



**VIAVI**

VIAVI Solutions

Brochure

# **VIAVI** **VoIP Analysis**

Manage, monitor, and maintain VoIP communication with Observer Analyzer

## The Detail and Diagnostics to Solve Today's VoIP Problems

Today's enterprise administrators face a multitude of VoIP management challenges. These challenges begin when first preparing for VoIP implementation, continue throughout VoIP deployment, and persist as VoIP traffic traverses across complex, heterogeneous network links.

Effective management of VoIP call quality requires hard data beyond subjective user assessments. Observer<sup>®</sup> Analyzer's VoIP Expert delivers the essential analysis required to manage, troubleshoot, and maintain the quality of VoIP communications. Regardless of whether you need management-level VoIP summaries or command-level details of a particular call, Analyzer delivers.

"We just deployed VoIP on a remote site where our user expectation was high. But we had a great deal of technical problems—for example the sound would drop out. Observer Analyzer helped us identify problem areas and helped us pinpoint the location to implement Quality of Service measures. With Analyzer, we ultimately improved the performance of VoIP on our network".

- Steve Phelan

The Carphone Warehouse Group

## Benefits

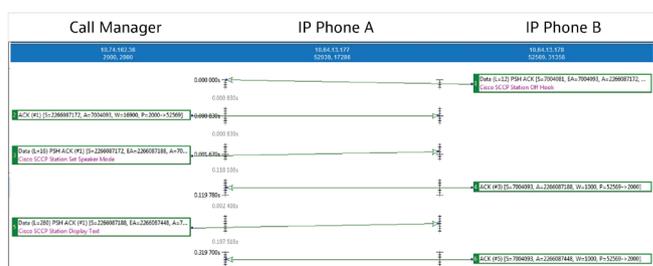
- Focus on real, live traffic issues with over 70 VoIP metrics
- View quick summaries of overall VoIP status
- Drill down to specific connection details for faster troubleshooting
- Immediately identify quality issues with over 50 VoIP Expert events
- Measure call quality individually and in aggregate:



From aggregate-level VoIP summaries...

ID/Stream	Station 1/Port	Station 2/Port	Status
2051020 (Jason Mauk) - 2051021 (Br...)	10.64.13.177	10.64.13.178	■
▶ RTP/PCMU(G.711)	17286	31356	■
▶ RTP/PCMU(G.711)	17286	31356	■
2051020 (Jason Mauk) - 2051021 (Br...)	10.64.13.177	10.74.162.36	■
▶ SCCP	52939	2000	■
▶ SCCP	52939	2000	■
2051021 (Brett Messinger) - 205102...	10.64.13.178	10.74.162.36	■
▶ SCCP	52569	2000	■
▶ SCCP	52569	2000	■

...to individual call detail records...

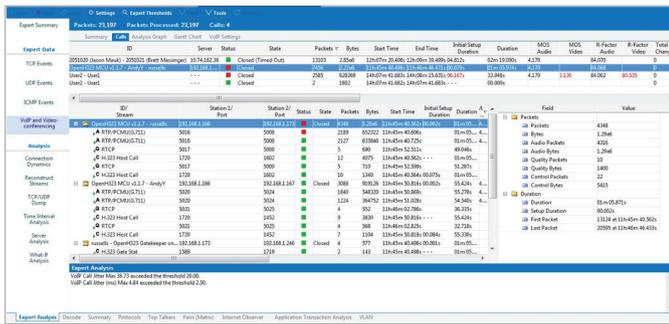


...to connection breakdowns.

## Key Features

### Monitor VoIP Communications In-Depth

Problems with VoIP audio quality are almost always the result of network delay, jitter, and packet loss, or some combination of these issues. Analyzer tracks network factors that affect quality, and reports call quality scores, which measure overall VoIP network health. Monitor aggregate statistics such as traffic summary, total number of VoIP packets, average jitter rates, and overall call quality. Or drill down into call detail records for individual information such as MOS/R-factor scores, QoS prioritization levels, call setup, duration, and teardown.



Source	State	Packets	Bytes	Start Time	End Time	Initial Setup	Duration	MOS	R-Factor
200000 (SIP) [M] 200000 (SIP) [M] 200000 (SIP) [M]	Closed (Timed Out)	15333	2,024,640	11/10/10 20:45:06	11/10/10 20:45:06	11/10/10 20:45:06	0:00:00	4.070	64.070
200000 (SIP) [M] 200000 (SIP) [M] 200000 (SIP) [M]	Closed	2305	626,500	11/10/10 20:45:06	11/10/10 20:45:06	11/10/10 20:45:06	0:00:00	4.270	116.420
200000 (SIP) [M] 200000 (SIP) [M] 200000 (SIP) [M]	Closed	2	1802	11/10/10 20:45:06	11/10/10 20:45:06	11/10/10 20:45:06	0:00:00		0

Call Detail Records

### Maintaining Quality of Service

Analyzer's VoIP Expert recognizes and decodes the IP Type of Service field precedence bit—a mechanism for prioritizing traffic for applications such as VoIP. Quality of Service (QoS) level is therefore reported by call, packet, and protocol. VoIP Expert also shows percentage of VoIP utilization as compared to the rest of network traffic, which is essential for planning network upgrades.

