How to debug, troubleshoot and monitor VoIP using MikroTik





Voicenter •Homer Whoami •SIP •RTP • RTCP **Voip Protocols** •QOS Provisioning Mikrotik VoIP Monitoring Setup Wireshark Sngrep CaptAgent **Tools for VolP** RtpAgent Analytics Troubleshooting Monitoring Alerting **Homer Cloud**





whoami <

Shlomi Gutman

CTO of **Voicenter** (Israel) **VP** of Cloud Products at **QXIP** (Amsterdam)



















whoami <

HI.

Shlomi Gutman.

Founder and CTO at Voicenter.

Open-Source Telephony expert.



Built my first computer when I was sand and sterisk

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Voicenter is A leading telecommunication technology company providing top-tier business telephony since 2007

We are delivering a 'One-stop-shop' solution for business all around the world





PBX







Voicenter – Cloud Contact Center

- Cloud-based Phone system
- Hybrid Solution
- Real Time Dashboards
- Workforce Management
- Dialers
- Api
- Integration





Voicenter – Cloud Contact Center

Voicenter is an Israel based business providing a solid array of services to its customers, including:

- Cloud based Phone Systems
- Hybrid Solutions
- Real Time Dashboards
- Workforce Management
- Dialers
- Api
- Integration

QXIP BV is an Amsterdam based R&D Company specializing in *Open-Source* and *Commercial* Voice Technologies deployed and trusted by thousands of businesses worldwide, include large telephony and network operators, voice service carriers, voip service providers, cloud service providers, call center operators and voice equipment vendors.

QXIP Capture Technologies are natively implemented in all major OSS voip platforms such as *Kamailio*, *OpenSIPS*, *FreeSWITCH*, *Asterisk*, *RTPEngine* and many tools such as *sipgrep*, *sngrep*.





Elephant in the server room







VoIP & RTC Problems

- Connectivity Problems
- Call Quality Problems
- Security Problems
- Multi Equipment management
- Hard to troubleshoot
- Mission Critical Application





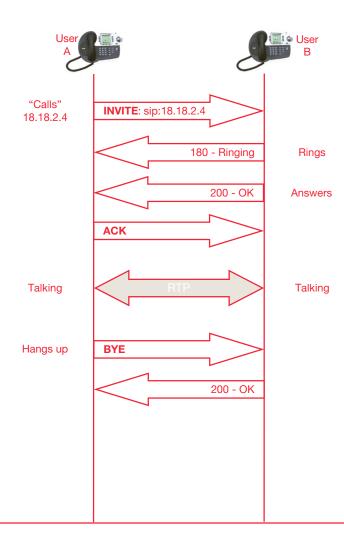
VoIP Protocols

- SIP / WEBRTC / TLS Signaling Protocols
- RTP Media Protocol
- RTCP Real Time Control Protocol





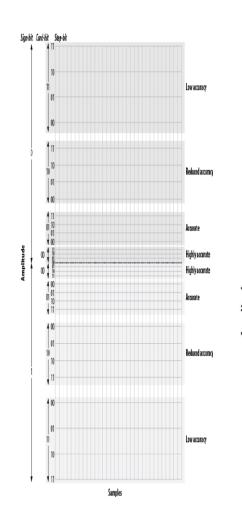
SIP Flows - Basic



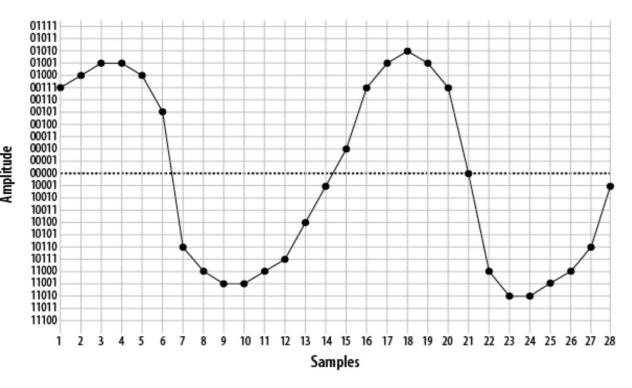




RTP - How Digital Audio Works



00111 01000 01001 01001 01000 00101 10110 11000 11001 11001 11000 10111 10100 10001 00010 00111 01001 01001 01001 00111 00000 11000 11010 11010 11001 11000 10110 10001







RTCP-RTP (Quality) Control Protocol

```
"event": {
     "media": "audio",
     "base": 48000,
     "lsr": 37971368,
     "lost": 0,
     "lost-by-remote": 0,
     "jitter-local": 18940,
     "jitter-remote": 0,
     "packets-received": 39,
     "packets-sent": 40,
     "bytes-received": 6708,
     "bytes-sent": 7280
```



1 RTCP packet per RTP stream each 5-10 seconds





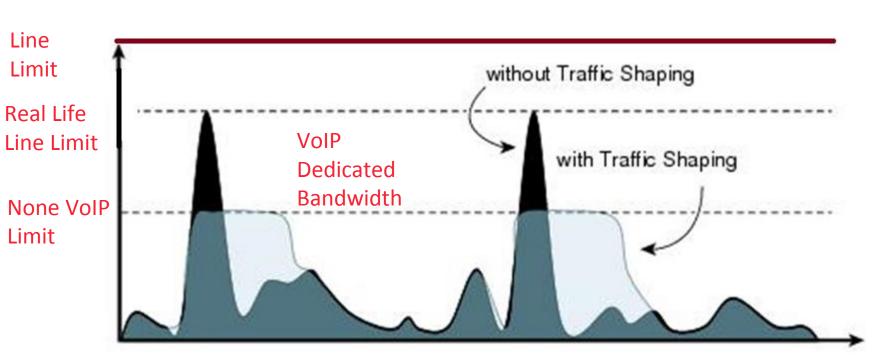
How Can Mikrotik Push my Voip Packets?







