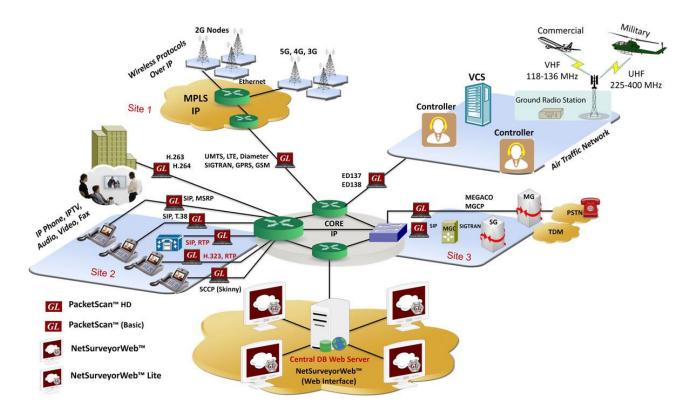
All-IP Signaling and Traffic Analysis RTP/RTCP/Fax (T.38) Analysis

PacketScan[™] All-IP Signaling and Traffic Analysis

(5G/4G/3G/2G/VoIP/RTP, RTCP/ Video Analysis)





What the Software does?

- Captures, segregates, and monitors packets; perform voice quality testing in real-time over VoIP network
- Non-intrusively capture real-time signaling and traffic packets for infinite time and provide call details of per call and aggregated protocol statistics
- Supports monitoring of 5G networks. It captures, segregates, monitors and collects statistics on all calls over N1/N2, N4, N8, N10, N11, N12 and N13 interfaces of the 5G network
- Can be deployed as a Probe for a centralized monitoring system with Oracle database



Applications

Real-time VoIP Traffic Analysis –

- Analyze 5G Calls
- ➤ View RTP, SIP, MSRP, H.323, and MEGACO Calls
- ➤ Trigger on Called and / or Calling Party, Packet Impairments
 - Save .HDL (GL's proprietary format) or .PCAP (Ethereal format) or *.PCAPNG file format
 - Save .WAV time stamped file names
 - Save CDRs
- ➤ MOS R-factor view quality as the call proceeds
- Traffic limited only by hard drive capacity
- > T.38 Fax packets decoding

Network Monitoring Solutions –

- Multiple probes can be used for network monitoring
- CDRs can be exported in a text format to a flat file or a remote computer

Network Monitoring Solutions (Contd.) –

- CDRs can also be exported to an Oracle data base
- ➤ Results can be accessed remotely using NetSurveyorWebTM, a simple web browserbased application
- Air Traffic Monitoring Solution
- Gateway Delay Measurement along with our TDM Protocol Analyzer



Main Features

Comprehensive Analysis Tool	Capture calls in real-time for infinite time non-intrusively
	Detail Signaling, Audio, Video QoS statistics
	 Call flow graph and Pictorial representation of the statistics
	 Inband/Outband Detection, Wave graph, Audio play back, Audio/Video recording
	 Ability to export Call Data Records of completed calls in CSV file format
	Complex Filtering and Search capabilities
	 Option to create multiple aggregate column groups and prioritize the groups as per the requirement to display the summary results efficiently
	 Allows the user to automatically create search/filter criteria from the current screen selection
	 Decode support for multi-layer tunneled traffic - GTP, GRE, VXLAN
	Support export frame summary for tunneled traffic
QOS Parameters and Performance Metrics	E-model (G.107) based MOS/R-Factor scores
	Media Delivery Index for video calls
	 H.263, H.264 codec support
	 Jitter, Delay, and Gap for Audio and Video traffic
	Minimum, maximum, and average Round Trip Delay (RTD)
	 Reports Inband (DTMF & MF) events, Outband events as per RFC 2833 or RFC 4733 events, RTP/RTCP packet count per direction
Triggers and Actions	 Captures calls based on filter criteria and performs set of actions for the completed calls such as recording, sending email, extraction of voice or fax traffic file



Main Features (Contd.)

SIP Registration Details	Registration statistics and trace messages depicted graphically
As a Probe with Central Monitoring System – NetSurveyorWeb™	 PacketScan[™] can send summary fields, frame octets, status, call detail records, along with traffic summary of captured calls to a central database. NetSurveyorWeb[™] displays the data from the database in a simple web-based browser. It features rich graphics, ladder diagrams, CDRs (Call Data Records), custom report and filter configurations
Single Point Analysis System	Enhanced to work with GL's Voice Band Analyzer and Call Data Records applications to provide useful call detail records for further analysis using built-in tool in Excel®.
Utilities	 Provides HDL File Conversion utility to convert ethereal format file (*.PCAP, *.CAP, and *.PCAPNG) to GL's file format (*.HDL) and vice-versa Includes Excel® Addins to import CDRs into Excel® to analyze using Pivot Table, and Pivot Charts.



Supported Protocols

- Session Initiation Protocol (SIP) RFC 3261
- Media Gateway Control Protocol (MGCP) RFC 2705/3435 (3991)
- Media Gateway Control (MEGACO) RFC 3525 and 3015
- Message Session Relay Protocol (MSRP)
- H.323
- RTP/RTCP
- SCTP RFC 2960
- Connection Oriented Transport Protocol (COTP, ISO 8073)

- 5G N1N2, N4, N8, N12, N13
- SCCP (Skinny)
- SS7 SIGTRAN
- ISDN-SIGTRAN
- GSM A over IP
- GPRS over IP
- UMTS over IP
- LTE
- Diameter

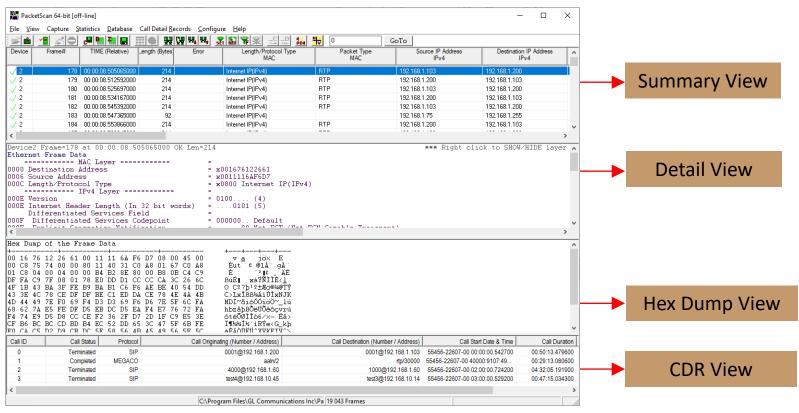


Supported Codecs

- G.711 (mu-Law and A-Law), G.711 Application II (A-law and μ-law with VAD)
- G726 (40, 32, 24, 16kbps)
- GSM (13.2kbps), GSM EFR (12.2 kbit/s), GSM HR
- G729, G729B (8kbps)
- G.722, G.722.1
- ILBC_15_2 (for 20 msec), ILBC_13_33 (for 30 msec)
- SPEEX (Narrow band and Wideband)
- SMV* (Modes 0, 1, 2 and 3)
- Video codecs include H263++ CIF 190, 350, 512 kbps, QCIF 64, 80, 128 kbps, and H264 codec offers video compression
- Other optional codec include (must be purchased with additional license)
 - ➤ AMR (Narrow band and Wideband)
 - > EVRC, EVRC0 (Rates 1/8, ½ and 1)
 - > EVRCB, EVRCB0 (Rates 1/8, ½ and 1); EVRC-C
 - > Opus and EVS (Narrow Band, Wideband, Super Wideband, Full Band)



PacketScan[™] Analyzer View

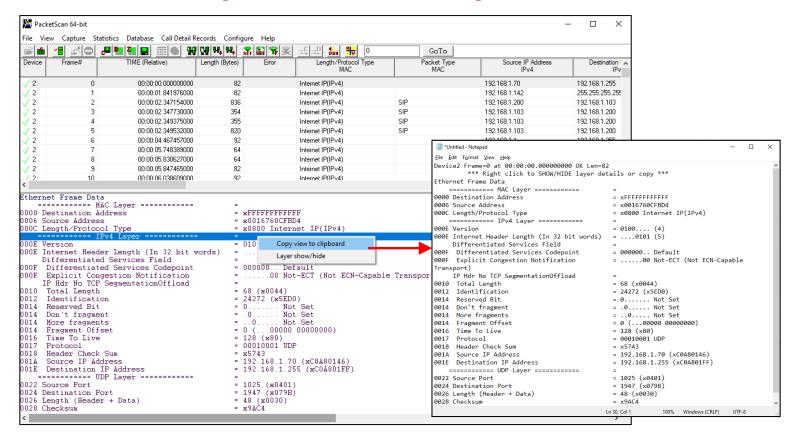


- Default panes summary, detail, and hex dump of the frame data views
- Optional panes statistics and call trace views



