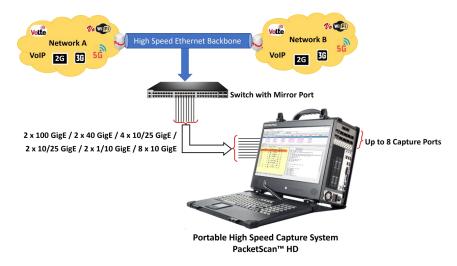
# High Density Ethernet Monitoring Appliance – PacketScan™ HD

(1G, 10G, 25G, 40/100G)



### **Overview**

PacketScan<sup>™</sup> HD is a high density Ethernet monitoring appliance with specialized network interface cards, large storage capacity and protocol analysis software. Customers can choose the specific Ethernet data rate for the network interface cards including 4 x 1 GigE, 2 x 10 GigE, 2 x 40 GigE, and 2 x 40 / 2 x 100 GigE variations. Capture and analyze high speed Ethernet traffic over 1 Gbps, 10 Gbps, 40 Gbps and 100 Gbps networks. Almost all VoIP and Wireless protocols over IP transport layer can be captured and decoded for troubleshooting network problems. PacketScan<sup>™</sup> HD appliance is also available in three variants.

Part Number	PKV120 (Rack system)/ PKV120P (Portable system)	PKV122 (Rack system)/ PKV122P (Portable system)	PKV124 (Rack system)/PKV124P (Portable system)					
Processor	Single Processor	Dual Processor Xeon /Single Processor	Dual Processor Xeon/Single Processor					
RAM	32 GB	32 GB	128 GB					
Storage		500 GB NVME SSD, customizal	izable up to 240 TB					
Data Rate	4x1GigE	4x1/10GigE or 2x1/10GigE	8x10GigE, 2x10/25GigE, 2x40GigE, 2x100GigE					

GL's <u>PacketScan<sup>™</sup> HD 5G Protocol Analyzer</u> can monitor 5G networks. It captures, decodes, and collects statistics over N1N2, N4, N8, N10, N11, N12 and N13 interfaces of the 5G network. The 5G Protocol Analyzer is an optional module available within PacketScan<sup>™</sup> HD on purchasing of additional licensing.

PacketScan<sup>™</sup> HD supports decoding of <u>eCPRI protocol</u> which enables analysis of eCPRI message types such as IQ Data, Bit Sequence, Generic Data Transfer, Remote Memory Access, One-way Delay Measurement, Remote Reset, and Event Indication.

GL's **TCP Analytics** application analyzes TCP connections between both internal LAN and external WAN computers including servers and clients. The application helps troubleshoot large bandwidth consumption, failed TCP sessions, packet loss, poor TCP throughput and more. TCP Analytics (PKV400) is an optional application with PacketScan<sup>™</sup> HD network monitoring appliance. For more details, refer to <u>TCP Analytics</u> webpage.

For more details, refer to <u>PacketScan<sup>™</sup> HD - Network Monitoring Appliance</u> webpage.

🔊 GL Communications Inc.

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### **Main Features**

- Supports the following configurations: 4 x 1 GigE, 2 x 10 GigE, 2 x 40 GigE, and 2 x 40 / 2 x 100 GigE
- PacketScan<sup>™</sup> HD works with <u>FastRecorder<sup>™</sup> and PacketExtractor<sup>™</sup></u> application for wirespeed IP traffic filtering and recording capabilities of up to 320 Gbps directly onto disk for offline filtering, extraction, and analysis
- Using PacketScan<sup>™</sup> HD system along with FastRecorder<sup>™</sup> application, users can capture the traffic and analyze the captured data using GL <u>IP Analytics<sup>™</sup></u> tool
- Supports 5G interfaces N1N2, N4, N8, N10, N11, N12, and N13
- Wirespeed unfiltered continuous capture to NVMe SSD up to hard disk size
- PacketScan<sup>™</sup> HD can monitor 20,000 simultaneous calls with bidirectional RTP traffic from 1 Gbps to 100 Gbps link rates. Up to 50,000 calls can be achieved by scaling with higher configurations
- Simultaneous operations with contiguous/multiple cards, (1GigE, 10 GigE, and 40 GigE) subject to the performance limitation and up to maximum of 4 cards are supported
- Provides wirespeed hardware filter capabilities to filter traffic of interest
- Supports almost all industry standard IP and Wireless Protocols (from SIP to LTE)
- Supports all RTP traffic Voice, Data, Video, Fax T.38, Digits, Tones, Impairments
- Capture and Call processing is enhanced to handle different Tunnel traffic (VXLAN, GRE and GTP) and multiple tunnelling
- Support for eCPRI decode

### As a Single Point Packet over IP CDR Analysis System

- PacketScan<sup>™</sup> HD can work with GL's <u>VoiceBand Analyzer (VBA)</u> and <u>Call Data Records (CDR)</u> applications to generate Call Detail Records as (\*.CSV files) along with voice files for each direction
- PacketScan<sup>™</sup> HD can send protocol fields, and call detail records, along with traffic summary of captured calls to a central database and <u>NetSurveyorWeb</u><sup>™</sup> displays the data from the database in a simple web-based browser, featuring rich graphics, custom search, report and filter configurations

#### **Filter and Search Capabilities**

PacketScan<sup>™</sup> HD supports three stages of filtering:

- Hardware Filter high speed, discards unwanted packets at the hardware level
- Capture Filter slower discards unwanted packets at the application level
- View Filter and Search (Post Capture Filter) performs filtering on the captured trace only for viewing purposes; filtered trace can be exported to PCAP or GL's HDL file format

#### **Supported Codecs**

- G.711 (a-Law and  $\mu$ -Law), G.711 App II (a-Law and  $\mu$ -Law with VAD)
- G.722, G.722.1 (Wideband), G.726, G.726, with VAD, G729, G729B (8kbps)
- GSM, GSM HR, GSM EFR
- SPEEX/SPEEX\_WB (Narrow band/Wideband)
- iLBC (20ms and 30ms), SMV
- AMR/AMR\_WB (Narrow band/Wideband) (requires additional license)
- EVRC, EVRC0, EVRC-B, EVRC-B0, EVRC-C (requires additional license). Visit Voice Codec webpage for more details

### **Supported Protocols**

- SIP, SIP-I, SIP-T, H.323, MEGACO, MGCP, Diameter, Skinny (SCCP)
- LTE, SIGTRAN SS7, ISDN, GSM A and Abis over IP, GPRS Gb and Gn over IP
- UMTS IuCS, IuH, IuPS, and IuUP over IP, T.38 Fax and Video calls. Visit <u>Supported Protocols</u> for more details



# Main Features (Contd.)

### **QOS** Parameters

- E-model (G.107) based MOS/R-Factor scores
- Media Delivery Index (Delay Factor: Media Loss Rate) for video calls
- H.263, H.264 codec support for video conference monitoring capability

#### **Traffic Handling**

- All RTP traffic supported Digits, Tones, Voice, Video, Fax
- SIP ED 137B for Air Traffic Monitoring (Air-to-Ground and Ground-to-Ground Calls)
- Segregation of IP traffic and signaling
- Listen and Record audio streams, Video QoS Statistics
- Filters based on WhiteList Calls, Criteria based Voice/Trace Recording