Traffic Shaping Concept



Losers make promises they often break.
Winners make commitments they always keep.





VolP QoS Best Practice

- Address List Maintenance
- Connection Marking
- Packet Marking
- Queues configuring







Media servers Import Script

Script < Media Serve	rsImport>								
Name:	MediaServersImport								
Owner:	admin								
Policy:									
	✓ read ✓ write ✓ policy test								
	password sniff sensitive romon								
	dude london								
Last Time Started:									
Run Count:	0								
	p://mikrotik.XXXX.com/script/MediaServer.rsc" mode=http "Error on downloading OF MediaServer Script> http://mikrotik.XXXX.com/script/IPList/MediaServer.rsc"};								
/import file-name=N	MediaServer.rsc "Error on Runing MediaServer.rsc"};}								





Media servers HTTP Response

```
:put " * *****Add XX.XX.XX.XX MGW 01 to MediaServer**********
:do {
/ip firewall address-list add address=XX.XX.XX.XX
list=MediaServerList comment="MGW 01 ->MediaServer"
   } on-error={ :put " Failed to add XX.XX.XX.XX MGW 01 to
MediaServer probably already there "};
:put " * *****Add XX.XX.XX.XX MGW 01 to MediaServer*******
:do {
/ip firewall address-list add address=XX.XX.XX.XX
list=MediaServerList comment="MGW 01 ->MediaServer"
   } on-error={ :put " Failed to add XX.XX.XX MGW 01
MediaServer probably already there "};
```





Connection Marking

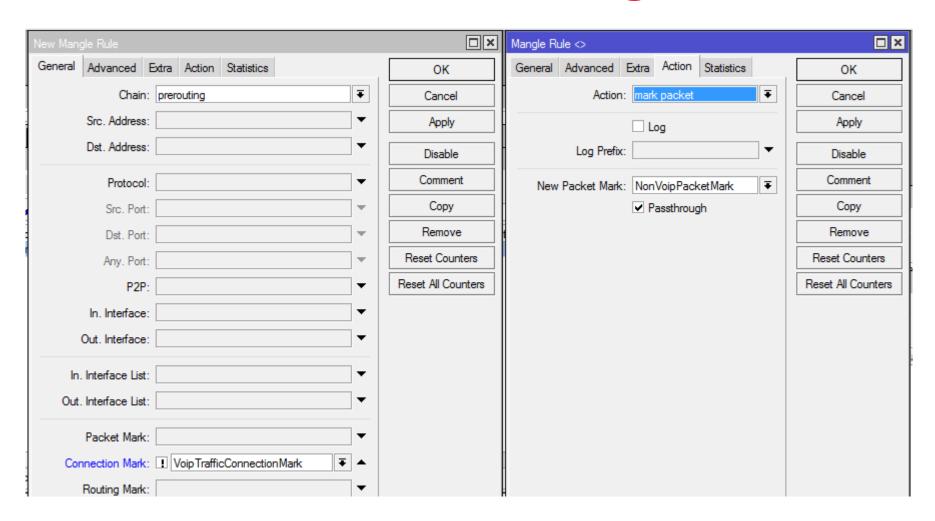
w Mangle Rule	
eneral Advanced Extra Action Statistics	
Chain: prerouting	
Src. Address:	
Dst. Address:	
Protocol: Udp	
Src. Port:	
Dst. Port: 5060,5061,10000-32767	
Any. Port:	
P2P:	
In. Interface:	
Out. Interface:	
In. Interface List:	
Out. Interface List:	
Packet Mark:	
Connection Mark:	
Routing Mark:	
Routing Table:	
Connection Type:	
Connection State: ☐ invalid ☐ established ☐ related ☑ new	

General Advanced Extra Action Statistics				
Src. Address List: Media ServerList	→			
Layer7 Protocol:				
Contant:	— •			
General Advanced Extra Action Statistics	ОК			
Action: mark connection ▼	Cancel			
Log	Apply			
Log Prefix: ▼	Disable			
New Connection Mark: VoipTrafficConnectionMark ▼	Comment			
✓ Passthrough	Сору			
	Remove			
	Reset Counters			