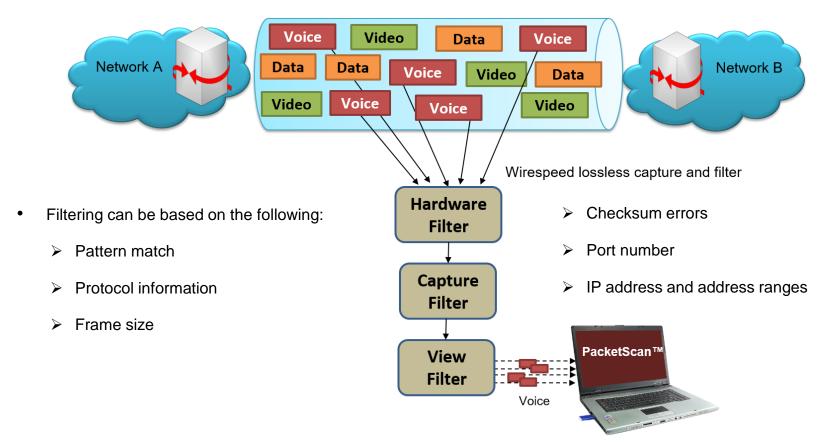
Layer Copy/Show/Hide Options

 Provides option to copy the entire layer decode information to the clipboard





WireSpeed Filtering



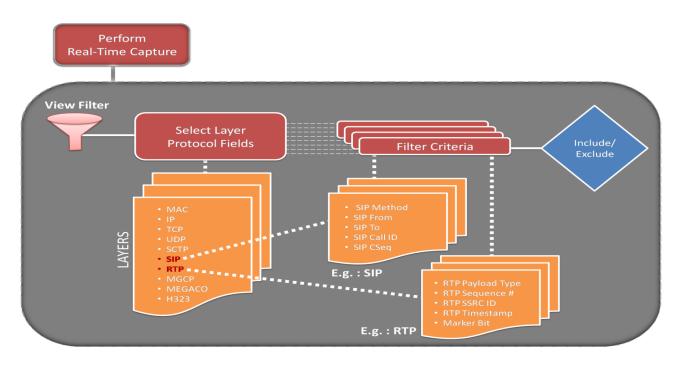


3 Stages of Filtering

- HardwareFilter (HWF) "Special NIC" with hardware filtering very fast
- Capture Filter (CF) Powerful software filtering but a little slower
- View Filter (VF) applies on the captured frames to filter only frames of interest
- PacketScan™ HD captured files to/from WireShark
- PacketScan™ HD PDA for detailed voice, fax, and video analysis



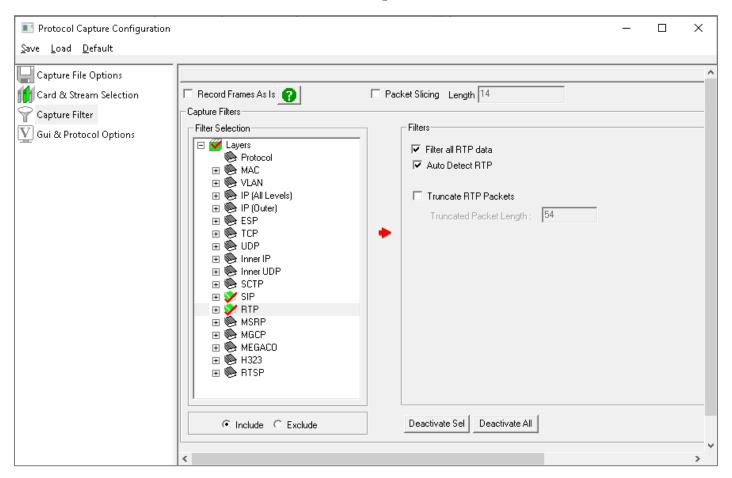
Real-time and Offline Filters



• Filtering and search capability isolates required frames from original frames in real-time / offline based on parameters set in Data Link layer, MAC layer, IP, TCP/UDP and more

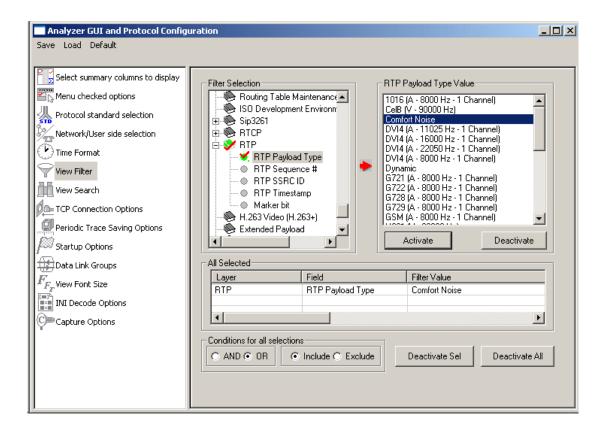


Real-time Capture Filter



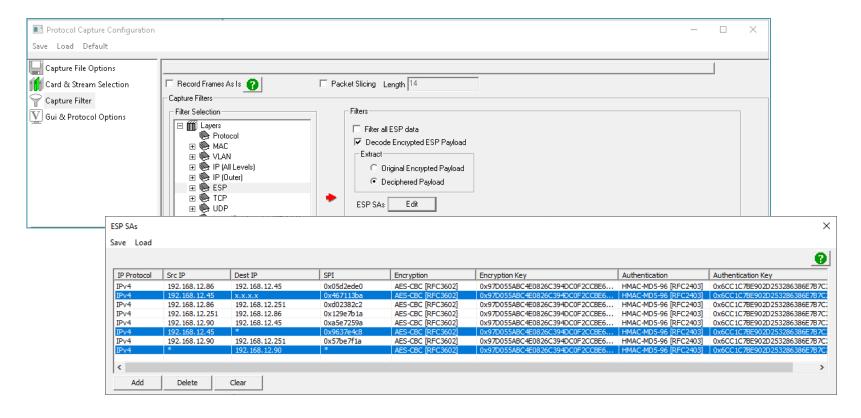


Display Filter (Offline)





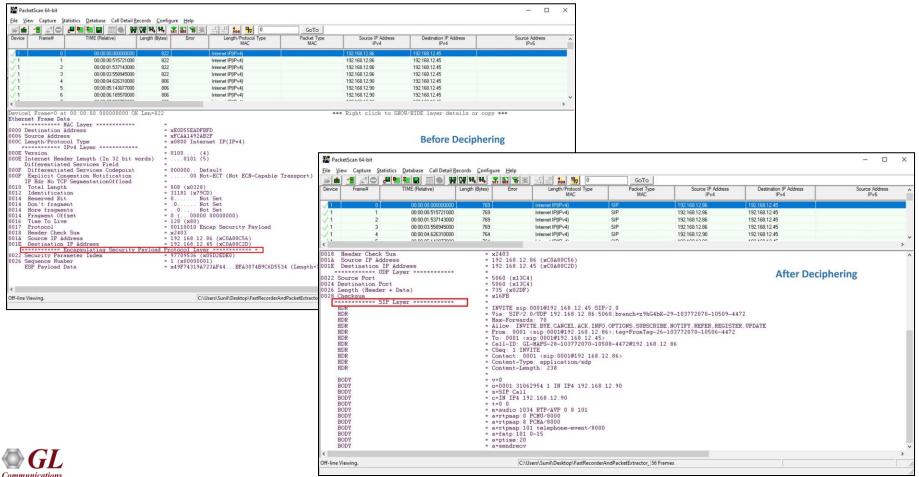
Encapsulated Security Payload (ESP) Deciphering





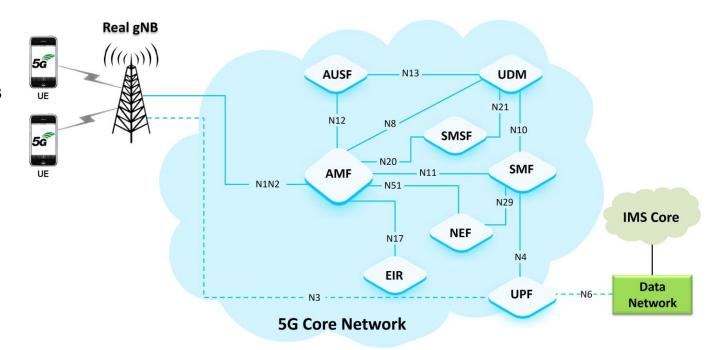
ESP filter is used to provide **ESP SAs value** to decrypt ESP packets

Comparison of Before and After ESP Deciphering



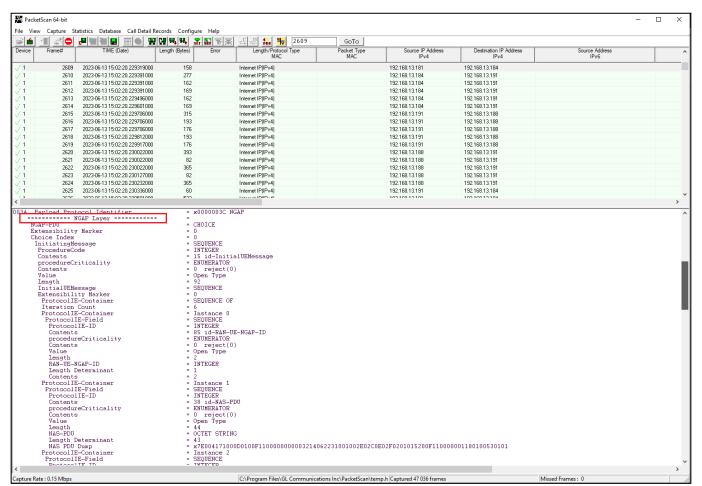
5G Analyzer

- Captures, segregates, monitors and collects statistics on all calls over N1/N2, N4, N8, N12 and N13 interfaces of the 5G network
- Provides VoNR call statistics such as caller, callee, MOS scores, discarded packets and voice storage