### **Performance Metrics**

- Signaling, audio, and video QoS parameters for each call
- Minimum, maximum and average round trip delay
- Inband (DTMF and MF) events, Outband events as per RFC 2833 or RFC 4733 events, RTP/RTCP packet count and reports per direction

#### **Triggers and Actions**

• Filter the completed calls captures based on different signaling parameters and then specify a series of actions to be taken

#### **Report Generation**

- Ability to export summary report of selected or all completed calls in Packet Data Analyzer to CSV file format
- Analyze the CSV files using custom Excel<sup>®</sup> addins
- Ability to save a particular call in HDL, PCAP, or PCAPNG file format for further detail analysis
- Generates alert summary when particular vital parameters go beyond a specified value

### Statistics

- Quality Metrics with E-Model R-Factor and MOS Factors graphs, Jitter Buffer Statistics, Degradation Factor, Burst Metrics, and Delay Metrics
- Active calls, Average jitter, Packets Discarded, RTP packets summary, Detail ladder diagram



# **Specifications**

Supported Interfaces	<ul> <li>4x 1 Gbps: 850/1310 nm SFP Module; Ethernet/ Optical SFP</li> <li>2x 10 Gbps: 10GBASE-SR SFP+; Optical only</li> <li>2x 40 Gbps: MTP/MPO Connector for CFP2; Optical only</li> <li>2x 40/2x 100 Gbps: MTP/MPO Connector for CFP2; Optical only</li> </ul>
OSI	MAC, ARP, IP, IGMP, ICMP, TCP, UDP, SCTP, FTP, HTTP, TLS, SMTP
Protocols	GSM, UMTS, LTE, IMS, SIP, RTP, T.38, RTCP, and much more (some protocol support requires additional licensing)
Capture Timestamp	Absolute, Relative, Difference, NTP 4 Nano-second resolution
Captured Trace Format	GL's Proprietary HDL, PCAP, PCAPNG Frame Decodes can be saved to text file format



PacketScan<sup>™</sup> HD Rack System -1G/10G/40G/100G



PacketScan™ HD Portable System 1G/10G/40G/100G

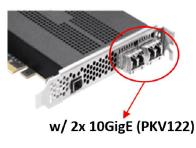
Pelican Carry Case



# Specifications (Contd.)

Filter	Hardware Filter at line rate, Application Level Capture Filter, and Post Processing Filter and Search
Performance	<ul> <li>4 x 1GigE: 20000 calls with bi-directional RTP traffic</li> <li>2 x 10GigE: 30000 calls with bi-directional RTP traffic</li> <li>Extracting/recording voice <ul> <li>2500 simultaneous calls (maximum)</li> <li>Option to record filtered calls of interest only</li> </ul> </li> </ul>
Protocols	5G, LTE, IMS, SIP, RTP, T.38, RTCP, UMTS, GSM, and more (some protocols require additional licensing)
Rack/Portable System Specifications	<ul> <li>Dual Processor Xeon/Single processor</li> <li>Expandable memory up to 128 GB</li> <li>Intel DQ67SW uATX LGA1155/Q67 motherboard</li> <li>17" 1280 x 1024 LCD (Optional Resistive Touch)</li> <li>LCD Specifications: 180 H/180 V viewing angle, 250 nits, 1500:1 contrast ratio 16.7M colors, 8ms response time</li> <li>DVI-A for integrated LCD video interface</li> <li>Standard I/O Interfaces: Integrated GbE, Serial Port, 2 USB3, 4 USB2, 2 eSATA, 2 SATA6, 2 SATA3, 1394, Audio/ Speaker</li> <li>PCI Expansion Slots: One PClex 16, one PClex 4 (or PCl)</li> <li>PCI Slot Lengths: 9-13" depending on configuration</li> <li>Removable hard drives up to 4x2.5" SATA/SSD</li> <li>500 GB NVME SSD, customizable up to 240 TB</li> <li>Optical Drive: DVD/CD Writer or BluRay Burner</li> <li>Video Projector Ports: DVI-I and Display Port</li> <li>Power Supply: 275 Watt 90 – 264VAC 50 – 60 Hz</li> <li>Size Closed: 16"W x 16.3"H x 5.4"D</li> <li>Size Open: 16"W x 16.3"H x 8"D</li> <li>Environmental: 0° - 50°C 10-90% Rel. humidity</li> <li>Transit Case (optional): Pelican 1610 with custom polyethylene foam</li> <li>Weight: 26 pounds; Total Weight of Computer with Transit 40-45 pounds</li> </ul>

# **Portable Platforms**





w/ 4x 1GigE (PKV120)



w/40 or 100 GigE (PKV124)



# **Comprehensive Filtering Capabilities**

The PacketScan<sup>™</sup> HD application permits users to filter out traffic of interest at two levels prior to capture.

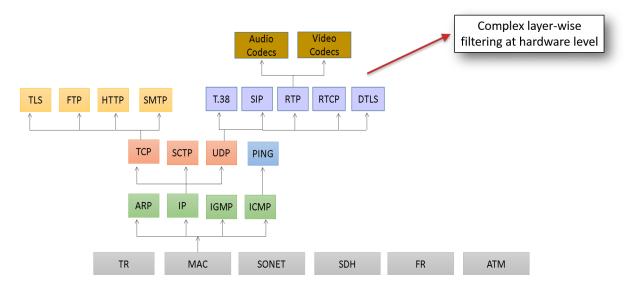
#### **Hardware Filter**

- Permits users to filter out packets of interest at hardware level on high density network and discard unwanted traffic
- Create up to 10 user defined hardware filters to filter-out traffic based on Layer-wise parameters such as Frame size and MAC, 802.1Q (VLANs), IPv4 /IPv6, TCP, UDP, SCTP, GTP, SIP, RTP and more
- Ability to set filter conditions either before capturing the packets, or while running real-time capture
- Complex filtering capabilities at the lower hardware level result in Low CPU load on the host server
- Users can create their own filters using custom filter option which provides flexibility to check the fields and use the logical conditions more efficiently

Capture File Options	Capture Filters Hardware Filte	ers						
∬ Card & Stream Selection ☆ Capture Filter ∑ Gui & Protocol Options	Packet Slicing Length 0	Field ID	Filter T Protocol IPLIST	Type Advanced Field Name	Operator	Value	Condition	IP List Type IP Address Pair List  IP Layer Type Outer Non Tunnel
	Filters       Piders       Prv4       Prv4ProtocoTrype       Prv4FotocoTrype       Prv4Eoderont       Prv4Eoderont	Selected F KeyList[ Assign[5 <	Insert Expression	n Delete Clear n DotIpv4; KeySet=0] = ([192. ] = (((Layer3Protoc) = IPv6 ions Final Applied Expressio	All	erIpv6) == 9 ) OR		IP Address       IP Address         192.168.12.216       >         192.168.12.209       IP Address         192.168.12.209       IP Address         Add       Edt         Delete       IP Address         Update       IP Address         Validate 8: Update       IP Address         Validate 9: Update       IP Address