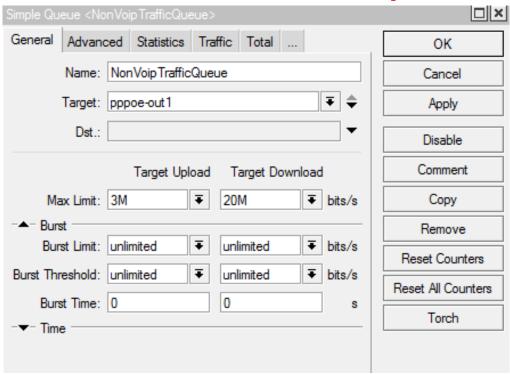
Pocket Marking







Queue for non voip traffic









VoIP Monitoring Using Mikrotik



Switch Layer mirroring

Good Performance

Packet Sniffer stream

Bad Performance

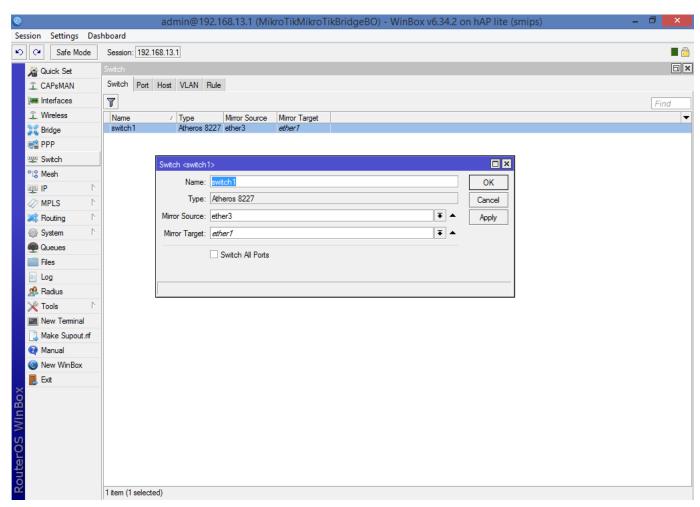
Pcap File analytics

Ugly from any perspective





Switch Layer mirroring

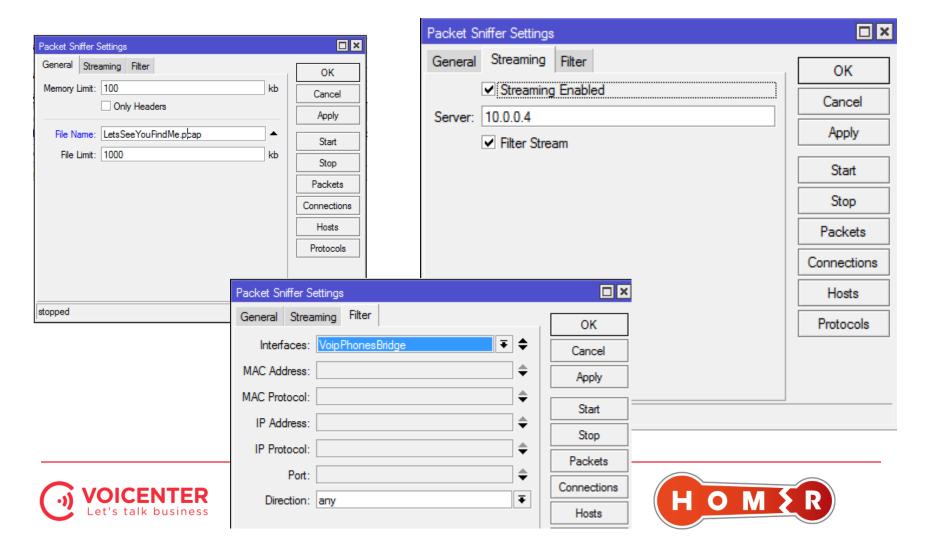




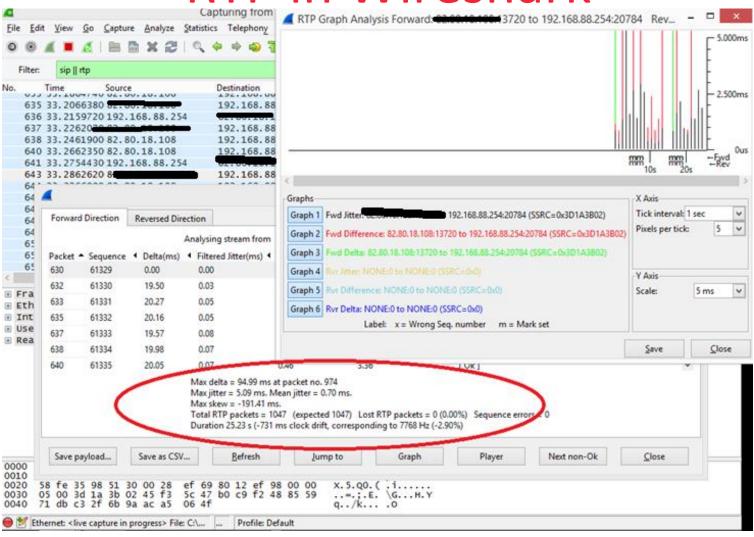


Packet Sniffer Setup

TZSP - Packet Sniffer encapsulation



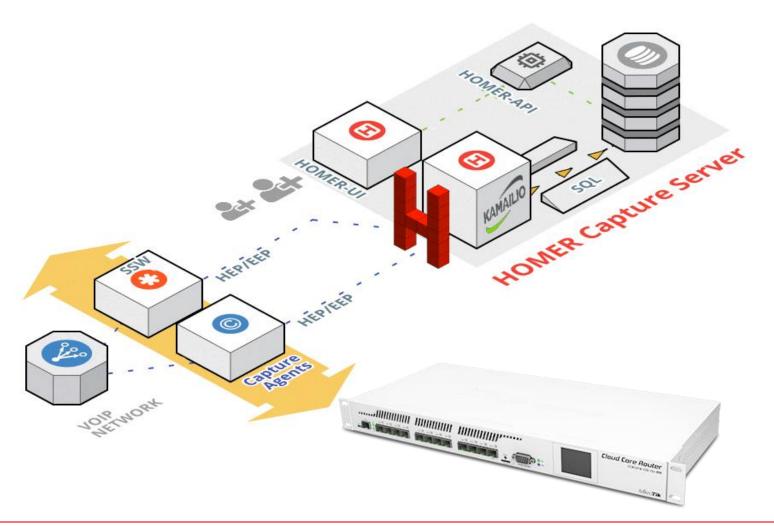
RTP in Wireshark







SIP + RTP in HOMER Cloud







CaptAgent / RTPAgent

modular capture agent supporting multiple sockets, protocols and transport methods

- TZSP support for MicroTik
 Packet Sniffer encapsulation
- HEP support for HOMER Cloud Analytics and Monitoring

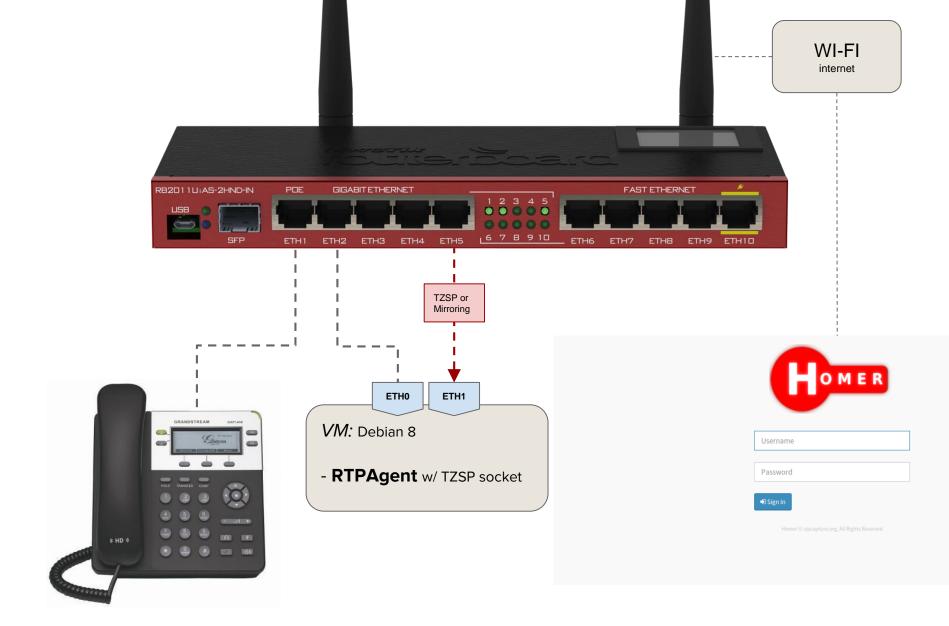
```
"CORRELATION_ID": "56a211936328-fgbtmubkimot",