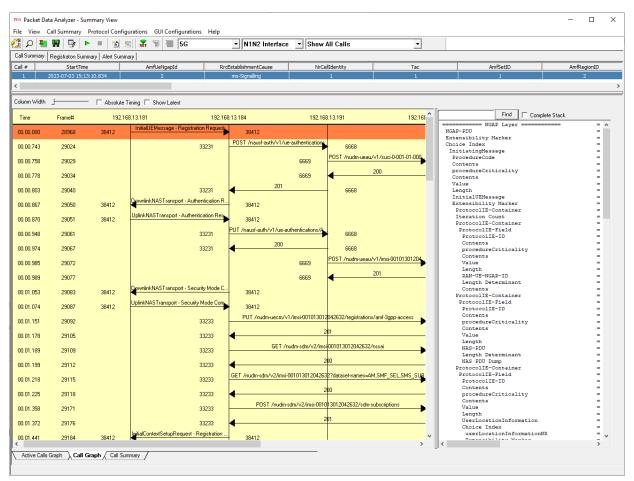


Decode View - 5G NGAP Layer





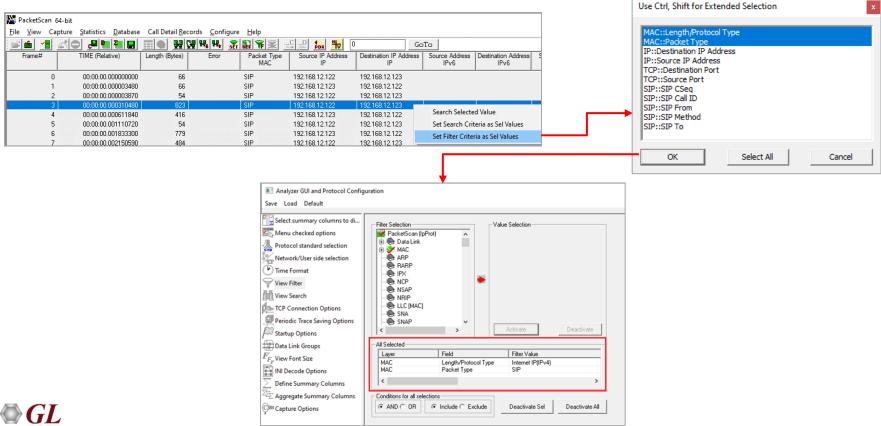
Call Graph – 5G N1N2





Filter Criteria From Screen Selection

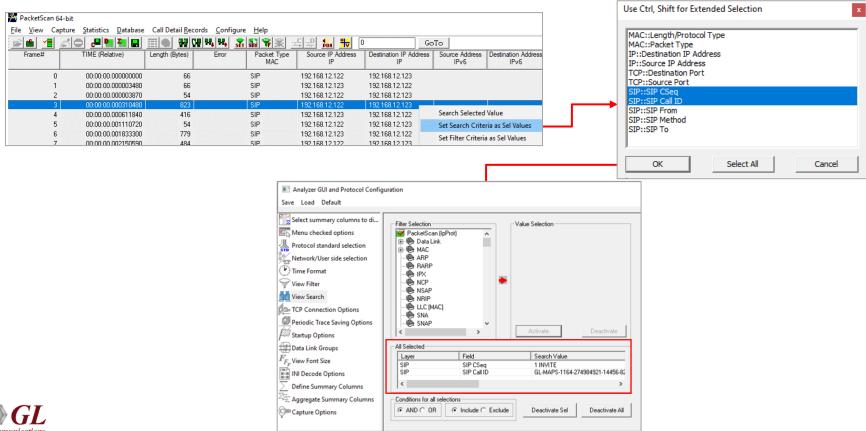
Allows the user to automatically create filter criteria from the current screen selection





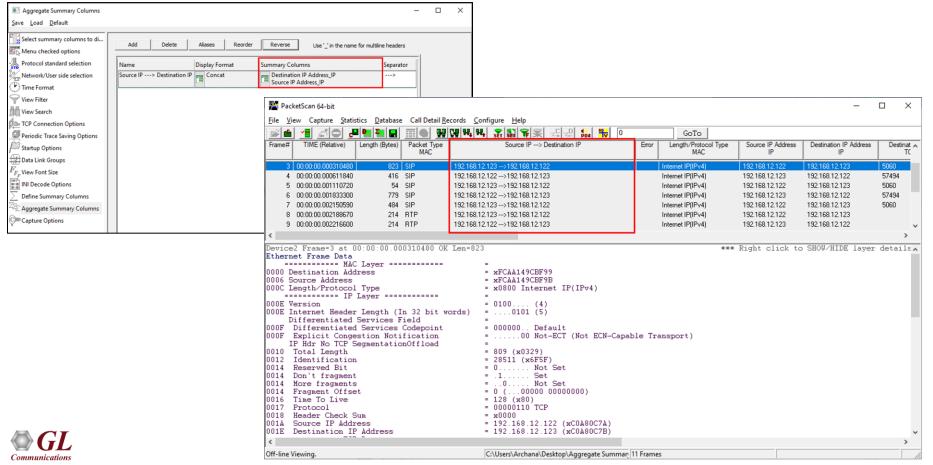
Search Criteria From Screen Selection

Allows the user to automatically create search criteria from the current screen selection



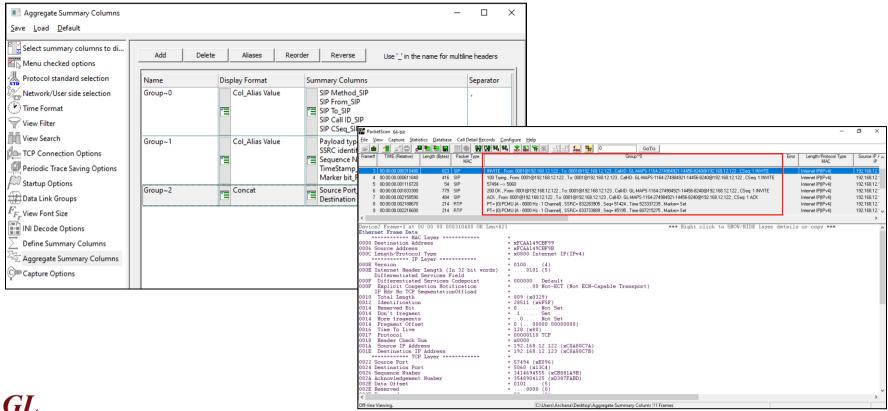


Aggregate Summary Columns



Aggregate Summary Column Group

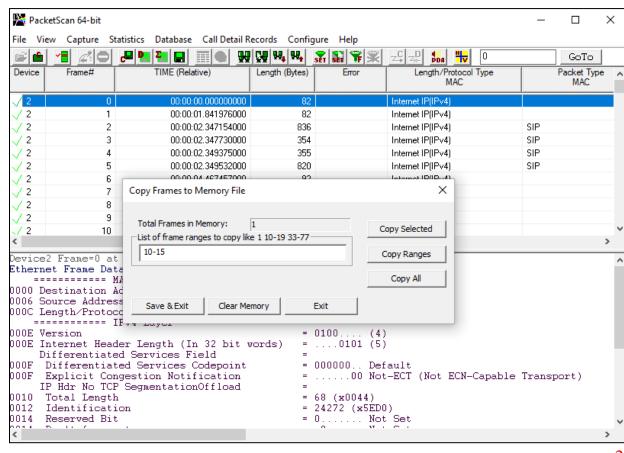
 The user can create multiple aggregate column groups and prioritize the groups as per the requirement to display the summary results efficiently





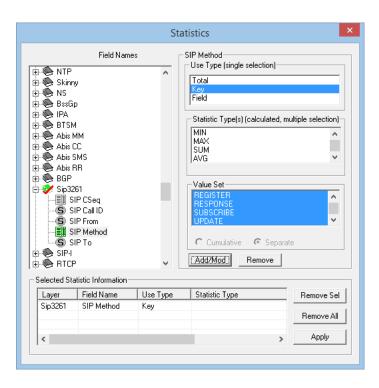
Copy Frames to Memory File

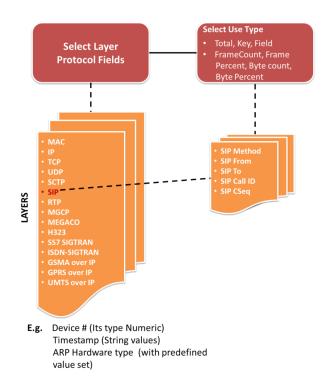
 The user can select and copy the frames (frame range) to the desired location





Statistics



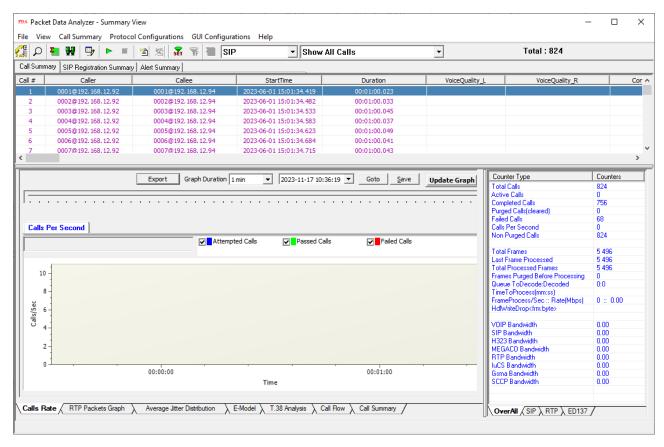


 Various statistics can be obtained to study the performance and trend in the VoIP network, based on protocol fields and different parameters