#### **Software Filter**

• Layer-wise complex software filtering further can be applied at the application level based on different signaling parameters further, with Triggers and action feature, one can perform automated actions on the filtered completed calls

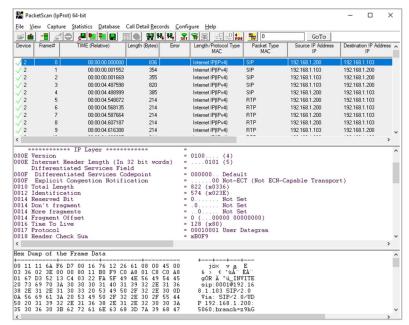


#### Document Number: PKV120-01

### **Summary Frame View**

#### Summary, Detail, and Hex Dump Views

The Summary View displays various information such as Frame Number, Time, Length, Message Types, IP source and destination addresses, and so on. Any field from the protocol headers can be added to Summary view, i.e., summary fields are completely user-configurable. Users can select a frame in Summary View to analyze and decode each frame in the Detail View. The Hex dump view displays the frame information in HEX and ASCII octet dump.



**Different Views** 

#### **Summary View**

Device	Frame#	TIME (Difference)	Length (Bytes)	Error	Length/Protocol Type MAC	Packet Type MAC	Source IP Address IP	Destination IP Address IP	Source Port UDP	Destination Port UDP	SIP Method Sip3261	SIP From Sip3261
12	0	00:00:00.000000	836		Internet IP(IPv4)	SIP	192.168.1.200	192.168.1.103	54098	5060	INVITE	0001@192.168.1.200
12	1	00:00:00.001552	354		Internet IP(IPv4)	SIP	192.168.1.103	192.168.1.200	54098	5060	SIP/2.0 100 Trying	0001@192.168.1.200
12	2	00:00:00.000117	355		Internet IP(IPv4)	SIP	192.168.1.103	192.168.1.200	54098	5060	SIP/2.0 180 Ringing	0001@192.168.1.200
12	3	00:00:04.485929	820		Internet IP(IPv4)	SIP	192.168.1.103	192.168.1.200	54098	5060	SIP/2.0 200 OK	0001@192.168.1.200
12	4	00:00:00.001401	385		Internet IP(IPv4)	SIP	192.168.1.200	192.168.1.103	54098	5060	ACK	0001@192.168.1.200
12	5	00:00:00.059073	214		Internet IP(IPv4)	RTP	192.168.1.200	192.168.1.103	1024	1024		
12	6	00:00:00.020063	214		Internet IP(IPv4)	RTP	192.168.1.200	192.168.1.103	1024	1024		
12	7	00:00:00.019529	214		Internet IP(IPv4)	RTP	192.168.1.200	192.168.1.103	1024	1024		
2	8	00:00:00.019523	214		Internet IP(IPv4)	RTP	192.168.1.200	192.168.1.103	1024	1024		
12	9	00:00:00.009121	214		Internet IP(IPv4)	RTP	192.168.1.103	192.168.1.200	1024	1024		
12	10	00:00:00 01:2377	214		Internet (PI(Pv4)	BTP	192 168 1 200	192 168 1 103	1024	1024		

#### **Detail Decode View**

Device2 Frame=0 at 16:58:57.799237 OK Len=836	*** Right click to SHOW/HIDE layer de
Ethernet Frame Data	
MAC Laver	-
0000 Destination Address	- x0011116AF6D7
0006 Source Address	= x001676122661
000C Length/Protocol Type	= x0800 Internet IP(IPv4)
IP Laver	- · · · · · · · · · · · · · · · · · · ·
800E Version	- 0100 (4)
000E Internet Header Length (In 32 bit words)	0101 (5)
Differentiated Services Field	- 11
000F Differentiated Services Codepoint	= 000000 Default
800F Explicit Congestion Notification	=
0010 Total Length	= 822 (x0336)
0012 Identification	= 574 (x023E)
0014 Reserved Bit	= 0Not Set
8014 Don't fragment	= 0. Not Set
0014 More fragments	Not Set
0014 Fragment Offset	= 0 (00000 0000000)
0016 Time To Live	= 128 (x80)
0017 Protocol	= 00010001 User Datagram
8018 Header Check Sun	- xB0F9
001A Source IP Address	= 192.168.1.200 (xC0A801C8)
001E Destination IP Address	= 192,168,1,103 (xC0A80167)
UDP Laver	- · · · · · · · · · · · · · · · · · · ·
0022 Source Port	= 54098 (xD352)
8024 Destination Port	= 5060 (x13C4)
0026 Length (Header + Data)	= 802 (x0322)
0028 Checksun	= xFA5F
Sip3261 Layer	-
HDR	- INVITE sip:00010192.168.1.103 SIP/2.0
HDR	= Via: SIP/2.0/UDP 192.168.1.200:5060;branch=z9hG4bK3811333536-332
HDR	= Max-Forwards: 70
HDR	= Allow: INVITE, BYE, CANCEL, ACK, INFO, PRACK, CONET, OPTIONS, SUBSCRIBE, NOTIFY, REGISTER, UPDATE
HDR	= From: 0001 <sip:00010192.168.1.200>;tag=GLPG_3011333536=333</sip:00010192.168.1.200>
HDR	= To: 0001 (sip:00010192.168.1.103)
HDR	= Call-ID: GLPG-483633760331
HDR	= CSeq: 1 INVITE
HDR	= Contact: 0001 <sip:00010192.168.1.103></sip:00010192.168.1.103>
HDR	= Content-Type: application/sdp
HDR	= Content-Length: 349

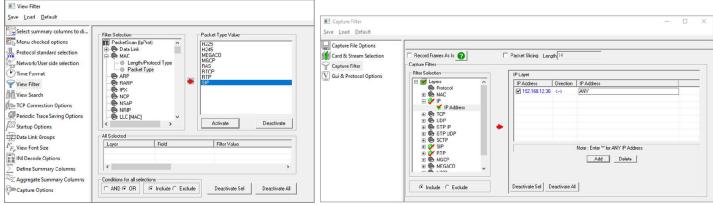
#### **Hex Dump View**

Her	c Di	ımp	of	the	e Fi	rame	e Da	ata								
+				-+				-+				-+				+++
00	11	11	6À	F6	D7	00	16	76	12	26	61	08	00	45	00	jöx v <u>a</u> E
03	36	02	3E	00	00	80	11	B0	F9	C0	Α8	01	C8	CO	Α8	6 > € °ù ÈÀ¨
01	67	D3	52	13	C4	03	22	FA	5F	49	4E	56	49	54	45	gÓR Ä "ú_INVITE
20	73	69	70	ЗÀ	30	30	30	31	40	31	39	32	2E	31	36	sip:0001@192.16
38	2E	31	2E	31	30	33	20	53	49	50	2F	32	2E	30	0D	8.1.103 SIP/2.0
ΔO	56	69	61	ЗÀ	20	53	49	50	2F	32	2E	30	2F	55	44	Via: SIP/2.0/UD
50	20	31	39	32	2E	31	36	38	2E	31	2E	32	30	30	ЗA	P 192.168.1.200:
35	30	36	30	ЗB	62	72	61	6E	63	68	ЗD	7A	39	68	47	5060;branch=z9hG
34	62	4B	33	38	31	31	33	33	33	35	33	36	2D	33	33	4bK3811333536-33
32	ΟD	ΟÀ	4D	61	78	2D	46	6F	72	77	61	72	64	73	ЗÀ	2 Max-Forwards:

### **Filtering and Search**

Filter and search capabilities adds a powerful dimension to the SIP analyzer. These features isolate required frames from original frames in real-time/offline. Users can record all or filtered traffic into a trace file.