### QoS Standards Recognized

- Default: RFC 1349, 1195, 1123, 791
- OSPF V2: RFC 1248, 1247
- DSCP: RFC 2474

### Quickly Solve VoIP Problems with VoIP Expert

Use Analyzer's Expert Edition help to efficiently find the source of VoIP trouble. Analyzer includes over 50 event-based and threshold-based VoIP Experts.

#### Examples Include:

- Alarms for unacceptable jitter level
- Lost packets
- Alterations in the QoS stream

### Manage VoIP Audio Quality

The Mean Opinion Score (MOS) is an overall quality index that calculates how a typical user would rate audio quality given the current network conditions. Placing an alarm on this score is a good way to proactively manage overall VoIP quality: If MOS starts to fall below 3.5 (out of 5), further investigation might be necessary to determine why VoIP quality is suffering. These factors can be customized depending on the network's specific VoIP equipment and overall network characteristics. For example, VoIP quality on calls made from the factory floor over multiple hops may be scored differently than a call made within a small office setting.

### Detailed Views with in-Depth Call Detail Records

- Addresses
- Call status (open, closed, fail)
- Number of packets, packet bytes, packet loss
- Start time, initial setup duration, duration
- Current jitter, maximum jitter
- MOS, R-factor
- QoS per call
- Number of packets that arrive out of order
- Detailed analysis for packet loss & delay
- Gap density, average gap duration
- Coding/decoding delay

### Quality Score Impairment Factors Include:

- Loudness rating
- Talker Echo rating
- Circuit noise
- Floor noise
- Room noise

### View Conversational Streams in Detail

Analyzer's comprehensive call display details statistics for all of the data streams that comprises a VoIP call, including call setup communications, the actual voice conversation, and call teardown communications. Use this to monitor call ID and stream information including gap density (the rate of low-level packet loss), burst density (the rate of packet loss during bursts of packet loss), and other critical statistics.

### VoIP Expert Includes Call Id and Stream Information such as:

- Number of packets
- Call setup
- Duration
- Teardown
- Lost packets
- Jitter
- Gap density
- Burst density

### Monitor VoWLAN

With Observer's Distributed Network Analysis (O-DNA™) architecture, all of Analyzer's VoIP enhancements are automatically available across multiple topologies. For example, Analyzer's VoIP Expert has the capability to monitor VoIP traffic even over wireless networks.

### Decode and Reconstruct Voice and Video Streams

Analyzer offers complete VoIP and video decodes, including H.323, Session Initiation Protocol (SIP), MGCP, and SCCP (Cisco "skinny"). Reconstruct streams into a .WAV or .AVI file for replay.

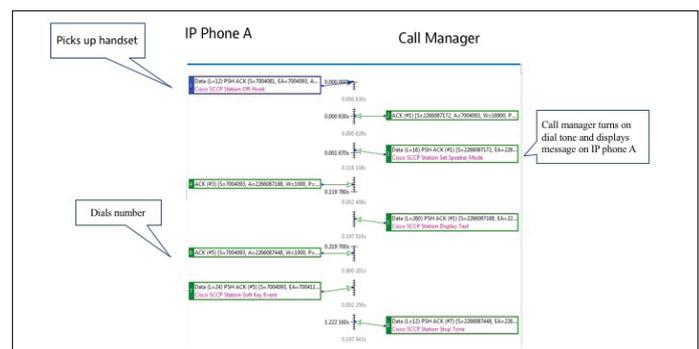
"So far, Observer Analyzer's VoIP capabilities have helped cut CI Travel's phone bill by about 25–30 percent".

- Paul Ingram  
CI Travel

### Troubleshoot Connection Problems

When VoIP users cannot obtain a dial tone, or if there is excessive delay in connecting to the other party's phone, examining a graphical display of how the call is progressing between the parties and the call manager may reveal the source of the communication breakdown.

Analyzer's VoIP Expert provides a Connection Dynamics display—a unique Analyzer feature for quickly identifying latency and delay. Simply right-click on any call or connection stream to determine which party is not responding or which party is responding slowly.



Connection Dynamics

## Compare Voip to Network Performance

Could total network bandwidth affect jitter and delay? Analyzer's VoIP Expert charts jitter in each direction of a VoIP call and compares it to total bandwidth utilization. Use this information to determine if network load is affecting VoIP call quality.

"Our next big project is VoIP, and Observer Analyzer's VoIP Analysis feature is going to assist us with implementation. Before, I wouldn't have known if our systems were prepared for VoIP deployment but now I have the confidence we can go forward. With Analyzer, I'm making more intelligent decisions about the future of our network".

- Mesmak "Mark" Giorgis  
Long Beach Transit



Contact Us **+1 844 GO VIAMI**  
(+1 844 468 4284)

To reach the VIAMI office nearest you,  
visit [viavisolutions.com/contact](https://www.viavisolutions.com/contact)

© 2021 VIAMI Solutions Inc.  
Product specifications and descriptions in this  
document are subject to change without notice.  
voip-br-ec-ae  
30176204 901 0717