Packet Data Analysis

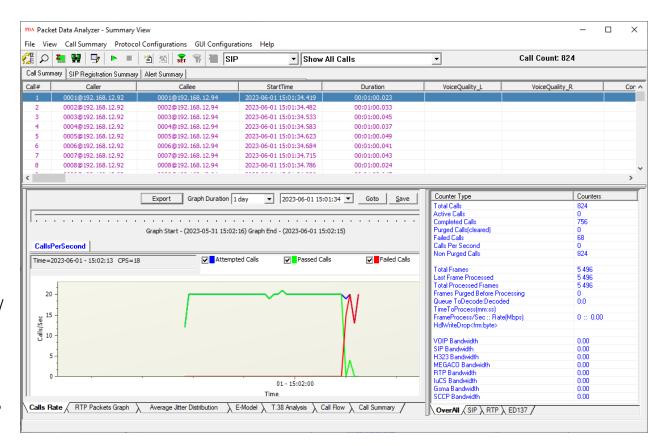
- Packet Data Analysis in PacketScan[™] displays the following views
 - > Summary view
 - Detail view
 - Registration summary view





PDA Summary View

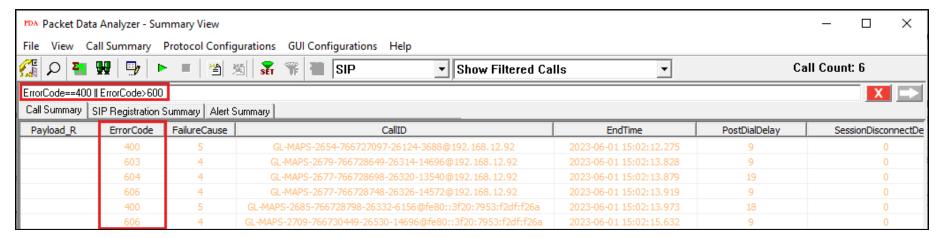
- Summary View displays -
 - Summary of data transmission in each direction including calling number, called number, call id, start time, duration, missing packets, etc.
 - Includes separate statistical counts on total packets, calls, failed calls, captured frames, etc., for SIP, H323, MEGACO, and RTP based calls
 - Provides various graphs to view active calls, average jitter distribution, E-model based measurements for R-factor / MOS/ Packet discarded, RTP packets, T.38 fax analysis, and call signaling, Gap, Jitter, Gap/Jitter Distribution, Wave and Spectral Display for media stream analysis, VoIP calls and more





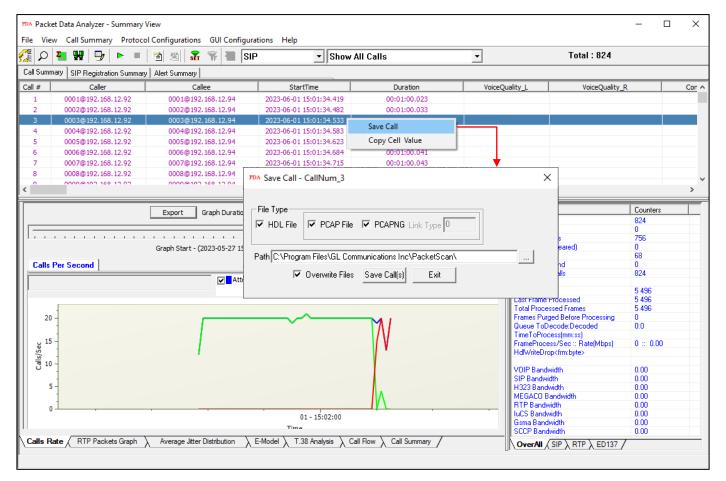
Displaying Filtered Calls using Filter Expressions

- Filter CDRs (Call Detail Records) based on parameters such as caller, time, message count, etc.
- The expression supports the following mathematical operators: ==, <=, >=, !=, <, >, &&, ||
- For example, the filter expression "ErrorCode==400||ErrorCode>600" will display calls with ErrorCode equal to 400 and calls with ErrorCode greater than 600



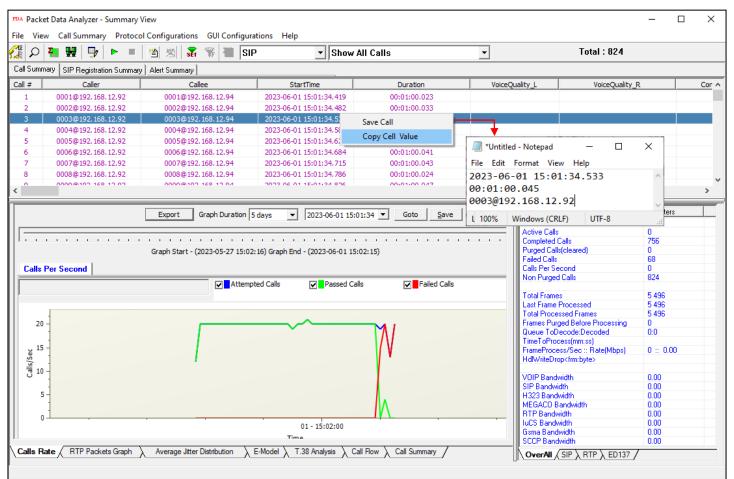


Save Call in *.hdl, *.pcap, and *.pcapng Formats





Copy Cell Value to Clipboard

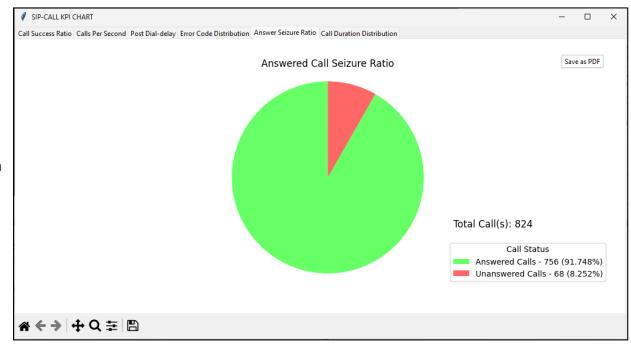




Key Performance Indicators (KPIs) Report for SIP Calls

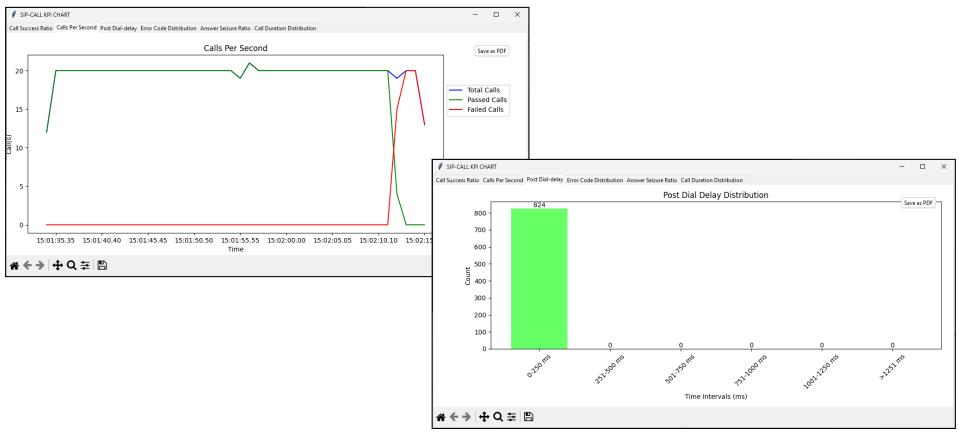
The SIP Call Summary KPI Report includes KPIs for the following fields:

- Call Success Ratio: Displays graph for "Successful" and "Unsuccessful Calls," including counts and percentages (%)
- Calls Per Second: Shows graph "Total," "Passed," and "Failed Calls per second."
- Post Dial Delay: Shows delay counts in milliseconds (0-250ms, 251-500ms, etc.)
- Error Code Distribution: Lists Top 10
 Call Failure Causes with counts and percentages (%)
- Answer Seizure Ratio: Shows "Answered" and "Unanswered Calls," with counts and percentages (%)
- Call Duration Distribution: Provides call counts for different durations (0-1 sec, 1-10 sec, etc.)