"RTP_SIP_CALL_ID": "56a211936328-fgbtmubkimot",
"DELTA":19.980.
"JITTER":0.023,
"REPORT_TS":1453461919,
"TL_BYTE":0,
"SKEW":-0.180,
"TOTAL_PK":510,
"EXPECTED_PK":510,
"PACKET_LOSS":0,
"SEQ":0,
"MAX_JITTER":1.892, "MEAN_JITTER":0.126,
"MAX_DELTA":35.547, "MAX_SKEW":-15.615,
"MIN_MOS":4.385, "MEAN_MOS":4.394, "MOS":4.394,
"RFACTOR":92.449, "MIN_RFACTOR":92.013, "MEAN_RFACTOR":92.444,
"SRC_IP":"192.168.178.34", "SRC_PORT":58320,
"DST_IP":"192.168.60.70","DST_PORT":32728,
"SRC_MAC":"00-04-13-29-64-22","DST_MAC":"34-31-C4-38-24-0D",
"CODEC_PT":9,"CLOCK":8000, "CODEC_NAME":"g722", "DIR":1,
"REPORT_NAME": "192.168.178.34:58320", "PARTY":0,
"TYPE": "PERIODIC"
```





















UI: Tracking Calls and Sessions in Real Time

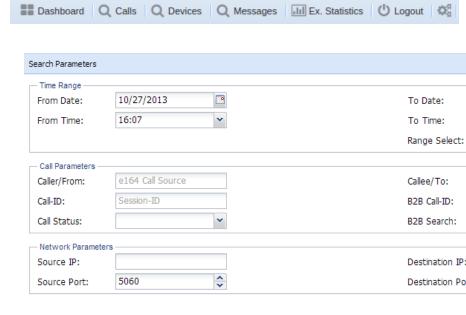
The **CALL SEARCH** functionality is one of the most used tools to locate, analyze and extract present and past call sessions.