Allows real-time filtering based on parameters set in Data Link layer, MAC layer, IP, TCP/UDP, and more. The offline filter allows filtering based on Frame Number, Time, Length, Message Types, and so on. Similarly, search capability helps users to search for a particular frame based on specific search criteria.



**View Filter** 

**Capture Filter** 

### Analysis of VoIP and Wireless Calls – Summary View

#### **Summary View**

TA Summary view displays summary of data transmission in each direction including calling number, called number, call id, start time, duration, missing packets, max/min RTD, average RTD and so on. Calls and sessions are classified as active, completed, or failed giving the users an idea about the calls and its status in the network. It includes separate statistical counts on total packets, calls, failed calls, and more, for SIP, H.323, MEGACO, RTP, GSMA, IuCS, and SCCP based calls.

#### Call Summary – Signaling, Audio, and Video QoS Statistics

The Call Summary displays the signaling, audio, and video parameters of each call for SIP, RTP, MEGACO, H.323, GSMA, IuCS, and SCCP protocols. Video QoS parameters such as Codec Info, Frame Rate, Missing Packets, Delay, Gap, Video Frame Count, Out Of Sequence count, Duplicate Packets count, Media Delivery Index (MDI), etc. are displayed for all video calls with H.263 and H.264 codecs.

Packet Data Analyzer - Summary View						-	
ile View Call Summary Protocol Configu	rations GUI Config	jurations Help					
🚰 🔎 🎦 👯 🗗 🕨 🗉	图 新 幣	SIP -	Show All Sessions	-			
Call Summary Registraton Summary Alert Summa	ry						
Call # SSRC Payload Packe Cor Rec M0	versati Listening Pa S/R MOS/R Di	ckets Missing Duplicate Out Of scard Packets Packets Sequen	Average Average Average Gap(ms) Delay Jitter	Average Cumulative Ma Inter Arri Packet	x/Min Gap Max/1/ Delay	Max/Min Jitter	Max/Min RTDelay(ms
Cal#000001 Cale::0001@192.168.1.200 Calee:	0001@192.168.1.103	Calld:GLPG-483633760331 Call StartTime	:2011-02-10 16:58:57.799 Call Du	ration: 00:00:25.489			
	0/93 4.20/93 0 0/93 4.20/93 0	/0.00 0/0.00 0/0.00 0/0.00 /0.00 0/0.00 0/0.00 0/0.00	20.01 0.00 0.00 20.01 0.00 0.00		.65 / 18.97 1 / -1		0.000 / 0.0
ε 3360545537 ΡCM076000 1265 4.2 ε	0733 4.20733 0	70.00 070.00 070.00 070.00	20.01 0.00 0.00	0 0 21	.61718.01 17-1	0.667 0.00	0.13070.1
Signaling Parameters	Value	Audio Parameters	Value ^	Video Parameters		Value	
Caller	0001@192.168	Stc RTP Channel	192.168.1.200	Sic Video Channel			
Callee	0001@192.168	Src Media Type	PCMU/8000	Sic Media Type			
Calld	GLPG-4836337	Stc SSRC	3365468417	Sic SSic			
Call Status	Terminated	Stc Packets Count	1273	Stc Packets Count			
		Stc Missing Packets / (%)	0 / 0.00	Stc Missing Packets / (%			
Call Initiated Time	2011-02-10 16:	Src Duplicate Packets / (%)	0 / 0.00	Src Duplicate Packets /			
Call Established Time	2011-02-10 16:	Src Out of Sequence Packets / (%)	0/0.00	Stc Out of Sequence Par			
Call Stop Time	2011-02-10 16:	Src Conversational MOS/R-Factor	4.20 / 93	Stc Video Frame count			
Call Duration	00:00:25.489	Src Listening MOS/R-Factor	4.20 / 93	Stc Frame Rate(Frames/s			
Call Terminator	Caller	Src Discarded Packets / (%)	0/0.00	Sic AvgDelay			
Call Failure Reason		Src Average Inter Arrival Jitter (RTCP)	0	Sic AvgGap			
		Src Average Jitter	0.00	Stc MDI (DF:MLR)			
Session Request Delay (msec)	1.669	Stc Average Delay	0.00	Stc AvgMDI(DF:MLR)			
Session Disconnect Delay (msec)	0.905	Stc Average Gap	20.01				
Post PickUP Delay (msec)	128.905			Dest Video Channel			
		Dest RTP Channel	192.168.1.103	Dest Media Type			
Total Signaling Frames	7	Dest Media Type	PCMU/8000	Dest SSrc			
		Dest SSRC	3380545537	Dest Packets Count			
		Dest Packets Count	1269	Dest Missing Packets / (			
		Dest Missing Packets / (%)	0/0.00	Dest Duplicate Packets /			
		Dest Duplicate Packets / (%)	0/0.00	Dest Out of Sequence Pa Dest Video Frame count	ackets / (%)		
		Dest Out of Sequence Packets / (%)	0/0.00				
		Dest Out of Sequence Packets / (%) Dest Conversational MOS/R-Factor Dest Listening MOS/R-Factor	4.20 / 93	Dest Frame Rate(Frames. Dest AvgDelay	/sec)		

Call Summary, Audio and Video Statistics

# Packet Data Analysis (PDA)

#### Features

- Call Quality Of Service (QoS) for all calls with E-Model based (G.107) Mean Opinion Score [MOS (ITU-T, G.107, E-model)] and R-factor with individual and summary statistics presented in graphical and tabular formats
- Provision for H.263+ and H.264 video capture and video conference monitoring capability
- Calculates minimum, maximum, and average round trip delay values for SIP calls
- Supports decoding of AMR and AMR\_WB codec with IuUP Header
- Save calls in HDL, PCAP, or PCAPNG file format for further analysis
- Ability to copy the cell value to clipboard (Notepad)
- The PDA Summary View can also export all terminated call details as a text file (CSV format) during the live capture. This feature requires activating the Export Terminated Calls option from PDA prior to live capturing
- This structured text file can be imported into Excel<sup>®</sup> using a custom add-in (Excel-Dashboard-Tool-IP.xlsm) to generate different chart types such as call volumes, call duration, call failure causes, CMOS, LMOS, packet loss and more
- Individual and summary statistics presented in graphical and tabular formats
- Graphs are provided for key statistics for network monitoring and troubleshooting. Graphs available include Active Calls, Average
  Jitter, E-Model MOS/R-Factor/Packets Discarded, RTP Packets Summary, ladder diagram for T.38 based fax calls and call signaling, Gap,
  Jitter, Gap/Jitter Distribution, Wave and Spectral Display for media stream analysis, VoIP calls and more
- Displays a summary of signaling, audio, and video parameters such as Source/Destination Video Channels, Media Type, SSRC, Average Delay/Gap, Packet Counts, Media Delivery Index and Frame Rate for all video calls
- Calls and sessions are classified as active, completed, or failed giving the users an idea about the calls and its status in the network
- Filter CDRs (Call Detail Records) based on parameters such as caller, time, message count, etc.
- Generates VoIP Key Performance Indicators (KPI) Reports: Call Success Ratio, Calls Per Second, Post Dial Delay, Error Code Distribution, Answer Seizure Ratio, and Call Duration
- Creates SIP Registration KPI Reports: Register messages per sessions ,Registrar(s) distribution, Registration(s) vs Deregistration(s) Over Time, Error code distribution
- Export KPI Report in PDF Format
- Generates alert summary when particular vital parameters go beyond a specified value