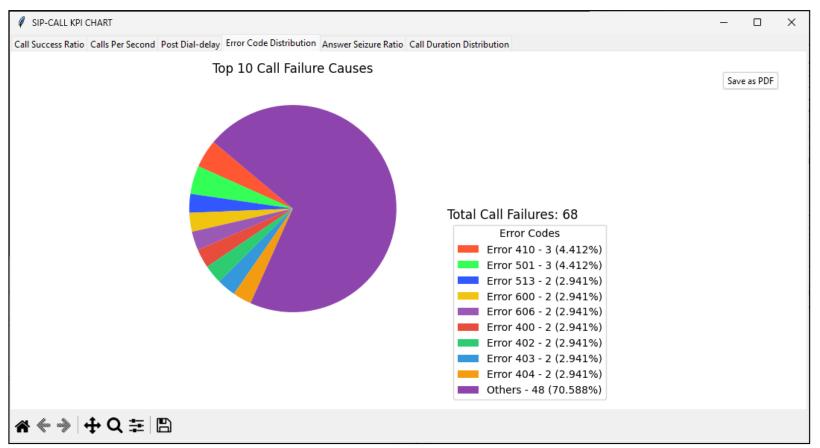


Calls Per Second and Post Delay KPIs



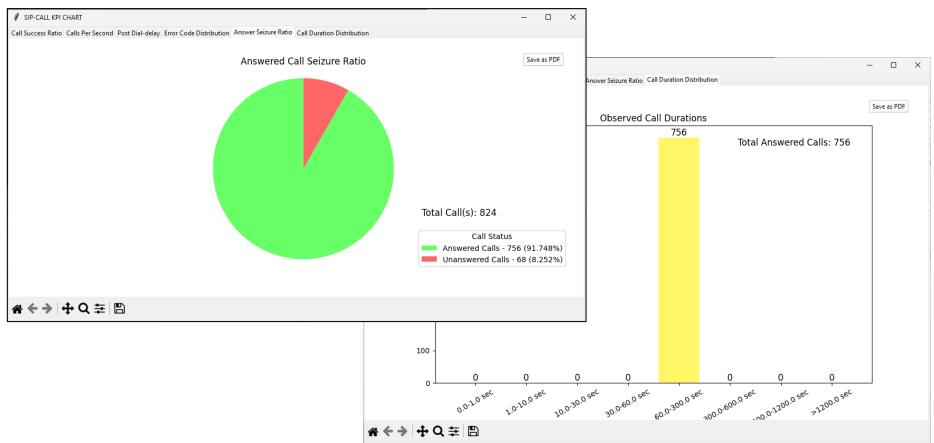


Error Code Distribution KPI





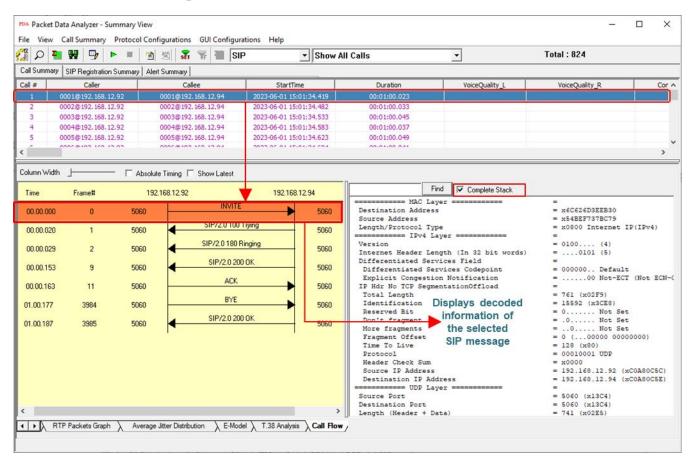
Answer Seizure Ratio and Call Duration Distribution KPIs





Call Graph – SIP Call

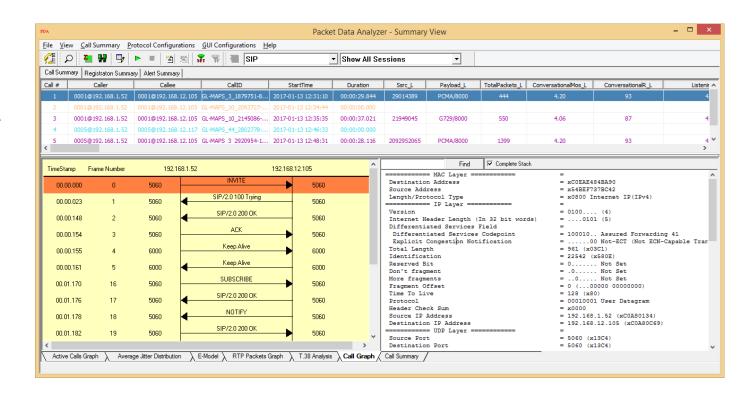
- Displays the message sequences of captured VoIP calls
- Decodes of the selected SIP message is displayed on the right pane
- The Complete Stack option enables the user to view the full call details for the selected message on the ladder diagram





Call Graph – SIP ED-137

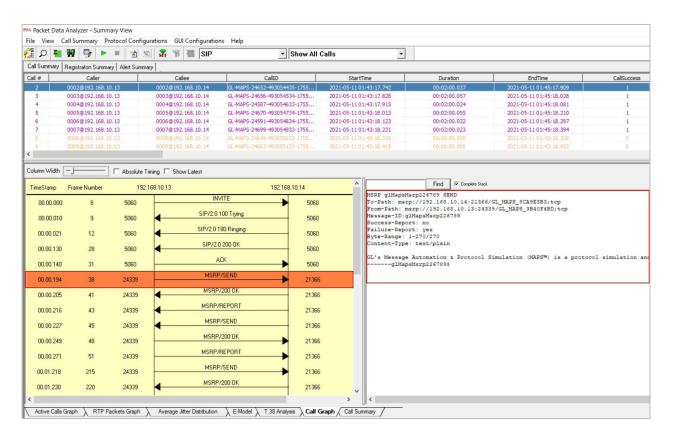
- Displays the message sequences of SIP ED-137 call
- Decodes of the selected
 SIP message is displayed
 on the right pane





Call Graph – MSRP Call

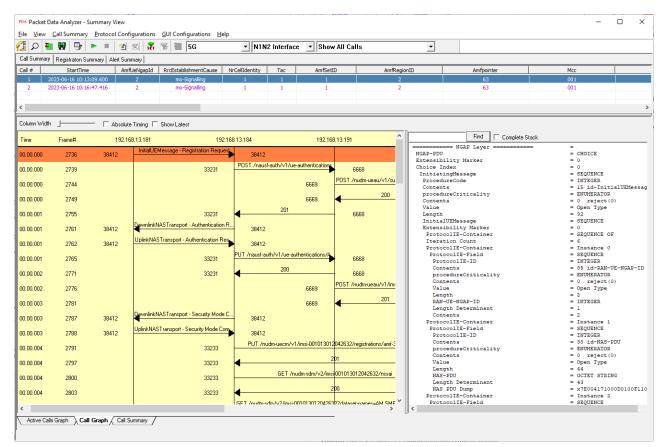
Decodes of the selected MSRP message is displayed on the right pane





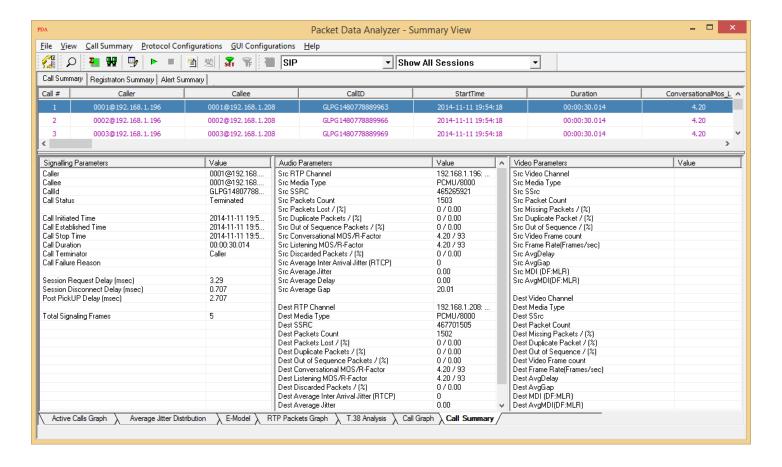
Call Graph – 5G N1N2 Call

Decodes of the selected 5G N1N2 message is displayed on the right pane





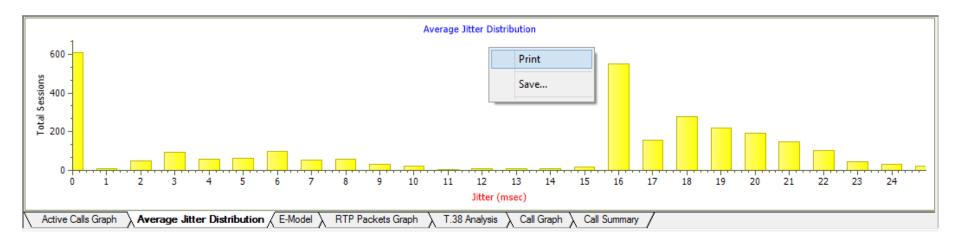
Signaling / Audio/ Video QoS Parameters





Average Jitter Distribution Graph

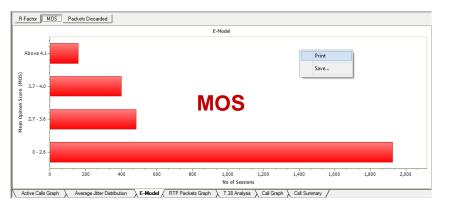
Distribution of the Average Jitter values across Total Sessions

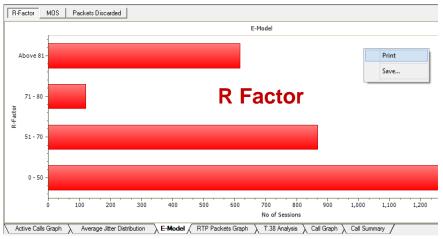




MOS Graph and R-Factor Graph

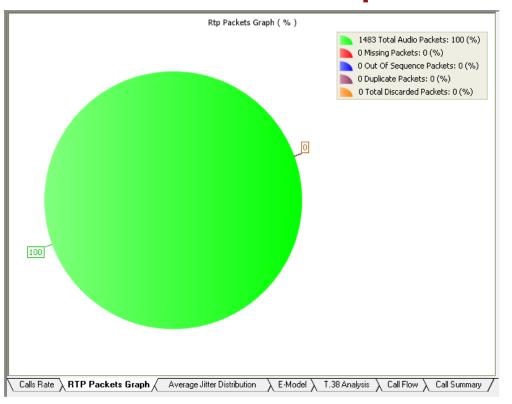
- E-Model graph provides R-factor, MOS, and packets discarded against number of sessions.
 All these three graphs show statistics of terminated calls
 - R-Factor A bar graph that plots R-Factor across No of Sessions
 - ➤ MOS A bar graph that plots Mean Opinion Score across No. of Sessions
 - Packets Discarded A bar graph that plots Packets Discarded across No. of Sessions