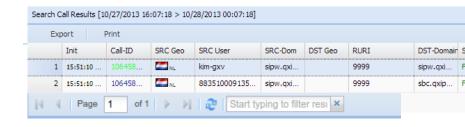
The Call Search tool obeys the global **TIME-RANGE** as its primary filter, extended by a customizable number of user defined parameters targeting session headers and parameters (more in the next slide)

The **SEARCH** functionality also offers programmable "Search Profiles" per group used to automatically include and match multiple dialing patterns (*international/national*) prefixes (00/+) or routing prefixes from a bare number with no additional user interaction required.

To maximize the platform's full potential, a "two-tier" approach is also possible and suggested, with a first initial group of results returned by the backend and complex filtering by any field can be performed client-side using advanced regex patterns and wildcard matching.

NEXT: TRACKING CALLS AND SESSIONS









UI: Tracking Calls and Sessions in Real Time

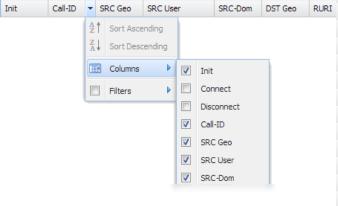
Search Results for Calls will be returned in a table, ordered by timestamp and ready to use

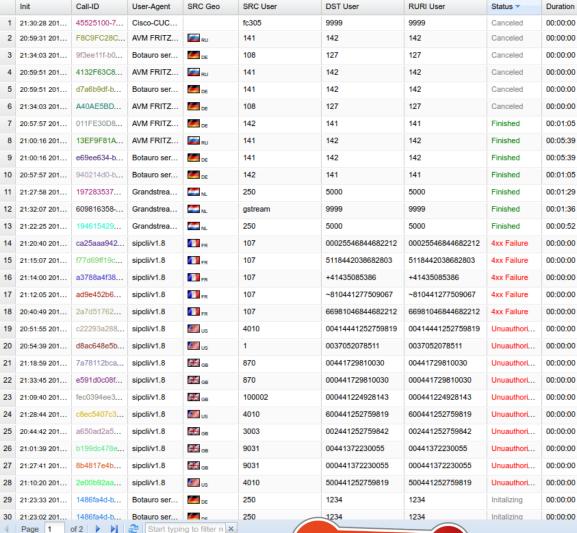
(continued)

PCAPTURE core is session aware and can display call status in realtime and aggregate all messages, statistics and logs in a single object, with automatic correlation to any other connected B2BUA legs

To investigate Session Details, just click on a result

Search Results table columns can be configured based on user preference to show or conceal any of the available session and protocol parameters.





Q Devices

Q Messages III Ex. Statistics

Dashboard Q Calls















UI: Tracking Calls and Sessions Details

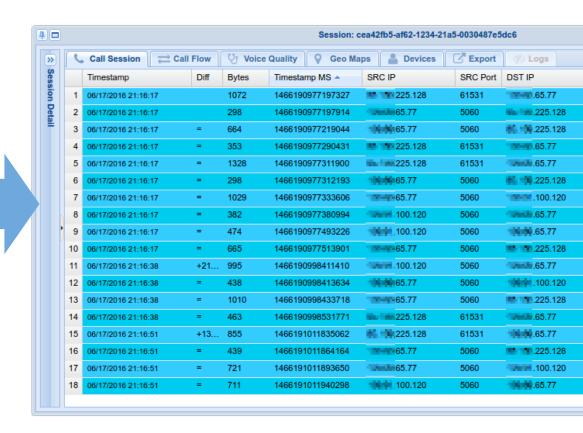
(continued)

Session Details will be returned when selecting one or more result rows. The API will automatically fetch all correlated call data in the current Time Range.

The "Session Detail" window features several Tabs presenting available correlated information about the Session (or Session Group) being displayed.

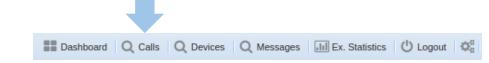
Call Session tab presents packets in Shark-View mode

Each packet and message can be inspected in any display mode by simply clicking the corresponding row or object to reveal the full original payload data









UI: Tracking Calls and Sessions Details

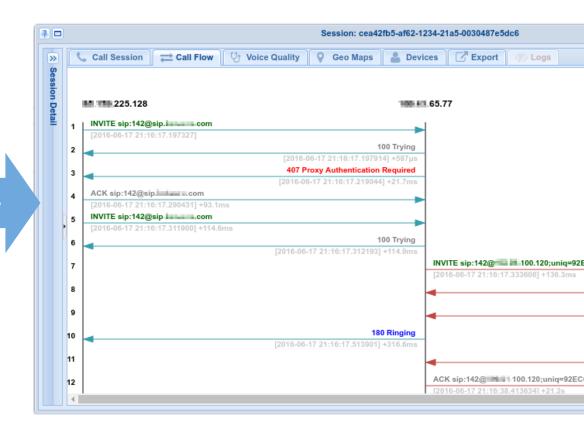
(continued)