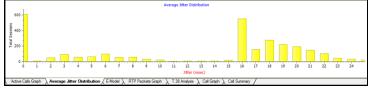
	ket Data Analyzer - Summary								- 0	×
<u> </u>	ew Call Summary Protoco									
<u>🎢</u> 🏳	) 🎦 🏪 🙀 🕑 🕨 I	🗉 🖄 🚮 🕈 📍	👫 📲 SIP	Show All C	Calls	-	Call	Count: 117		
Call Sun	nmary SIP Registration Summar	ry Alert Summary								
Call#	Caller	Callee	StartTime	Duration	VoiceQuality_L	VoiceQuality_R	Payload_L	Payload_R	Result	^
1	0628@[fe80::3f20:7953:f2df	)628@[fe80::6c2b:	. 2023-12-20 11:20:30.261	00:00:48.864			AMR_WB/16000	AMR_WB/16000	Pass	
2	0629@[fe80::3f20:7953:f2df		. 2023-12-20 11:20:30.313	00:00:48.799			AMR_WB/16000	AMR_WB/16000	Pass	
3	0630@[fe80::3f20:7953:f2df		. 2023-12-20 11:20:30.365	00:00:48.746			AMR_WB/16000	AMR_WB/16000	Pass	
4	0631@[fe80::3f20:7953:f2df		. 2023-12-20 11:20:30.406	00:00:48.713			AMR_WB/16000	AMR_WB/16000	Pass	
5	0632@[fe80::3f20:7953:f2df		. 2023-12-20 11:20:30.451	00:00:48.652			AMR_WB/16000	AMR_WB/16000	Pass	
6	0633@[fe80::3f20:7953:f2df		. 2023-12-20 11:20:30.504	00:00:48.601			AMR WB/16000	AMR WB/16000	Pass	~
<										>
				[		Counter Type		Count	210	
		Export Gra	ph Duration 1 min 💌	2023-12-20 11:20:30	Goto	Save Lounter Type		173	510	
						Active Calls		101		
						Completed Calls		0		
		Graph Start - (202	3-12-20 11:19:41) Graph End	- (2023-12-20 11:20:4	ю)	Purged Calls(cle	ared)	56		
Call	sPerSecond					Failed Calls Calls Per Second	4	56 0		
	spersecond					Non Purged Call		117		
			Attempted Calls	Passed Calls	🔽 📕 Failed	Call				
						Total Frames		4 787		
	4					Last Frame Proc Total Processed		4 622		
2	10 -			~			Frames Before Processing	6 374		
	1		$\sim$			Queue ToDecoo	de:Decoded	0:0		
1	.5 -		/ \			TimeToProcess(		0::48		
	-		<b>\</b>				ec :: Rate(Mbps)	0 :: 0	.00	
Calls/Sec	10 -			N		HdfWriteDrop <fr< td=""><td>m:byte&gt;</td><td></td><td></td><td></td></fr<>	m:byte>			
3	"]					VOIP Bandwidth	1	0.00		
	-					SIP Bandwidth		0.00		
	5 -		N N	Λ		H323 Bandwidth		0.00		
	1		V			MEGACO Bandy	width	0.00		
	0					RTP Bandwidth IuCS Bandwidth		0.00		
1	1:20:00					Gsma Bandwidth		0.00		
			Tim	e		SCCP Bandwidt		0.00		
1										
<						>				
	Rate / RTP Packets Graph	Average Jitter Dis	tribution & E-Model & T	.38 Analysis 🔪 Call Flo	w & Call Summary	<u>-</u>		/		
Cails		A Average oncer Dis				_/ _   <b>∖OverAll</b> <u>/</u> SIP	$\lambda$ H323 $\lambda$ RTP $\lambda$ MEGAC	Ο <u>λ</u> ED137 /		

Call Summary View in PDA

## **PDA Graphs**

#### Calls Rate Graph in PDA – Summary View

**Calls Rate** – A line graph, depicting the Number Of Calls Vs Time. **Average Jitter Distribution** – Distribution of the Average Jitter values across the Total Sessions.



#### Average Distribution and Calls Rate Graphs

**E-model** - This 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 values across No. of Sessions
- Packets Discarded A bar Graph that plots Packets Discarded across No. of Sessions
- RTP Packets Graph Plots and compares out of ordered packets, missing packets and duplicate packets against Total Audio Packets



Export Graph Duration 1 min V 2023-12-20 11:20:30 V Goto Save

Failed Cal

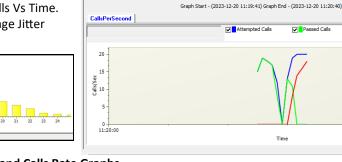


T.38 Analysis - Fax (T.38 data) over VoIP monitoring and decoding capability.

Call Flow - Displays the message sequence of protocols such as SIP, SIP ED137B, MEGACO, and H.323 captured VoIP calls.

Packet Data	Analyzer - Sumn	nary View												- 0	×					
	Il Summary Pro	otocol Configur	rations <u>G</u> UI Con	figurations <u>H</u> elp																
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Call Summary S	IP Registration Sur	mmary Alert Sur	mmary																	
Call#	Caller		Calee	StartTime	Dura			VoiceQualit	y_L	Voice	Quality_R	Payload		Payload_P						
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Column Width	J	Absolute Tin	ning 🔲 Show Lat	est																
Time	Frame#	67.120	0.213.197	122.166	.105.193				Find 🔽	Complete S	tack									
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00.03.429	3	5060	SIP/2	2.0 180 Ringing	5060	£ 12	) 🍓 🖡	9 🕌 😏	► = 🎽	刘 🚮	🌾 🚻 SIF	2	▼ Sho	w All Cal	ls	•		Call Co	ount: 824	
00.07.926	4	5060		7/2.0 200 OK	5060	-	mmary SIF		Summary Alert											
00.08.252	5	5060		ACK	5060	Call#	0001@	Caller 192.168.12.9		lee	StartTin 2023-06-01 15:		Duration 00:01:00.023	Voice	eQuality_L	VoiceQuality_R	Payload_L PCMU/8000	Payload_R PCMU/8000	Result Pass	ErrorCode ^
00.08.313	6	6002	BTP (	PCMU/8000)	6002	2	0002@	192.168.12.93	2 0002@192	. 168. 12.94	2023-06-01 15:	01:34.482	00:01:00.033				MuLAW_2/8000	MuLAW_2/800	IO Pass	0
				INVITE		3		192.168.12.9			2023-06-01 15: 2023-06-01 15:		00:01:00.045				PCMA/8000 ALAW_2/8000	PCMA/8000 ALAW_2/8000	Pass Pass	0
00.09.520	47	5060	4		5060	5		192.168.12.9			2023-06-01 15: 2023-06-01 15:		00:01:00.049				G729/8000 G7298/8000	G729/8000 G7298/8000	Pass	0
00.09.847	58	5060	SIF	2/2.0 200 OK	5060	7		192.168.12.9			2023-06-01 15:		00:01:00.043				GSM/8000	GSM/8000	Pass	0 🗸
00.09.881	59	5060	◀	ACK	5060	<														>
02.51.889	3227	5060		BYE	5060	Column	width _		Absolute	e Timing ∣	Show Latest				0					
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<						00	0.00.020	1	5060	4	SIP/2.0 100	D Trying		60	Length	Address /Protocol Type			= x54BEF737B = x0800 Inte	rnet IP(IPv4)
Aver	age Jitter Distributio	n ) E-Model	X T.38 Analysis	Call Flow Call Su	mmary /		0.00.029	2	5060		SIP/2.0 180	Ringing	50	60	Versio				= 0100 (	
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						01.	.00.177	3984	5060		BYE		- 50	60		ification ved Bit	Displ	lays	= 15592 (x3C = 0 N	
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						<								>	<					>
						4 >	Averag	ge Jitter Distribu	ution X E-M	odel 🔪 T.3	38 Analysis Ca	II Flow / Ci	all Summary /							
						1														

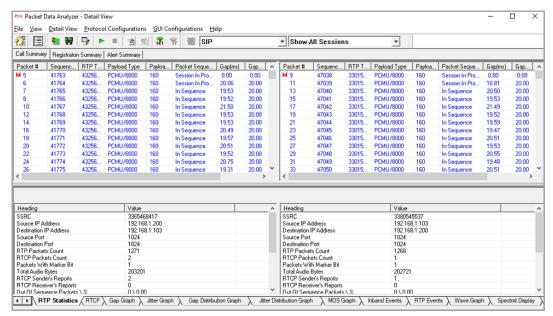
SIP Call Graph



### Analysis of VoIP and Wireless Calls – Detail View

### **Detail View**

This display assists in any comparisons that are to be made between the two RTP sessions of a call. Each frame of the selected session is dissected and its contents are displayed in a tabular form for easier viewing and comparisons. Vital aspects from the RTP frame needed for close analysis are included in the table.



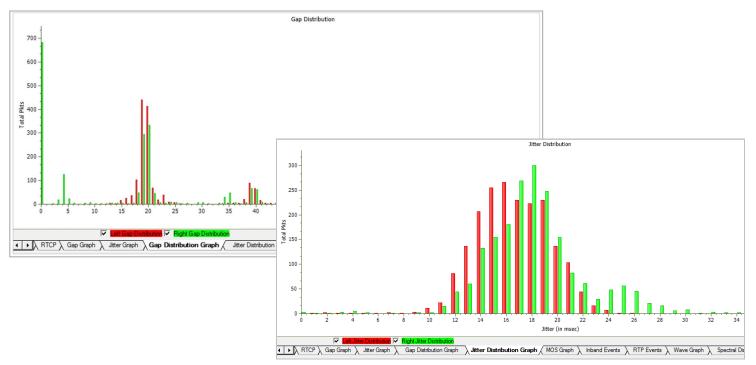
**Traffic Analyzer Detail View** 

#### **Graphs in Detail View**

Gap/Jitter graphs - Plots the Gap (in milliseconds)/Jitter versus the packet number.

**Gap Distribution Graph** - Number of packets with a particular value of gap is plotted against the (gap) value.

Jitter Distribution Graph - Number of packets with a particular value of jitter is plotted against the jitter value.



Gap/Jitter Distribution Graph

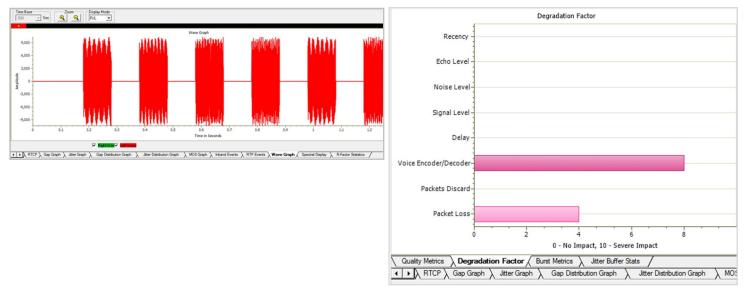
### Analysis of VoIP and Wireless Calls – Detail View (Contd.)

MOS Graph – Plots Mean Opinion Score values throughout the duration of the call.

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.

**Degradation Factor** – A pie chart plots and compares different statistics such as Good Quality, Packets discarded, Echo level, Packet loss, and Regency against total Packets for each individual sessions.



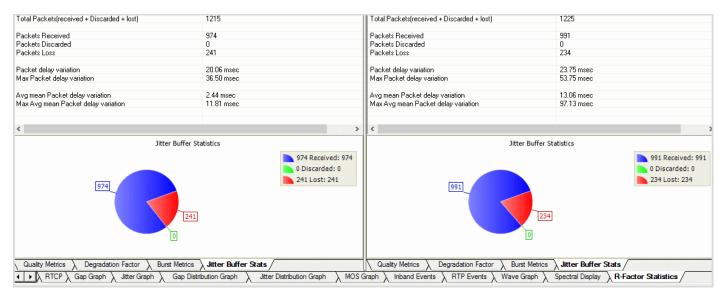
#### Wave Graph and Degradation Factors

### **R-Factor Statistics**

Quality Metrics based on E-model includes R-Factor and MOS Factor. R-Factor bar graph will display statistics such as R Listening, R Conversational, R-G107, and R-Nominal values.

MOS Factor bar graph will display statistics such as MOS CQ, MOS PQ, and MOS Nominal values during a call.

Jitter Buffer Statistics – A pie chart plots and compares packets received, packets discarded and packets lost against total Packets for each individual sessions. Also provides a tabular data on average.



**Jitter Buffer Statistics** 

### **Other Features**

#### Play Audio and Write to File

The Play Audio plays the selected call to the PC speaker. Write to File is similar to the Play Audio option. The basic difference being that the output is written to a file instead of playing to the speaker. PDA can monitor video calls and display both audio and video RTP streams in summary view.

Options Jitter Buffer Option											
Corrected     C Static      Dynamic     As Is     Jitter Buffer Len     100     msec     Dynamic Jitter Buffer Option											
Mix Option     Min Delay     40     msec       Image: Mix     Mix     Max Delay     100     msec       Image: Stareo     Image: Stareo     Image: Stareo     Image: Stareo       Image: Stareo     Image: Stareo     Image: Stareo     Image: Stareo											
Start from beginning File Record Use SSRC for File Name C:\Program Files\GL Communications Inc\PacketScan\Sample1.wav Invoke Cool Edit after write											

Write to File

#### Save Call

The Save Call feature enables the users to save a particular call either in GL's proprietary \*.HDL file format or in Ethereal \*.PCAP file format or \*.PCAPNG file format. Call Summary details could also be saved for a particular call as a \*.rtf file. This is especially useful to get data from real-time traffic locations to the lab for detail analysis of a flawed call.

PDA Save Call	×
Call(s) CallNum_2 Goto Goto	
File Type       Image: HDL File       Image: PCAP	ary
Path C:\Program Files\GL Communications Inc\PacketScan\	
✓ Overwrite Files Save Call(s) Exit	

Save Call

### Other Features (Contd.)

#### **RTP/RTCP Statistics, Inband Events, Outband Events**

The users can get the complete details of a single selected call such as total packets count, SSRC, RTP packet count, RTCP packet count, total Audio bytes, and more.

Inband Events display Inband DTMF and MF digits as they are received with details such as Timestamp, Type, Event, On-Time, Power, and Frequency. Outband Events display RTP events as per RFC 2833 or 4733 with details such as Timestamp, Event, Power, and Duration.

#### **Triggers and Action Settings**

Triggers and Action Settings allow the users to filter calls based on certain SIP, RTP, MEGACO, H.323, GSMA, and IuCS parameters followed by a set of actions for the completed calls. The filtered file can be saved in either GL's proprietary HDL file, Ethereal PCAP, or PCAPNG file format. It extracts fax image for the selected fax calls. Additionally, a summary of call signaling and audio parameters can be saved as \*.rtf file, or generate Call Detail Records in CSV file format along with voice files for each direction. The CSV files can be used for further analysis and retrieval of **calls of interest**.

Triggers and Action Sett	ings - Untitled	×
<u>F</u> ile		
Trigger List	Filter Selection SIP SIP SIP SIP Salid Party Fax Calls Fax Calls Failed Calls Sip Error Code Call Duration (mins) Session Request Del Session Disconnect D	
Enter Trigger Name		And © Or
Action Save Call Audio Recording User Defined Send e-mail Alert Summary Call Detail Record Extract Fax Image	Save Call To File Options File Name Mask [%]_%Y_%M_%D_%h-%m-%s Files Destination Directory [C:\Program Files\GL Communications [] Create File Options If File Exists © Overwrite © Skip Operation © Append	Save Options HDL File PCAP File PCAPNG Call Summary d Sequence Number
L	<u>D</u> k <u>C</u> ancel	

**Trigger and Action Settings** 



# Other Features (Contd.)

#### **Alert Summary**

Generates alerts when particular vital parameters go beyond a specified value and display in Alert Summary table. The users can specify the criteria based on which the alerts are to be generated. The tab provides an active list of the alerts that have occurred during the test session in tabular columns.

PDA Pack	cet Data An	alyzer - Summary View						
File Vi	ew Call Su	ummary Help						
Call Sum	mary SIP F	Registration Summary Alert S	ummary					
Call#	Protocol	Message	Туре	Threshold	Value	Caller	Callee	Calld
9	SIP	Mos value is between 3 - 4	Major	12		0188	0188	GL-MAPS-31153140-4234814636-215113866-3524@fe80::858d:4953:7823:3bc6
6	SIP	Mos value is between 3 - 4	Major	12		0185	0185	GL-MAPS-31153098-4234811642-215113843-4612@fe80::858d:4953:7823:3bc6
8	SIP	Mos value is between 3 - 4	Major	12		0187	0187	GL-MAPS-31152178-4234814354-215113858-5712@fe80::858d:4953:7823:3bc6
7	SIP	Mos value is between 3 - 4	Major	12		0186	0186	GL-MAPS-31152619-4234812329-215113851-2884@fe80::858d:4953:7823:3bc6
10	SIP	Mos value is between 3 - 4	Major	12		0189	0189	GL-MAPS-31152913-4234816510-215113874-5676@fe80::858d:4953:7823:3bc6
4	SIP	Mos value is between 3 - 4	Major	12		0091	0091	GL-MAPS-31152073-4234715545-215113129-5712@192.168.12.102
3	SIP	Mos value is between 3 - 4	Major	12		0090	0090	GL-MAPS-31152516-4234715545-215113130-2884@192.168.12.102
2	SIP	Mos value is between 3 - 4	Major	12		0088	0088	GL-MAPS-31152560-4234715451-215113115-3580@192.168.12.102
1	SIP	Mos value is between 3 - 4	Major	12		0089	0089	GL-MAPS-31152992-4234715451-215113119-4612@192.168.12.102
5	SIP	Mos value is between 3 - 4	Major	12		0092	0092	GL-MAPS-31153035-4234715638-215113143-3524@192.168.12.102

#### **Alert Summary View**

#### **Registration Summary**

- Provides the registration summary of each SIP registration including the user agent, registrar, status, registered time, expiry time, time to live, remaining time, registration request delay (RRD), and Re-registration attempts
- Provides graphical view of the active registrations and registration trace of each registration

		-	mmary View								_	· 🗆	×
File Vie	ew Call Si		Help I Registrations	▼ Call Count: 1	5								
Call Sum	mary SIP F	Registration	Summary Alert Sun	nmary									
Call#	Method		RegisterRequest		Registrar		esult	Status	ErrorCode	CallD	RegisteredTime	Requests	^
0 1 2 3 4 <b>&lt;</b>	Register DeRegist Register Register Register	er	2023-11-15 11:1 2023-11-15 11:1 2023-11-15 11:1 2023-11-15 11:1 2023-11-15 11:1 2023-11-15 11:1	8:2         0001@192.168.12.112           9:1         0001@192.168.12.112           9:2         0001@192.168.12.112           9:4         0001@192.168.12.112	192.168.12.112 192.168.12.112 192.168.12.112 192.168.12.112 192.168.12.112	Pa Fa Fa Fa	assed assed ailed ailed ailed		0 0 404 403 423 400	GL-MAPS-16-33884 GL-MAPS-16-33884 GL-MAPS-23-33937 GL-MAPS-33-33971 GL-MAPS-33-33971		1 1 1	>
Column \	width - <u>]</u> -	Frame#		ning 🗔 Show Latest	192.168.12.112			Find	Complete	• Stack			
	0:00.000	214	5060	REGISTER	5060		II F	REGISTER sip:192.168.1	12.112 SIP/	2.0			
<	0:00.222	216	5060	SIP/2.0 200 DK		>	F F F C C C F C C	Via: SIP/2.0/UDP 152.1 Route: <sip:152.168.12 Max-Forwards: 70 Allow: INVITE,BYE,CANG From: 0001 <sip:00018192.168 Call-ID: GL-MAPS-16-33 Cseq: 1 REGISTER Expires: 120 Contact: 0001 <sip:000 Contact: 0 001 <sip:000 Contact. 0 000 Contact. 0 000 Contact.</sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:000 </sip:00018192.168 </sip:152.168.12 	2.112:5060; CEL,ACK,INF 192.168.12. .12.112> 3884996-950	<pre>lr&gt; '0, PRACK, OPTIONS, SI 112&gt;; tag=FromTag- 5-9928@192.168.12</pre>	UBSCRIBE,NOTIFY, 17-33884996-9506	REFER, REGI	STER
Regis	tration Graph	Regis	tration Trace										

**Registration Summary** 



### **Filtered Calls using Expressions**

The PacketScan<sup>M</sup> analyzer offers the option to filter call detail records based on parameters such as caller, time, and message count. 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 as shown in the below screenshot.

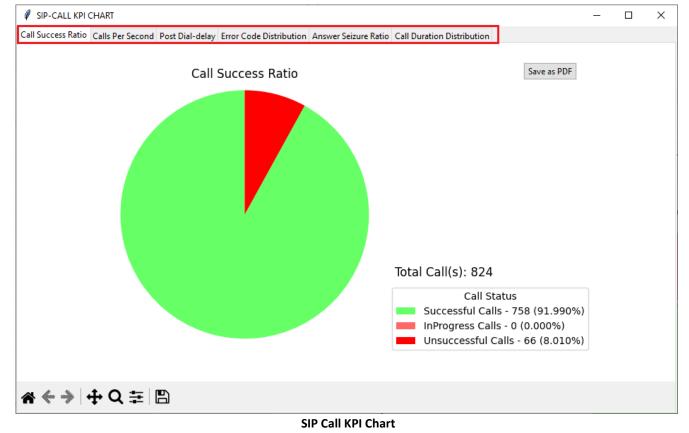
PDA Packet Data	a Analyzer - Sui	mmary View				– 🗆 X		
ile View Ca	all Summary	Protocol Config	jurations GUI Configurations Help					
📱 🔎 🏭	P         ■         ■         ■         ■         SIP         Show Filtered Calls         Call Count: 6							
ErrorCode==400	ErrorCode>600					X →		
Call Summary	SIP Registration	Summary Alert S	Summary					
Call Summary 9	SIP Registration ErrorCode	Summary Alert S	CalID	EndTime	PostDialDelay	SessionDisconnect		
				EndTime 2023-06-01 15:02: 12.275	PostDialDelay 9	SessionDisconnect		
	ErrorCode		CallID					
	ErrorCode 400	FailureCause 5	CalID GL-MAPS-2654-766727097-26124-3688@192.168.12.92	2023-06-01 15:02:12.275	9			
	ErrorCode 400 603	FailureCause 5 4	CalID GL-MAPS-2654-766727097-26124-3688@192.168.12.92 GL-MAPS-2679-766728649-26314-14696@192.168.12.92	2023-06-01 15:02:12.275 2023-06-01 15:02:13.828	9			
	ErrorCode 400 603 604	FailureCause 5 4 4	CalID GL-MAPS-2654-766727097-26124-3688@192.168.12.92 GL-MAPS-2679-766728649-26314-14696@192.168.12.92 GL-MAPS-2677-766728698-26320-13540@192.168.12.92	2023-06-01 15:02:12.275 2023-06-01 15:02:13.828 2023-06-01 15:02:13.879	9 9 9 19			

**Displaying Filtered Calls using Expressions** 

# **KPI Report for SIP Calls**

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

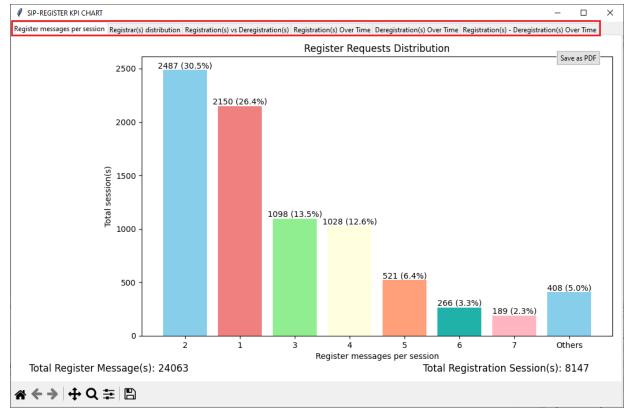
- Call Success Ratio: Displays graph for "Successful" and "Unsuccessful Calls," including counts and percentages (%)
- Calls Per Second: Shows graph "Total," "Passed," and "Failed Calls per second."
- Post Dial Delay: Shows delay counts in milliseconds (0-250ms, 251-500ms, etc.)
- Error Code Distribution: 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.)



### **KPI Report for SIP 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 Register messages per session



#### **SIP Registration KPI Chart**



# **Buyer's Guide**

Item No	Product Description
<u>PKV120</u>	PacketScan™ HD – High Density IP Traffic Analyzer w/ 4x1GigE - includes PKV100 – Online (not Offline) for tempo- rary audio codec support
<u>PKV120p</u>	PacketScan™ HD w/4 x 1GigE - Portable
<u>PKV122</u>	PacketScan™ HD–High Density IP Traffic Analyzer w/ 2x10GigE includes PKV100 – Online (not Offline) for tempo- rary audio codec support )
<u>PKV122p</u>	PacketScan™ HD w/2 x 10 GigE - Portable
<u>PKV124</u>	PacketScan™ HD – High Density IP Traffic Analyzer w/ 40/100 GigE
<u>PKV124P</u>	PacketScan™ HD – High Density IP Traffic Analyzer w/ 40/100 GigE - Portable

Item No	Related Software
<u>PKV112</u>	5G Analyzer (Optional with PacketScan™)
<u>PKV113</u>	Offline 5G Analyzer (Optional with PacketScan™ and NetSurveyorWeb™)
<u>PKV105</u>	SIGTRAN Analysis
<u>PKV103</u>	IP Based GSM and UMTS Analysis
<u>PKV110</u>	IMS Protocol Decodes (Optional with PacketScan™)
<u>PKV107</u>	LTE (Long Term Evolution) Analyzer, requires PKV100
<u>PKV104</u>	FaxScan <sup>™</sup> – Decodes T.38 Fax images in TIFF format from captured PCAP files

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# Buyer's Guide (Contd.)

Item No	Related Software
PCD103	AMR Narrowband Codec for PacketScan™
<u>PCD107</u>	Optional Codec – AMR Wideband
<u>PCD104</u>	EVRC Codec for PacketScan™
PCD105	EVRC-B Codec for PacketScan™
PCD106	EVRC-C Codec for PacketScan™
<u>PKV170</u>	NetSurveyorWeb™
PKV171	Network Surveillance Agent Toolkit
PKV172	Network Surveillance for GSM – GPRS Systems
<u>PKS118</u>	MAPS™ ED137 Radio
<u>PKS119</u>	MAPS™ ED137 Telephone (Includes PKS102)
<u>PKS117</u>	MAPS™ ED137 Recorder (Includes PKS102)
PKS107	RTP EUROCAE ED137
<u>PKV123</u>	FastRecorder™ and PacketExtractor™
<u>PKV169</u>	NetSurveyorWeb™ Lite
<u>PKV 400</u>	TCP Analytics

Note: PCs which include GL hardware/software require Intel or AMD processors for compliance.

For more details, refer to <u>PacketScan<sup>™</sup> HD - Network Monitoring Appliance</u> webpage.



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