RTP Packets Graph

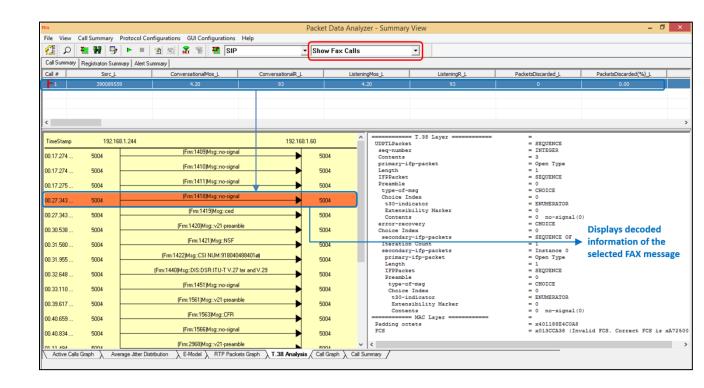


RTP Packets graph plots and compares out of ordered packets, missing packets and duplicate packets against Total Audio Packets



T.38 Analysis - Fax over IP

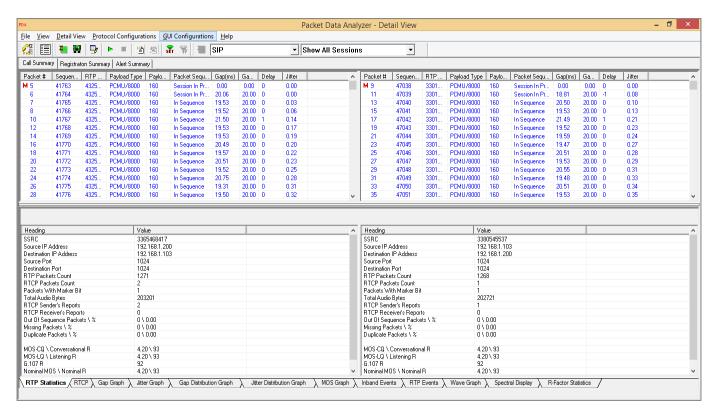
- Supports capturing and decoding of Fax (T.38 data)
 calls over VolP
- Decodes of selected FAX message is displayed on the right pane
- Captured fax calls by
 PacketScan[™] can also be
 analyzed using GLInsight[™] by
 saving the fax calls directly in
 (*.PCAP) Ethereal file format





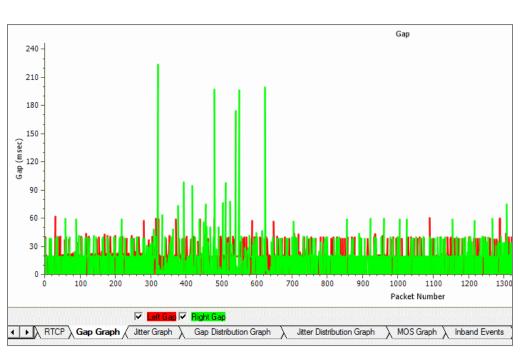
Call Detail View

- Provides a detail look at the two (or one) RTP sessions that are part of a single call
- Left and right panes accommodate the two sessions

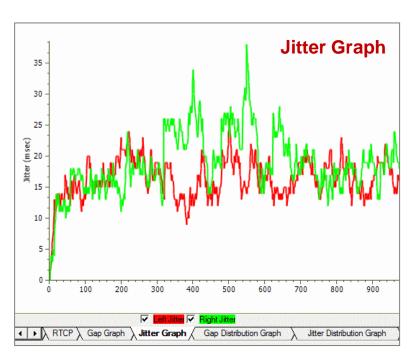




Gap and Jitter Graph



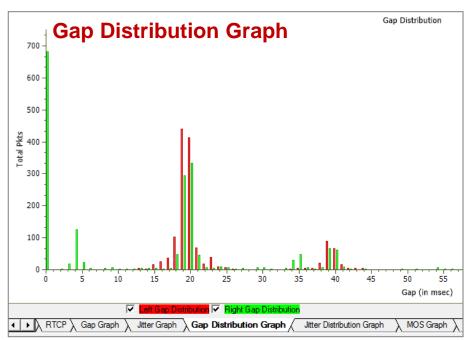
 Gap graph plots the Gap (in milliseconds) versus the packet number

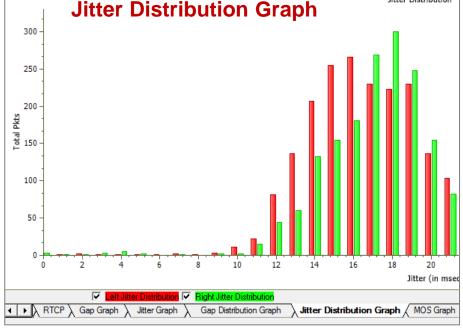


 Jitter graph plots the Jitter versus the packet number



Gap and Jitter Distribution Graph





 Number of packets with a particular value of gap is plotted against the (gap) value

 Number of packets with a particular value of jitter is plotted against the jitter value



Jitter Distribution

MOS Graph



MOS Graph plots Mean Opinion Score values throughout the duration of the call



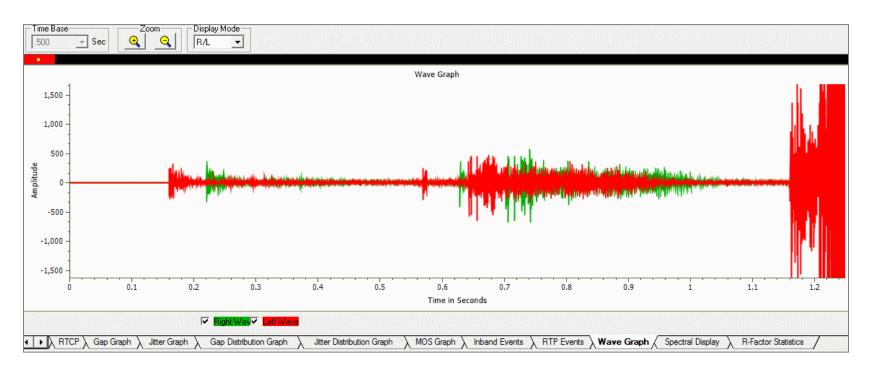
Inband and Outband (RTP) Events

TimeStamp	Туре	Event	On(ms)	Power(dBm)	Freq1(Hz)/Pow	Freq2	2(Hz)/Pow 🔼				
00:00:00.000	IDLE		45470	0.00							
00:00:45.470	DTMF	1	80	-1.89	6987-6.01	1210	0/-4.01				
00:00:45.550	IDLE		80	0.00							
00:00:45.630	DTMF	2	80	-1.87	6987-6.00	1337	7/-3.99				
00:00:45.710	IDLE	_	80	0.00							
00:00:45.790	DTMF	3	80	-1.85	6987-5.98	1479	2/200				
00:00:45.870	IDLE		80	0.00	000r 0.00		TimeStamp	Event	Volume (-dBm)	Duration (ms)	
00:00:45.950	DTMF	4	80	-1.86	7717-5.98		12:09:02.652	DTMF 1	6	80	
		4			77175330		12:09:02.812 12:09:02.971	DTMF 2 DTMF 3	6 6	80	
00:00:46.030	IDLE	_	80	0.00	774 1 5 00		12:09:02:971	DTMF 4	6	80	
00:00:46.110	DTMF	5	80	-1.86	7717-5.98		12:09:03:132	DTMF 5	6	80	
00:00:46.190	IDLE		80	0.00			12:09:03.452	DTMF 6	6	80	
00:00:46.270	DTMF	6	80	-1.87	7717-5.99		12:09:03.612	DTMF 7	6	80	
		0			1117-0.00		12:09:03.772	DTMF 8	6	80	
00:00:46.350	IDLE		80	0.00			12:09:03.931	DTMF 9	6	80	-
00:00:46.430	DTMF	7	80	-1.86	853/-5.98	1 1 1 1 1 1 1	12:09:04.092	DTMF 0	6	80	
00:00:46.509	IDLE		80	0.00			12:09:04:252	DTMFA	6	80	
00:00:46.590	DTMF	8	80	-1.89	8537-6.01		12:09:04.412	DTMF B	6	80	
		0			000/-0.01	1.3	12:09:04.572	DTMFC	6	80	
00:00:46.670	IDLE		80	0.00			12:09:04.732	DTMF D	6	80	
	/ DID			\ .	. 10: 1 1 0	<u> </u>	12:09:04.891	DTMF #	6	80	
iband Events	Y RIPE	Events λ	. Wave G	iraph 🚶 Spec	ctral Display 🔪 🛭 R	-Fact	12:09:36.324	MF 1	6	80	
					,		•				Ŀſ
							nband Events \(\rangle \) F	TP Events / Wave Gr	aph 🔪 Spectral Display	λ R-Factor Statistics	7

- Inband Event tab displays Inband DTMF and MF digits as they are received on selected RTP stream
- Outband (RTP) Events tab displays all Outband RTP events defined in RFC 2833 or RFC 4733



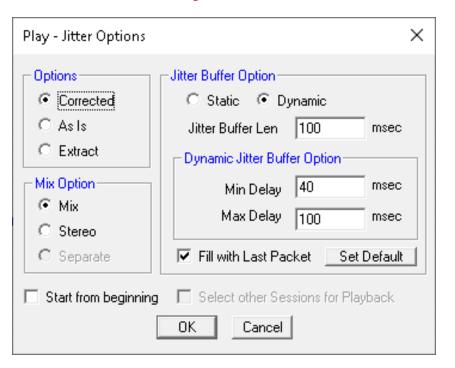
Wave and Spectral Graphs



- Wave graph Displays the amplitude of the incoming signal in a selected call as a function of time
- Spectral Display Displays the power of incoming signal while the capturing is going on as a function of frequency



Play Audio

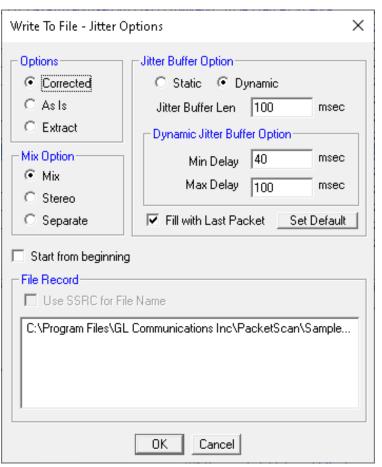


- Plays the RTP streams of a call to the PC speaker using a soundcard
- Provides a host of options such as jitter buffer settings, audio mixing, and so on to play a live call in real-time or play captured voice files



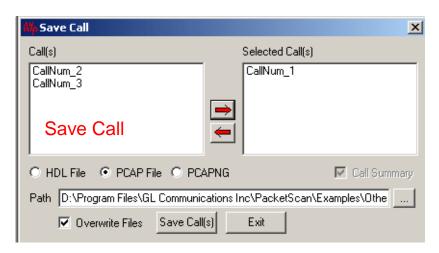
Write to File

- Various options are provided to save captured calls
- Use the files with voice quality analysis software to calculate the mean opinion score of the call
- Records the RTP stream to a file in *.wav format





Save Call and Extract Fax Image



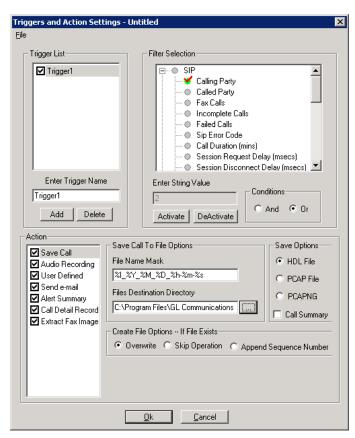
- Saves a particular call in either GL's proprietary HDL file format or Ethereal PCAP file format or PCAPNG file format
- Saves the Call Summary details including signaling and audio / fax/
 video parameters for a particular call in *.rtf file
- Helps in getting data from real-time traffic locations to the lab for detailed analysis



Extracts Fax image in the TIFF format from the selected fax call



Trigger and Action Settings

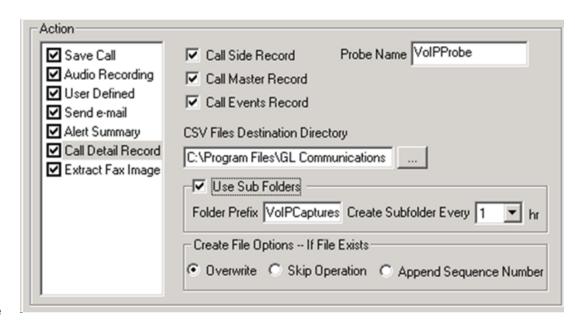


- Set the triggers and actions criteria to filter calls and perform additional actions on the completed calls
- Trigger actions on certain SIP, RTP, MEGACO, and H.323 parameters
- Triggering factors includes calling number, called number, incomplete calls, fax calls, call duration, MOS factor, sip error code, average jitter, and more
- Actions include
 - ➤ Saving call to a file -*.hdl, *.pcap, or *.pcapng
 - > Recording audio to a file
 - ➤ Sending an email alert
 - Generates alerts when particular vital parameters go beyond a specified value
 - Outputs call detail records as CSV
 - Extract Fax in Tiff format



Call Detail Records (CSV)

- Creates three types of Comma Separated Value (CSV) files such as Call Side Record, Call Master Record, and Call Events Record
 - ➤ Call Side Record: It is a record concerning each party participating in the call. For example: Probe ID, Call ID, Side, Address, File Name, SSRC, Codec, Total Packets, and so on
 - Call Master Record: It contains fields concerning the call, For example: Probe ID, CALL ID, Side 1, Side 2, Protocol name, Start & Released dated and time, and so on
 - ➤ Call Event Record: It gives an event-byevent account of the call. For example: Probe ID, Call ID, Side, Class ID, Start, Duration, Source IP address, Destination IP Address, and so on
- Use Sub Folders option to automatically create the subfolders after some time duration





Alert Summary

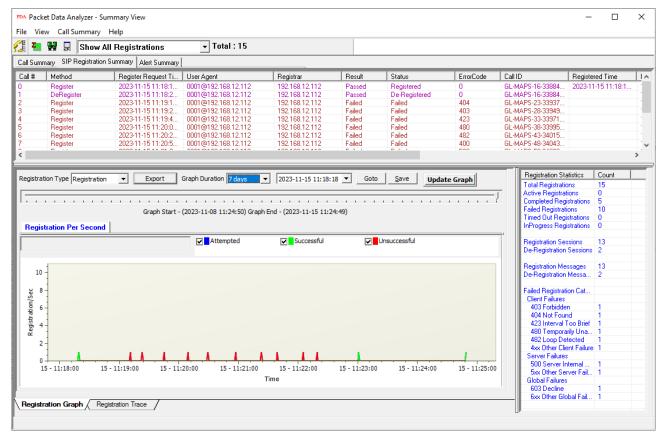
Call Su	ımmary Re	egistraton Summary Alert Sumr	mary					
Call#	Protocol	Message	Туре	Threshold	Value	Caller	Callee	Callid
1	SIP	mos value between 3 to 4	Warning	2.00-4.00	3.57	0005@192.168.1.236	0005@192.168.1.234	GLPG143457205760
2	SIP	mos value between 3 to 4	Warning	2.00-4.00	3.39	0006@192.168.1.236	0006@192.168.1.234	GLPG143617205763
3	SIP	mos value between 3 to 4	Warning	2.00-4.00	2.77	0008@192.168.1.236	0008@192.168.1.234	GLPG143617205769
3	SIP	mos value between 1 to 2.5	Critical	1.00-2.50	2.36	0008@192.168.1.236	0008@192.168.1.234	GLPG143617205769
4	SIP	mos value between 3 to 4	Warning	2.00-4.00	3.48	0009@192.168.1.236	0009@192.168.1.234	GLPG143617205772
5	SIP	mos value between 3 to 4	Warning	2.00-4.00	3.30	0011@192.168.1.236	0011@192.168.1.234	GLPG143777205778
6	SIP	mos value between 3 to 4	Warning	2.00-4.00	2.77	0012@192.168.1.236	0012@192.168.1.234	GLPG143927205781
6	SIP	mos value between 1 to 2.5	Critical	1.00-2.50	2.31	0012@192.168.1.236	0012@192.168.1.234	GLPG143927205781
7	SIP	mos value between 3 to 4	Warning	2.00-4.00	2.27	0001@192.168.1.231	0001@192.168.1.237	GLPG13407127763982
7	SIP	mos value between 1 to 2.5	Critical	1.00-2.50	2.27	0001@192.168.1.231	0001@192.168.1.237	GLPG13407127763982
8	SIP	mos value between 1 to 2.5	Critical	1.00-2.50	1.47	0002@192.168.1.231	0002@192.168.1.237	GLPG13417127763987
9	SIP	mos value between 1 to 2.5	Critical	1.00-2.50	1.04	0003@192.168.1.231	0003@192.168.1.237	GLPG13425567763992

- Generates alerts when vital parameters go beyond a specified value
- Provides an active list of the alerts for the events in a tabular column
- Displays the summary of call#, user-defined message, threshold value, actual value for which the alert occurred, callee, caller, and callid



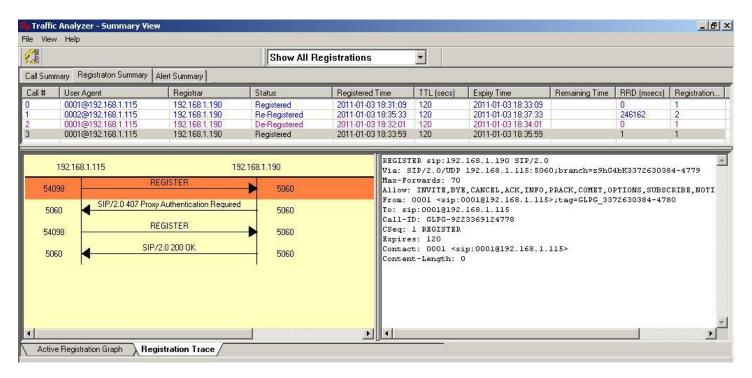
- Displays the SIP registration information in a tabular format which includes user agent, registrar, registered time, status, and so on for each user agent
- Displays the active registration graph of the entire registration summary
- Provides the trace display of each registration

Registration Summary





Registration Trace

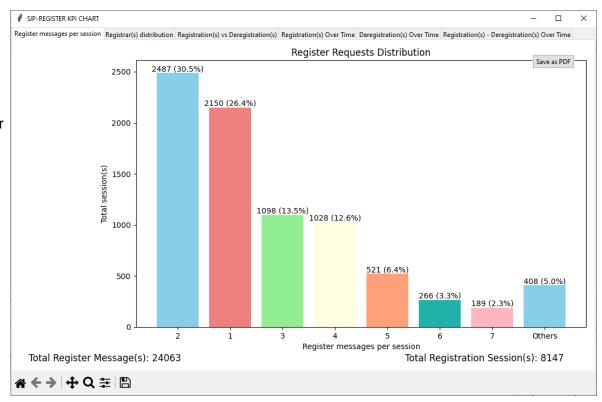


- Displays the message sequence of registered calls
- Message sequence pictorially displays the messages exchanged for a particular scenario between a user agent and the registrar