Sessions involving several devices with hops traversing multiple systems can quickly get complex

PCAPTURE's *Call Flow* tool automatically correlates hosts and messages and presents them in an easy to interpret format well familiar to voice experts of all seasons and capable of handling unlimited legs

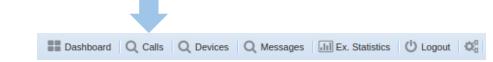
Call Flow tab presents packets in Signaling Flow mode

Each packet and message can be inspected in any display mode by simply clicking the corresponding row or object to reveal the full original payload data









UI: Tracking Calls and Sessions Details

(continued)

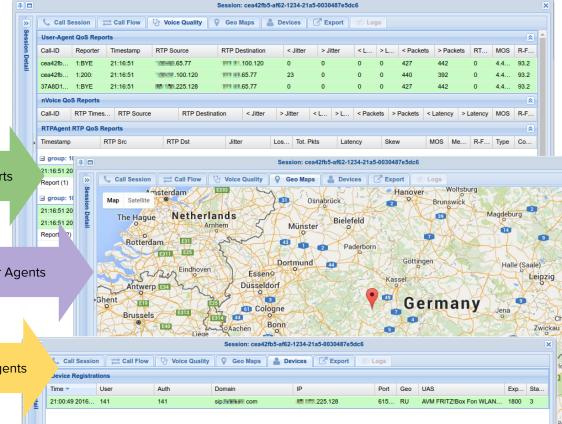
When additional data about the Session being inspected is available, PCAPTURE will automatically present it to the end-user without any interaction.

The **Export Tab** provides dynamic methods to Save, Archive or Share the current set in different formats.

Voice Quality tab presents stream RTP-RTCP quality reports

Geo Maps tab presents the approx. IP Geolocation of User Agents

Devices tab presents details about Registered SIP User-Agents







UI: Tracking Media Session Quality

(continued)

The **Voice Quality** tab presents metrics related to media sessions as reported by User-Agents, Passive Network Probes and Media Control Protocols, providing correlated data useful when analyzing complex RTP media paths between SIP Endpoints

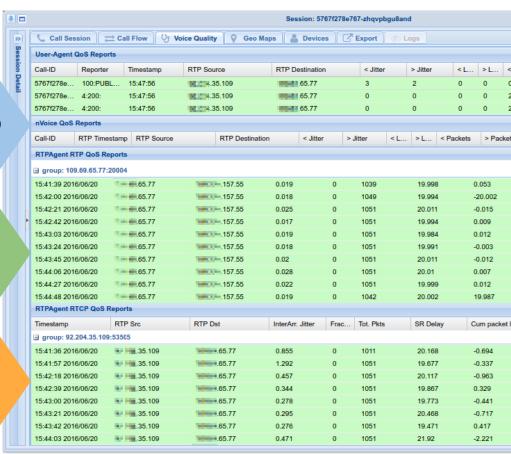
User-Agent generated media reports (RTCP-XR, X-RTP-Stat, P-RTP)

CAs can automatically report for monitored streams at variable or static rates, with each report carrying all RTP metrics and **MOS**

RTP Report from passive network analysis with granular metrics

CAs can also capture and aggregate RTCP control protocol messages and extract metrics and statistics to calculate a **MOS**

RTCP Report from analysis and aggregation of User-Agent reports



Dashboard Q Calls Q Devices Q Messages III Ex. Statistics U Logout 💸







UI: Tracking Registrations and Devices

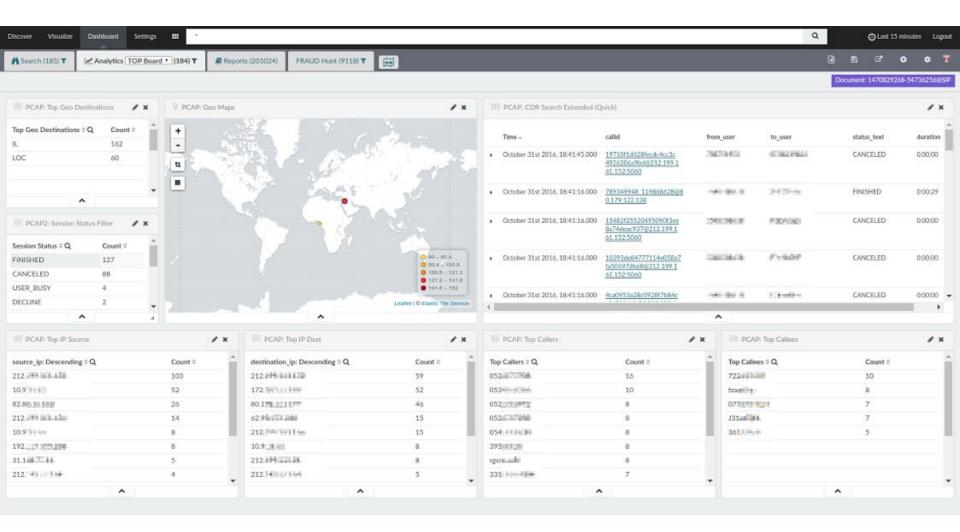
PCAPTURE features a dedicated tool for *Searching* and *Filtering* registration with *Expiration tracking*, integrated with Call Search tools:

nit	Call-ID	UAC	SRC Geo	User	Auth	Domain	Status	Expiration	Proto	SRC IP	SP	DST IP	DP
22:30:45 2016-06-19	1292262599	ACN IRIS X	DE DE	145	145	sip. com	Registered	Expired	SIP/UDP	36.186	407	.65.77	5060
22:30:52 2016-06-19	219111350	Grandstream	ML NL	lab	lab	sipvenet	Registered	Expired	SIP/UDP	44.127	5062	6.157.55	5060
22:31:23 2016-06-19	57EF618711	AVM FRITZ!	RU	141	141	sipcom	Registered	00:08:12	SIP/UDP	225.128	615	.65.77	5060
22:32:32 2016-06-19	386d439b1b	snom360/8	DE DE	123456789_pc	123456789_pc	sip	Registered	Expired	SIP/UDP	20.17	2048	.65.77	5060
22:33:43 2016-06-19	491551672	ACN IRIS X	MD MD	147	147	sip	Registered	Expired	SIP/UDP	161.3	187	Main #1.65.77	5060
22:35:02 2016-06-19	386d439b1b	snom360/8	DE DE	123456789_pc	123456789_pc	sip	Registered	Expired	SIP/UDP	20.17	2048	.65.77	5060
22:35:27 2016-06-19	818978511	ACN IRIS X	DE DE	204	204	sip. com	Registered	Expired	SIP/UDP	100.120	114	MIN #1.65.77	5060
22:35:37 2016-06-19	918051609	Grandstream	ML NL	gstream	gstream	sipvenet	Registered	-	SIP/UDP	44.127	550	6.157.55	5060
22:35:54 2016-06-19	219111350	Grandstream	ML NL	lab	lab	sipvenet	Registered	Expired	SIP/UDP	44.127	5062	6.157.55	5060
22:37:33 2016-06-19	386d439b1b	snom360/8	€	123456789_pc	123456789_pc	sip	Registered	Expired	SIP/UDP	20.17	2048	Main #1.65.77	5060
22:38:15 2016-06-19	1292262599	ACN IRIS X	€	145	145	sip. com	Registered	00:00:03	SIP/UDP	36.1			
22:38:40 2016-06-19	3024034456	S450 IP/022	DE DE	101	101	sip	Registered	Expired	SIP/UDP	36.1 »	€ Ca	II Session	l Flow
22:39:54 2016-06-19	AC6A5EE4E	AVM FRITZ!	WA.	139	139	sip	Registered	00:16:42	SIP/UDP	1.19	Tim	estamp	Diff
22:40:03 2016-06-19	2086355230	Grandstream	ML NL	lab	lab	sipventet	Registered	Expired	SIP/UDP	44.1		7/2016 21:16:17	
22:40:03 2016-06-19	386d439b1b	snom360/8	DE DE	123456789_pc	123456789_pc	sip	Registered	Expired	SIP/UDP	20.1		7/2016 21:16:17	_
22:40:06 2016-06-19	1533980629	Grandstream	ML NL	gstream	gstream	sipventet	Registered	-	SIP/UDP	MFTTT 44.1		17/2016 21:16:17	=
22:40:37 2016-06-19	918051609	Grandstream	ML NL	gstream	gstream	sipvenet	Registered	-	SIP/UDP	MF TTT 44.1	5 06/1	7/2016 21:16:17	=
22:40:56 2016-06-19	219111350	Grandstream	ML NL	lab	lab	sipventet	Registered	Expired	SIP/UDP	MFTTT 44.1		7/2016 21:16:17	=
22:41:12 2016-06-19	491551672	ACN IRIS X	€ MD	147	147	sip	Registered	00:03:00	SIP/UDP	161.		7/2016 21:16:17	=
22:42:33 2016-06-19	386d439b1b	snom360/8	DE DE	123456789_pc	123456789_pc	sip	Registered	Expired	SIP/UDP	20.1		7/2016 21:16:17	-
22:42:57 2016-06-19	818978511	ACN IRIS X	DE DE	204	204	sip	Registered	00:04:45	SIP/UDP	100	10 06/1	7/2016 21:16:17	=
22:45:03 2016-06-19	386d439b1b	snom360/8	DE DE	123456789_pc	123456789_pc	sip	Registered	Expired	SIP/UDP	20.1	11 06/1	7/2016 21:16:38	+21
2:45:37 2016-06-19	918051609	Grandstream	ML NL	gstream	gstream	sipventet	Registered	-	SIP/UDP	44.1		7/2016 21:16:38	=
												7/2016 21:16:38	=
												17/2016 21:16:38	+13
												17/2016 21:16:51	=





