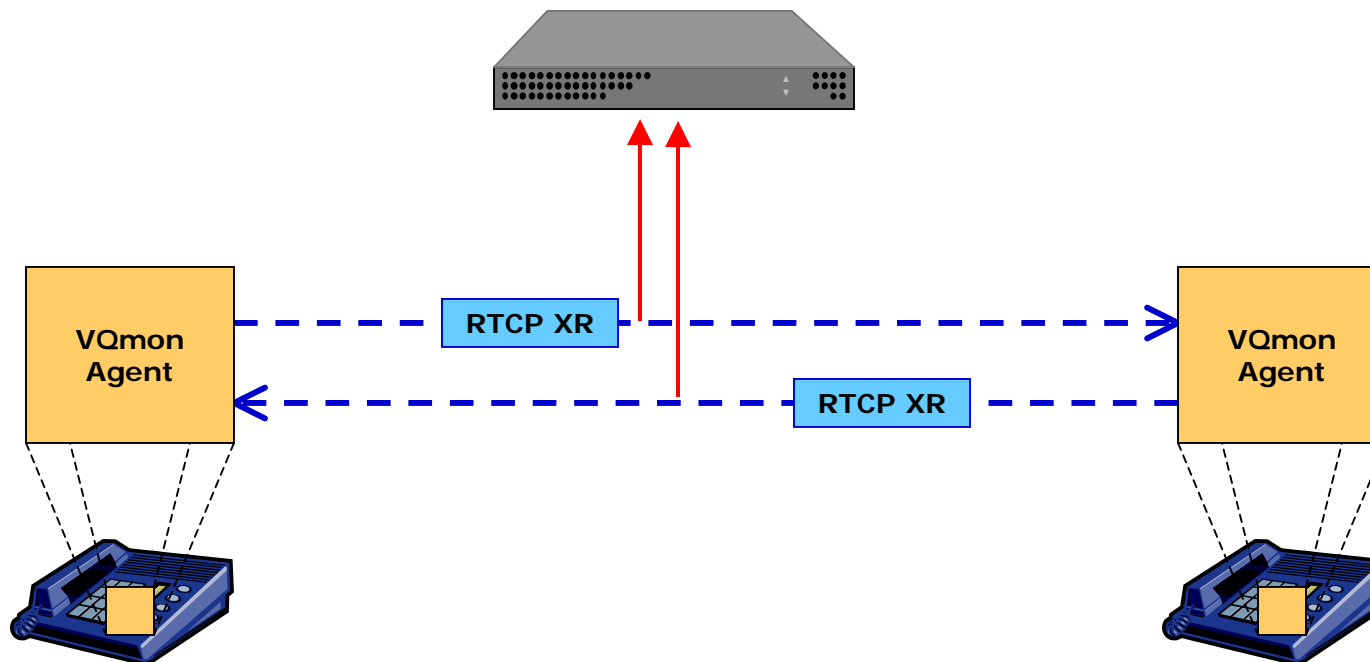


# RTCP XR as a reporting protocol

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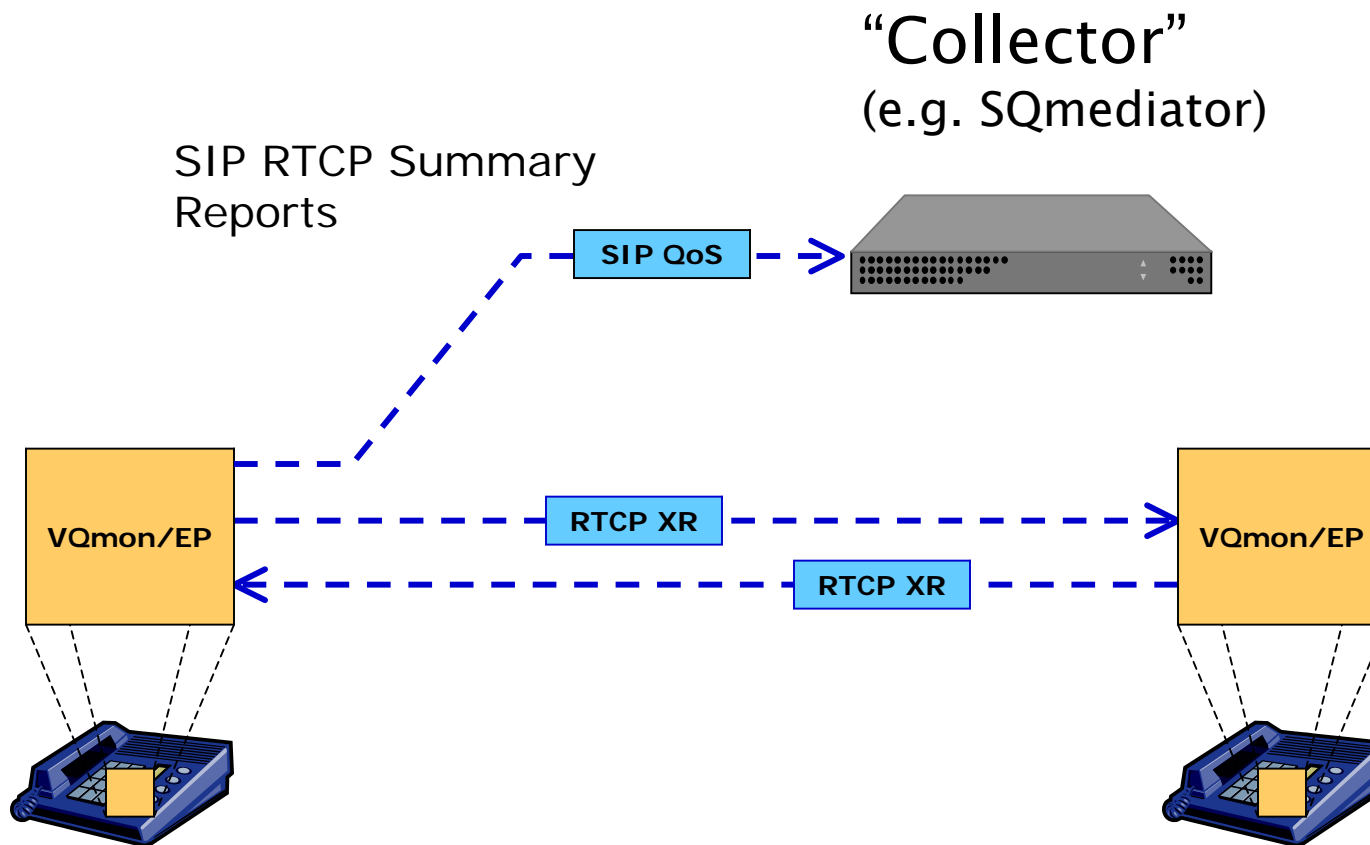
## Probe or Analyzer

- can incorporate signal, noise, echo, delay from XR reports
- Can report for both midpoints and endpoints



# SIP as a reporting protocol

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# SIP RTCP Summary

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PUBLISH sip:collector@example.com SIP/2.0

.....

Content-Type: application/rtpcpxr

VQSessionReport

LocalMetrics:

TimeStamps=START:10012004.18.23.43 STOP:10012004.18.26.02

SessionDesc=PT:0 PD:G.711 SR:8000 FD:20 FPP:2 PLC:3 SSUP:on

CallID=1890463548@alice.uac.chicago.com

.....

Signal=SL:2 NL:10 RERL:14

QualityEst=RLQ:90 RCQ:85 EXTR:90 MOSLQ:3.4 MOSCQ:3.3

QoEEstAlg:VQMonv2.1



# Passive Monitoring – Pro's and Con's

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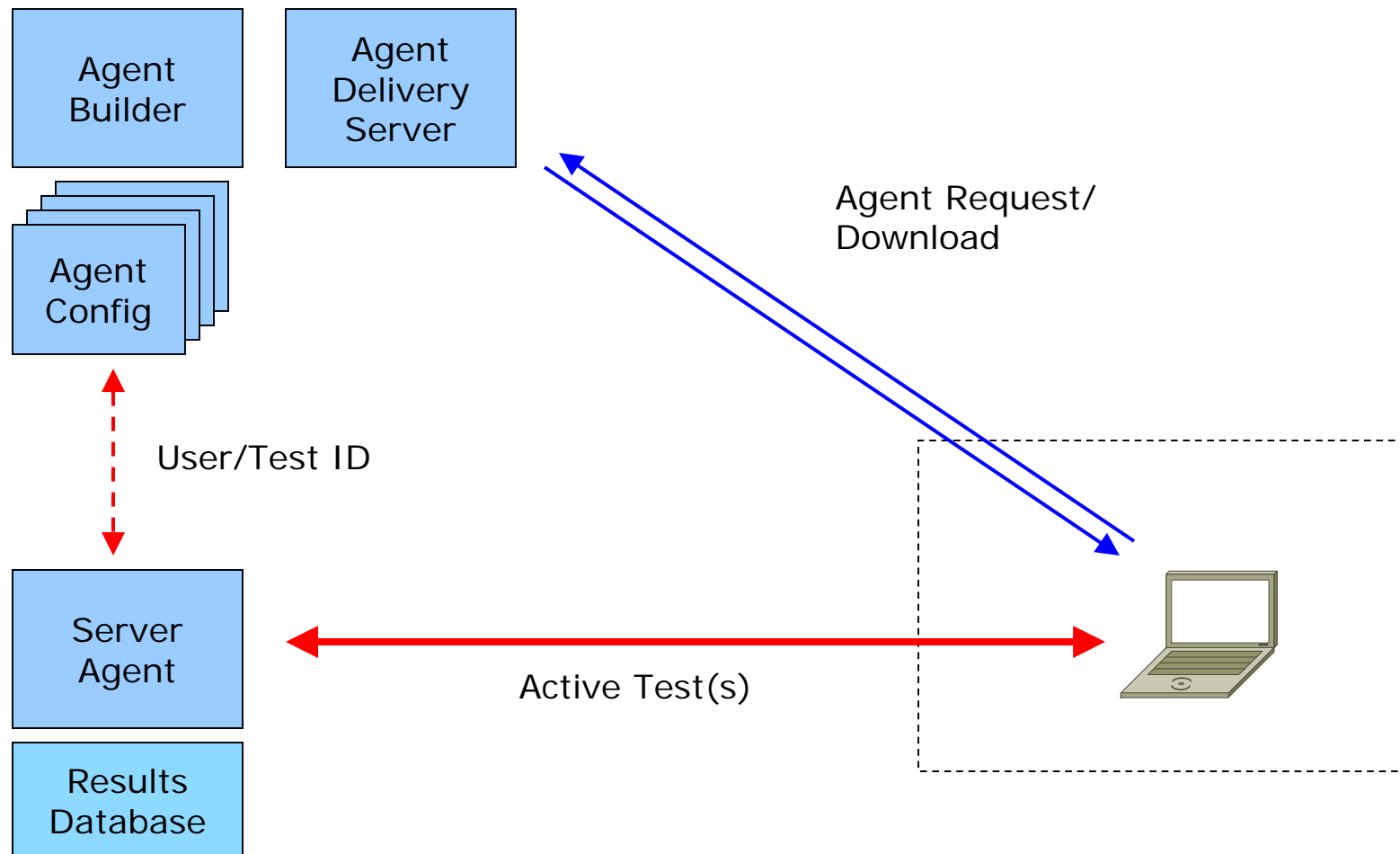
- Pro's
  - Understand problems affecting actual calls
  - Capture information that can be post-analyzed to address customer problem reports
  - Generates virtually no overhead traffic
- Con's
  - Can't detect a problem until it impacts a live call
  - Limited ability to run additional diagnostic tests to isolate and diagnose problem
  - Can't deal with failed connections

## Downloadable Active Test Agents

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- Test agents (VoIP, IPTV...) downloadable from web server
- Agents run tests against a network based server
- Agents initiate tests – minimizes firewall problems
- Results retained by server
- Agents can be static or transient

# Downloadable Agent - Architecture



# Example VoIP test process

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- Call connect
  - *Try SIP on port 5060*
  - *Try SIP on alternate port*
  - *Try encrypted SIP on alternate port*
  - *Agent reports what it had to do in order to connect*
- Media path quality
  - *Establish bidirectional RTP stream using voice payloads, measure quality*
  - *Establish bidirectional “non-RTP” stream, measure quality*

# What can active tests discover?

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- Call connect tests
  - Service provider may be blocking SIP
  - Residential gateways can (mistakenly) block SIP
  - Problem accessing server
  - .....
- Media path quality
  - Network congestion
  - Layer 1-2 problems, duplex mismatch, bit error problems.....
  - Net (non) neutrality

## Use Active or Passive tests?

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- Need both
- Active tests
  - Predeployment testing
  - SLA monitoring
  - Troubleshooting
- Passive monitoring
  - Service assurance
  - SLA monitoring
  - Customer service

## Looking forward to IPTV

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- IP Video is much more sensitive to transmission impairments than VoIP
- Much more complex video structure makes quality estimation more difficult
- IPTV is usually encrypted, which makes quality estimation even more difficult
- Making IPTV work over existing DSL infrastructure
- Service providers need to prove that IP based services work

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