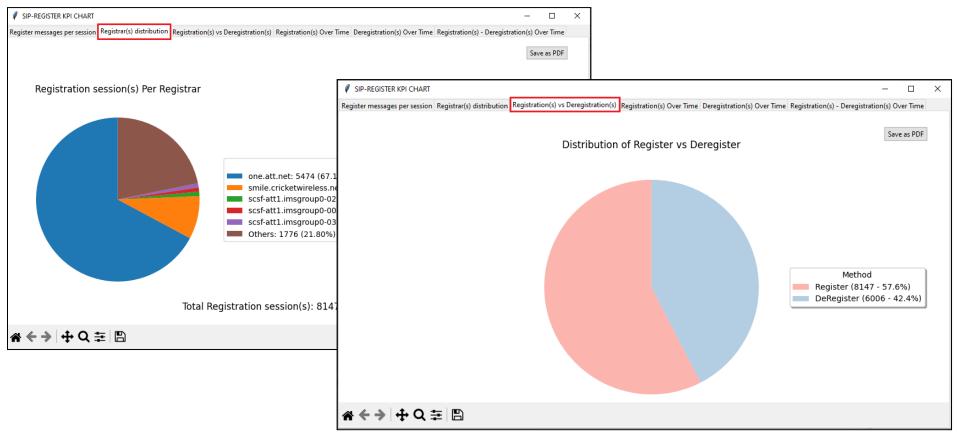
KPI Report (Registration)

- The SIP Registration Summary KPI Report includes KPIs for the following:
- Register Messages per Session: Shows a graph for the distribution of Register Requests
- Registrar(s) Distribution: Displays a graph for the number of Registration sessions per Registrar
- Registration(s) vs Deregistration(s): Illustrates a graph comparing the distribution of Register and Deregister counts with percentages (%)
- Registration(s) Over Time: Show the graphs for "Successful," "Failed," and "Total Attempts" per second
- Deregistration(s) Over Time: Displays a graph for "Successful" and "Total Attempts" per second
- Registration(s) Deregistration(s) Over Time:
 Shows a graph for overall "Register & Deregister attempts," "Register & Deregister passed," and "Register & Deregister failed" attempts per second



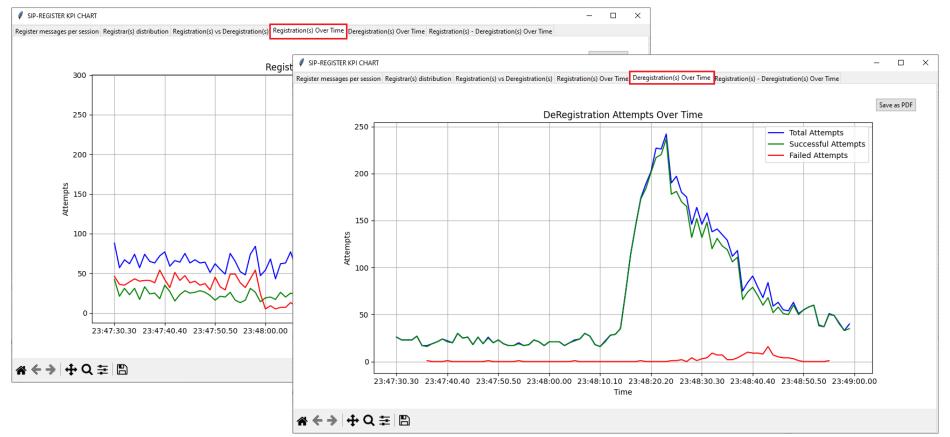


Registrar(s) Distribution, Registration vs Deregistration KPIs



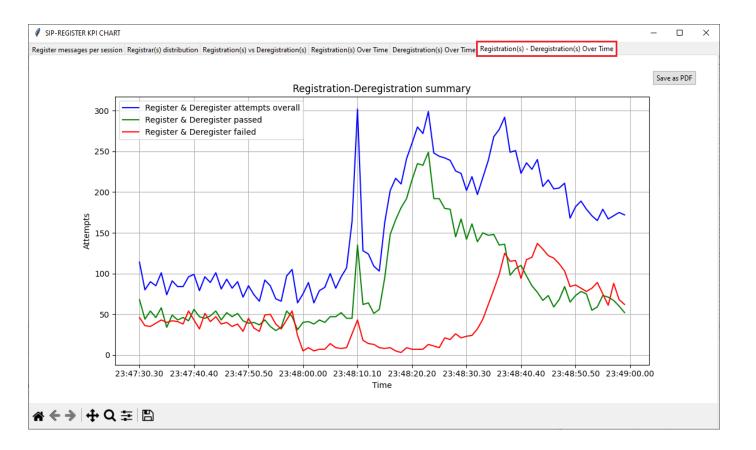


Registration(s) over Time, Deregistration over Time KPIs





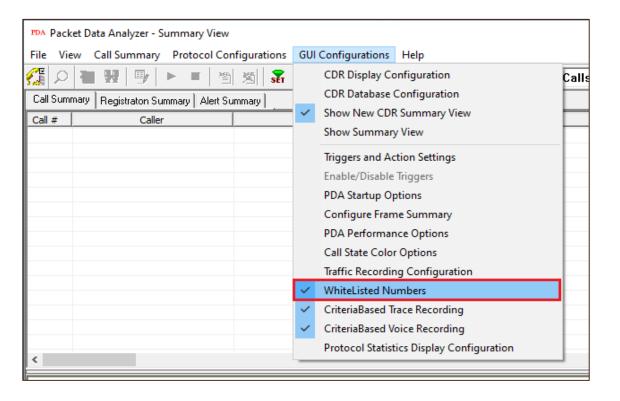
Registration(s)-Deregistration(s) over Time KPI





Whitelist Configuration

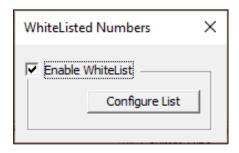
• On the **PDA** main window, click on **GUI Configurations** → **WhiteListed Numbers** to configure Whitelist number

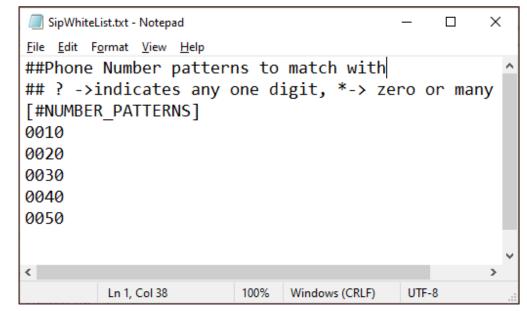




Whitelist Configuration (Contd.)

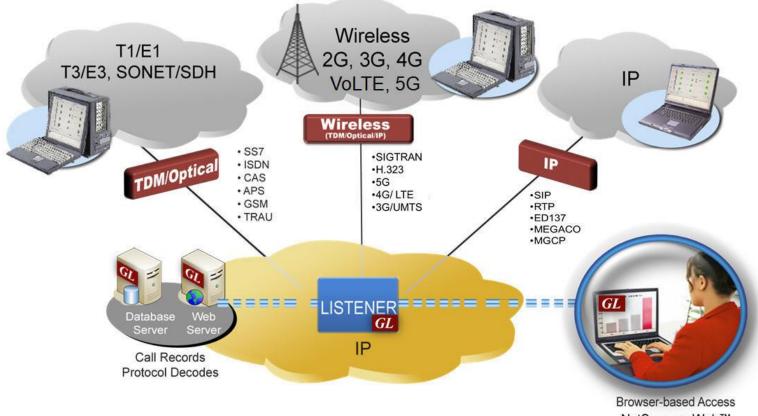
- Check the Enable WhiteList option and click on Configure List. This will invoke SIPWhiteList.txt in the Notepad application
- Enter the SIP Caller or Callee Number
- The following numbers should be added in the SipWhiteList.txt file
 - 0010
 - 0020
 - 0030
 - 0040
 - 0050
- Save and Close the file





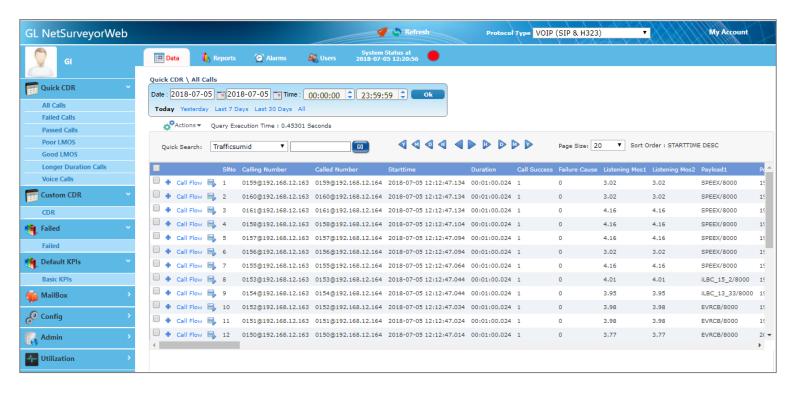


NetSurveyorWeb™ - Network Surveillance System





NetSurveyorWeb™ Main GUI



- Multiple PacketScan[™] probes can be used for network monitoring, with call detail reports exported to a central data base
- Results can be accessed remotely using NetSurveyorWeb™, a simple web browser based application



Thank you