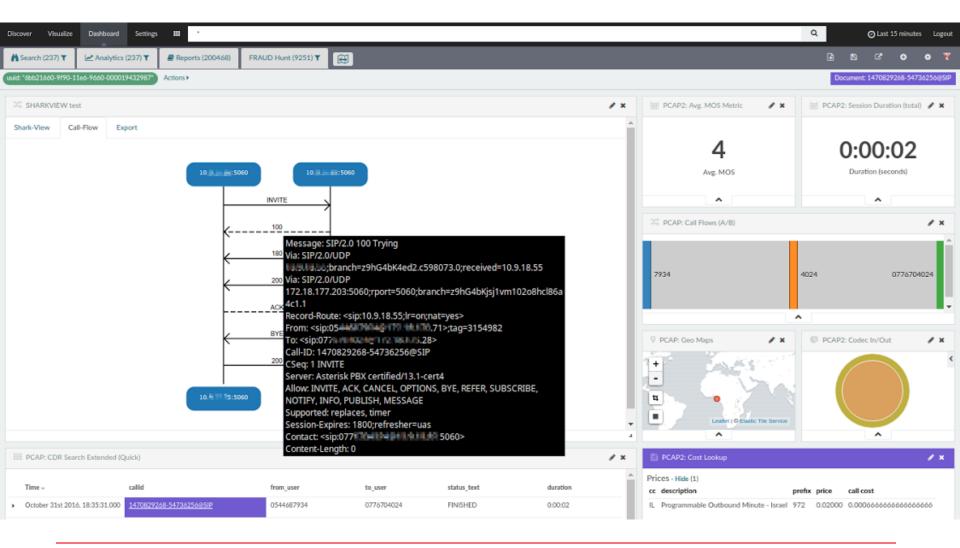
Voice - Top Board







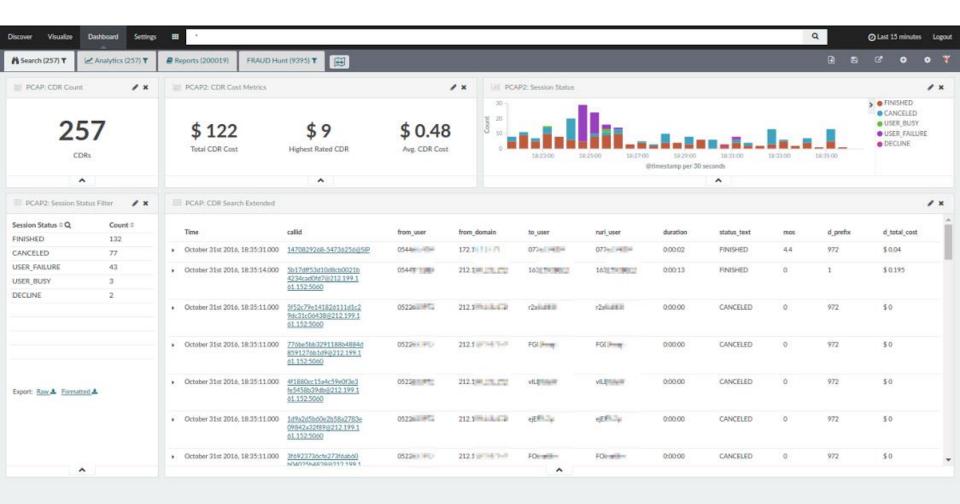
SESSION/PROTOCOL DRILLDOWN:







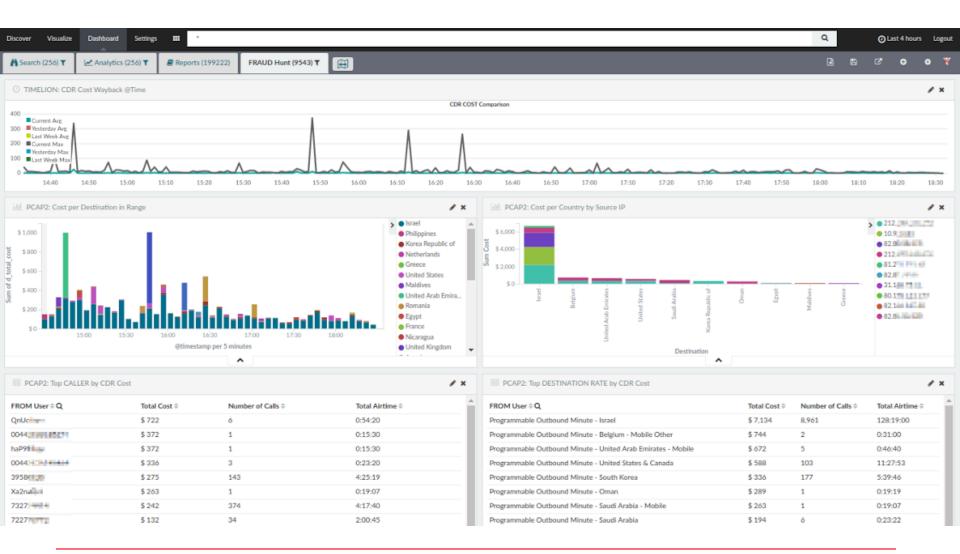
CDR SEARCH & FILTER:







FRAUD DETECTION PATTERNS:







Alerting and Fraud detection

SA: Time Series Visualization in Kibana

Complex reports can be created leveraging all available metrics and time series, including comparisons across different data ranges:

