

Manual of the Aarenet VoIP Switch Support Tools

Introduction

The VoIP Switch Administrators, Operators, Supporters find here information about the Aarenet VoIP Switch "On Board" tools for analyzing and supporting general VoIP Switch issues and customer problems:

- ◊ The ConfigCenter "Support Log" displays the VoIP Switch internal activities
- ◊ The ConfigCenter "Traces" displays the signaling between the VoIP Switch and external devices
- ◊ The ConfigCenter "Call Data" lists the CDR of all incoming or outgoing connections or connection attempts. The "Call Data" provide shortcuts to SIP, SDP trace and RTP statistic information of a single call.
- ◊ The ConfigCenter "Address Registration" displays if a SIP device or MGCP MTA has registered a telephone number
- ◊ The ConfigCenter "System Components" displays the state and activity of the VoIP Switch components
- ◊ The ConfigCenter "Channels" displays the state of connections
- ◊ The ConfigCenter "System Utilization" displays a statistical overview of the VoIP Switch resource utilization

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VoIP Switch ConfigCenter Support Tools

The ConfigCenter Support Log

The "Support Log" provides the supporter with information from the internal processes of the ServiceCenter:

- ◊ Registration
- ◊ Connection setup, release and exceptions
- ◊ Call Routing
- ◊ Used Ruleset
- ◊ Emergency calls
- ◊ etc

The "Support Log" provides filters for:

- ◊ Time based selection: From ? Until, From ? Duration
- ◊ Text filter
- ◊ Registration events
- ◊ Call events
- ◊ etc.

The "Support Log" has a limited history. The history may last from a few hours up to some days. The length of the history may be different from VoIP switch to VoIP switch and depends on the length of log files and amount of logging events.

Note

The "Support Log" is tenant sensitive. This means a supporter of tenant A is not able to see events of tenant B!

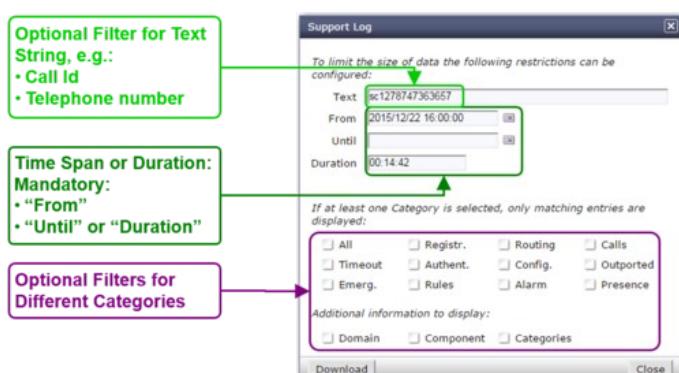
Navigate to the "Support Log"

ConfigCenter:

- Menu "Support"
- Menu "Support Log"

Get a "Support Log"

Dialog: "Support Log":



When the dialog "Support Log" opens it contains by default in "From" the actual date/time (-5min) and in "Duration" a duration of 5min:

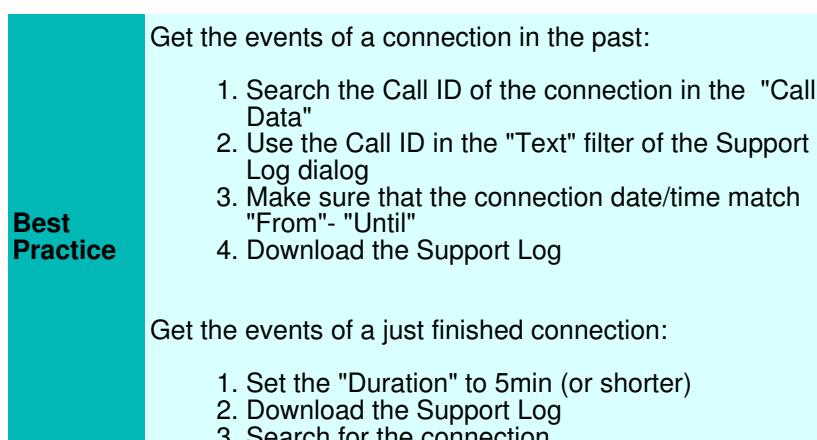
1. Click the Button [Download]
2. Via HTTP an ASCII formatted file with the last 5 minutes will be downloaded

Retrieving a "Support Log" in the past:

1. Insert the in "From" the desired start date/time
2. Insert in "Duration" the needed length
3. Press on the PC keyboard the 'Enter' key : The "Until" date/time will be computed
4. Click the Button [Download]

or

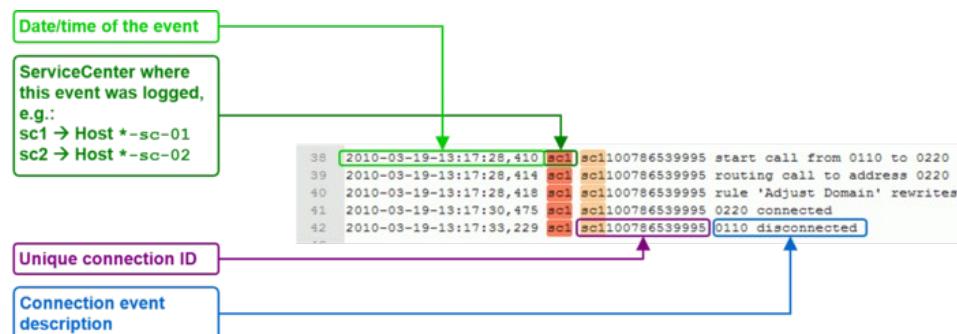
1. Insert the in "From" the desired start date/time
2. Insert the in "Until" the desired stop date/time
3. Press on the PC keyboard the 'Enter' key: The "Duration" will be computed
4. Click the Button [Download]



Interpretation of a "Support Log"

The interpretation of a "Support Log" is quite easy and straight forward. With a little experience one will be soon familiar with the interpretation.

Interpretation and example of a call setup and release:



ConfigCenter Trace

The "Trace" provides the supporter with information from the message traffic between the VoIP switch and external VoIP devices, such as PSTN gateway, SIP CPE, SIP or MGCP telephones.

The "Trace" contains:

- ◊ Session Initiation Protocol SIP registration and connection signaling messages
- ◊ Media Gateway Control Protocol MGCP audit and endpoint control messages
- ◊ Session Description Protocol SDP streaming media initialization parameters

The "Trace" provides filters for:

- ◊ Time based selection: From ? Until, From ? Duration
- ◊ Text filter

The "Trace" has a limited history. The history may last from a few hours up to some days. The length of the history may be different from VoIP switch to VoIP switch and depends on the length of log files and amount of logging events.

The interpretation of a "Trace" (PCAP formatted file) has to be done in an external application like Wireshark network protocol analyzer. Wireshark offers deep and rich VoIP analysis .

Note The "Trace" is **not** tenant sensitive. This means a supporter of tenant A is able to see signaling messages of tenant B!
Due to this open display of information it may be possible that the "Trace" is not available for the supporters and operators on a multi tenant VoIP Switch.

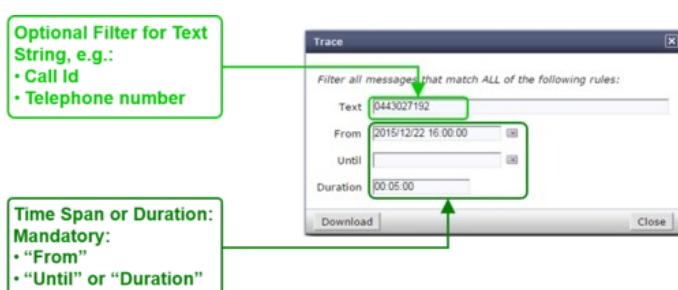
Navigate to the "Trace"

ConfigCenter:

- ➔ Menu "Support"
- ➔ Menu "Trace"

Get a "Trace"

Dialog: "Trace":



When the dialog "Trace" opens it contains by default in "From" the actual date/time (-5min) and in "Duration" a duration of 5min:

1. Click the Button [Download]

2. Via HTTP an PCAP formatted file with the last 5 minutes will be downloaded

Retrieving a "Trace" in the past:

1. Insert the in "From" the desired start date/time
2. Insert in "Duration" the needed length
3. Press on the PC keyboard the 'Enter' key: The "Until" date/time will be computed
4. Click the Button [Download]

or

1. Insert the in "From" the desired start date/time
2. Insert the in "Until" the desired stop date/time
3. Press on the PC keyboard the 'Enter' key: The "Duration" will be computed
4. Click the Button [Download]

Best Practice

Get the events of a connection in the past:

1. Search the connection in the "Call Data"
2. Click the Button [Trace]

Get the events of a just finished connection:

1. Set the "Duration" to 5min (or shorter)
2. Download the Trace
3. Search for the connection

Interpretation of a "Trace"

The interpretation of a "Trace" needs experience!

For more information:

- ◊ See also article "Brief Tutorial of the SIP Signaling and SDP Media Protocols"
- ◊ Get a Wireshark training

Example of a Wireshark call capture, SIP setup and release:

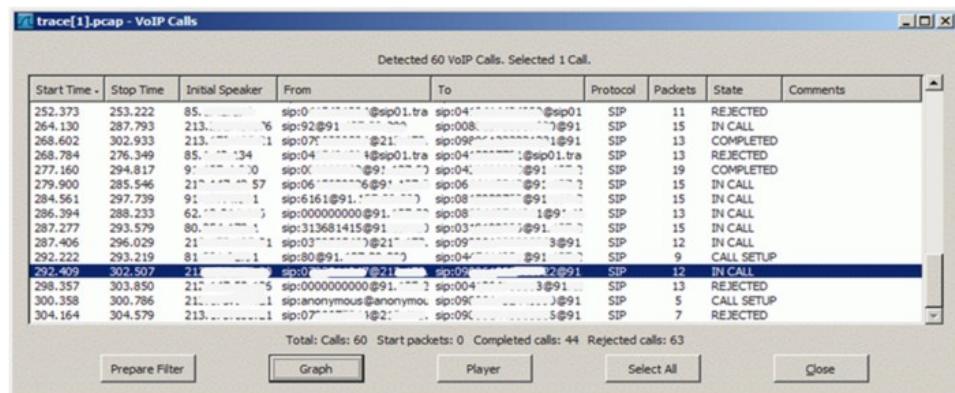
Frame 64264: 794 bytes on wire (6352 bits), 794 bytes captured (6352 bits)
Ethernet II, Src: 00:00:00:00:5c:01 (00:00:00:00:5c:01), Dst: 00:00:00:00:00:00 (00:00:00:00:00:00)
Internet Protocol Version 4, Src: 81.100.100.100 (81.100.100.100), Dst: 81.100.100.100 (81.100.100.100)
User Datagram Protocol, Src Port: 5062 (5062), Dst Port: 5060 (5060)
Session Initiation Protocol (INVITE)
Request-Line: INVITE sip:041...@81.100.100.100 SIP/2.0
Message Header
Via: SIP/2.0/UDP 81.100.100.100:5062;branch=z9hG4bKdfcbfbe3adb48886
Max-Forwards: 70
From: <sip:041...@81.100.100.100>;tag=f82a933b62
To: <sip:041...@81.100.100.100>
Call-ID: 2132cad34d6be47
CSeq: 28929 INVITE
Contact: <sip:041...@81.100.100.100:5062;transport=udp>
Supported: replaces
User-Agent: Patton SN4960 4E60V UI 00A0BA01CE73 R6.T 2013-03-14 H323 RBS MST SIP Stack/4.1.12.18
Content-Type: application/sdp
Content-Length: 294
Message Body
Session Description Protocol
Session Description Protocol Version (v): 0
Owner/Creator, Session Id (o): MxSIP 0 559 IN IP4 81.100.100.100
Session Name (s): SIP Call
Connection Information (c): IN IP4 81.100.100.100
Time Description, active time (t): 0 0
Media Description, name and address (m): audio 4926 RTP/AVP 8 0 18 125 101
Media Attribute (a): rtpmap:8 PCMA/8000
Media Attribute (a): rtpmap:0 PCMU/8000
Media Attribute (a): rtpmap:18 G729/8000
Media Attribute (a): rtpmap:125 CLEARMODE/8000
Media Attribute (a): rtpmap:101 telephone-event/8000
Media Attribute (a): fntp:18 annexb-no
Media Attribute (a): fntp:101 0-16
Media Attribute (a): sendrecv

Example of a Wireshark call list:

Navigate in Wireshark:

- Menu "Statistics"
- Menu "VoIP Calls"

Wireshark dialog where all calls are listed of the actual trace:

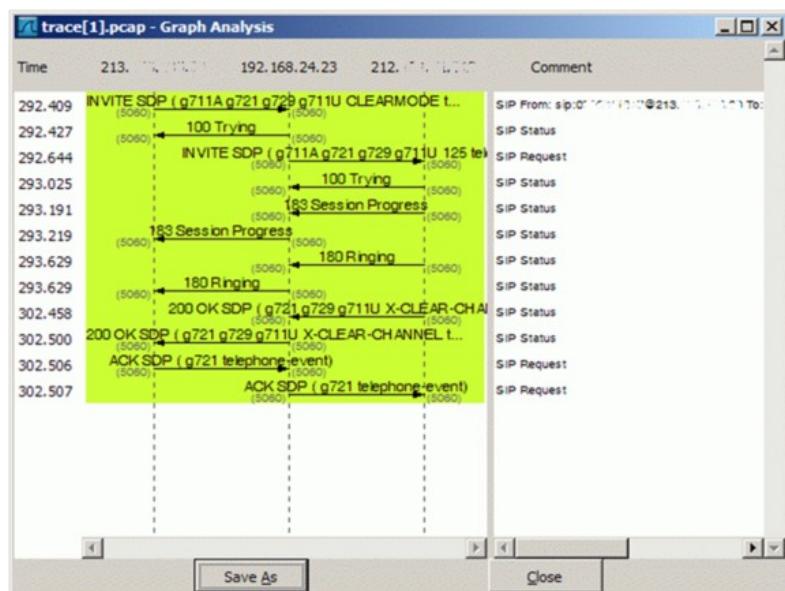


Example of a Wireshark call flow:

Navigate in Wireshark:

- Menu "Statistics"
- Menu "VoIP Calls"
- Select the call of interest
- Click Button [Graph]

Wireshark dialog where the message flow is shown of the selected call:



The ConfigCenter Call Data

The "Call Data" lists the CDR of all incoming or outgoing connections or connection attempts. Extended filters enable the supporter to search for specific calls. The filters can be combined with logical AND.

Filter CDRs according:

- ◊ Call start and end date/time
- ◊ Call duration
- ◊ Call charges
- ◊ Telephone number of caller and/or callee.
- ◊ Tenants & account
- ◊ Price list attributes "Destination Type" & "Destination"

The "Call Data" has a limited history. The length of the history may be different from VoIP switch to VoIP switch and depends on the CDR storage length in the date base.

Selected CDR details allow direct access to the information of:

- ◊ SIP Trace:
The SIP message contents of this specific connection or call attempt is shown. For the interpretation of the trace consult the article "Brief Tutorial of the SIP Signaling and SDP Media Protocols", chapter "Knowhow SIP Signaling" .
- ◊ RTP/RTCP Media:
The RTP/RTCP information and statistics of this specific connection or call attempt is shown. For the interpretation of the media information consult the article "Brief Tutorial of the SIP Signaling and SDP Media Protocols", chapter "Knowhow Media Stream" .

Note

- The "Call Data" has a limited history. The length of the history may be different from VoIP switch to VoIP switch and depends on the CDR storage length in the date base.
- Not all filter options may be available on the VoIP Switch.
- The "Call Data" is tenant sensitive. This means a supporter/operator of tenant A is not able to see events of tenant B!

Warning

Depending on the settings of a VoIP system it may be possible to change values in CDR.

Changing a CDR's contents may be a legal violation in the country of operation of the VoIP Switch!

Navigate to the "Call Data"

ConfigCenter:

- Menu "Rating"
- Menu "Call Data"

Get the "Call Data"

Dialog: "Call Data":

Filter for ranges of:

- Time
- Duration
- Charge

Hint:

Insert at start "Duration" "00:00:00" for displaying call attempts.

Filter for telephone numbers

Export the displayed CDRs in a MS Excel file.

Do not use for support reasons!

Start the searching

By clicking on the line of a CDR a dialog pops up, which provides a) more details of the connection and b) one click access to the call's SIP trace and media RTP/RTCP information and statistics:

Call details

CDR: sc1336652568403

Call ID	sc1336652568403
Charges Account	0.00
Charges Tenant	0.00
Time	2017/08/24 10:54
Duration	00:00:00
Restricted	no
Calling Number	04-12-051
Called Number	051-1
State	canceled
Leg	A
CLIP	0449980912
Destination	Switzerland-058 (Corporate)
Dest. Type	VAS
Postrating	
Rate Import Time	
Billing Info	

Do not use for support reasons!

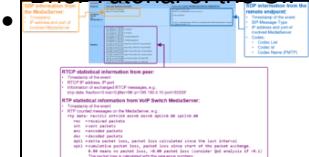
Get a trace of this call as PCAP file.

Get the RTP information and statistics of this call as HTML file.



For the interpretation of the trace consult the article:

"Brief Tutorial of the SIP Signaling and SDP Media Protocols", chapter "Knowhow SIP Signaling"



For the interpretation of the media information consult the article:

"Brief Tutorial of the SIP Signaling and SDP Media Protocols", chapter "Knowhow Media Stream"

The ConfigCenter Address Registration

The ConfigCenter "Address Registration" displays if a SIP device or MGCP MTA has registered the telephone number. The supporter finds the following information of the registering devices:

- ◊ Type of registration, SIP, notifications, presence, etc
- ◊ IP address
- ◊ SIP user agent
- ◊ Registration time left.

Registrations can be de-registered on the VoIP Switch by force.

Hint:

The device cannot be informed that it was de-registered on the VoIP Switch. That means you have to wait until it re-registers automatically or force the device manually to re-register.

Navigate to "Registrations"

ConfigCenter:

→ Menu "Addresses"

or

→ Menu "Accounts"

→ Click on the line of the desired account

→ Click on the right arrow at "Addresses"

For details:

→ Click on the line of the desired address

→ Click on the right arrow at "Registration"

Interpretation of "Registrations" Information

Display of "Addresses" and registration overview:

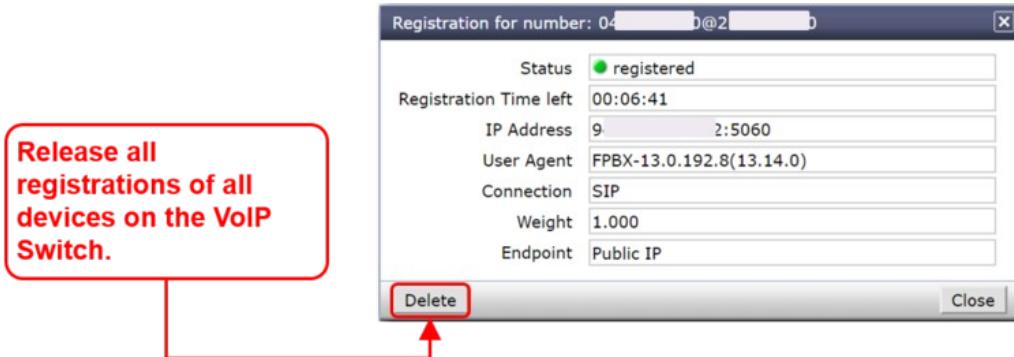
The screenshot shows the 'Addresses of Account' dialog box. On the left, a green box contains a status summary:

Status and type of registration:
● Active registration
● No registration
Note:
If no icon is shown then more than 100 addresses are listed.

The main table lists addresses with their registration status. A green arrow points to a 'sip' entry, and a red arrow points to a 'mgcp' entry. The table has columns for Number, Endpoint name, and Registration status.

Number	Endpoint name	Registration
0: 10		
0: 1		
0: 2		
0: 3		
0: 4		
0: 5		
0: 6		
0: 7		
0: 8		
0: 9		
0: 10		
0: 11	0@8c	
0: 12	1@8c	
1: 1		
1: 2		

By clicking on the line of an address and then the right arrow at "Registration" a dialog pops up, which provides informations of all registrations of the address:



The ConfigCenter Components

The "Components" displays the state and activity of the VoIP Switch components. The components are the entities of the VoIP Switch that provide all functionality and features. The display is automatically updated every few seconds and shows the actual state and load of every component.

Note On most VoIP Switches the "Components" display is not available for the supporters and operators.

Navigate to "Components"

ConfigCenter:

- Menu "System"
- Menu "Components"

Interpretation of "Components" Information

Display of "Components":

Name of all installed components.

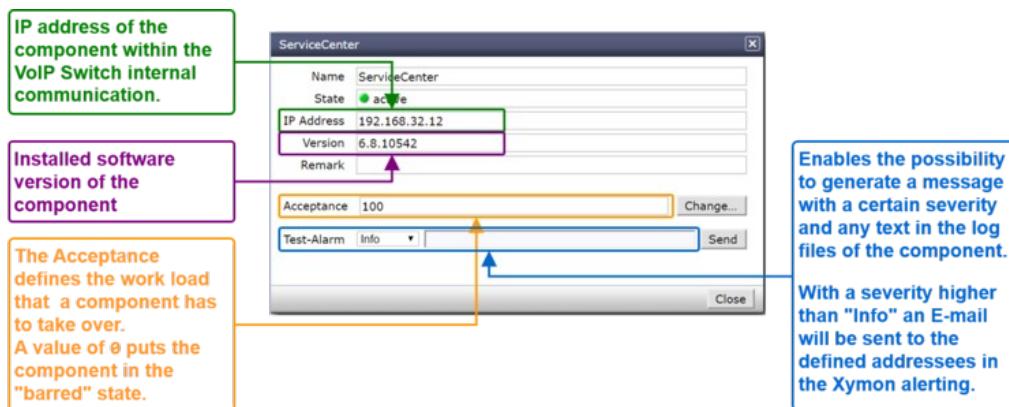
Presents the state of a connection:

- **active:** The component is working correctly and is active.
- **passive:** The component is correctly working and ready for jump in.
- **barred:** The component is correctly working but is suspended from its task.
- **unavailable:** The component is not working correctly!

In the remarks mostly the load of an active component is displayed. In an exceptional situation a short description is given.

Name	State	Remark
HealthCheck 1	● active	
HealthCheck 2	● passive	
LoadBalancer 1	● active	
LoadBalancer 2	● active	
CallBalancer 1	● active	153/598 messages
CallBalancer 2	● passive	0/0 messages
MediaServer 1	● active	919 streams
MediaServer 2	● active	914 streams
MediaServer 3	● active	434 calls
MediaServer 4	● active	453 calls
ServiceCenter 1	● active	
ServiceCenter 2	● active	
MediaCenter 1	● active	
MediaCenter 2	● active	
FaxServer 1	● active	
FaxServer 2	● active	
CallAgent 1	● active	
CallAgent 2	● active	
CdrCollector	● active	
RatingCenter 1	● passive	
RatingCenter 2	● passive	
AdminCenter 1	● active	28 sessions
AdminCenter 2	● active	6 sessions
ConfigCenter 1	● active	
ConfigCenter 2	● active	
Database 1	● active	209 connections
Database 2	● active	95 connections

By clicking on the line of a component a dialog pops up, which provides more informations or enables to send messages or handle the work load of the component:



The ConfigCenter Channels

The ConfigCenter "Channels" is a live display of the current active connections and connection build-up. The administrator can filter and search the connections. If needed a connection can be forced to be released.

Note On most VoIP Switches the "Channels" display is not available for the supporters and operators.

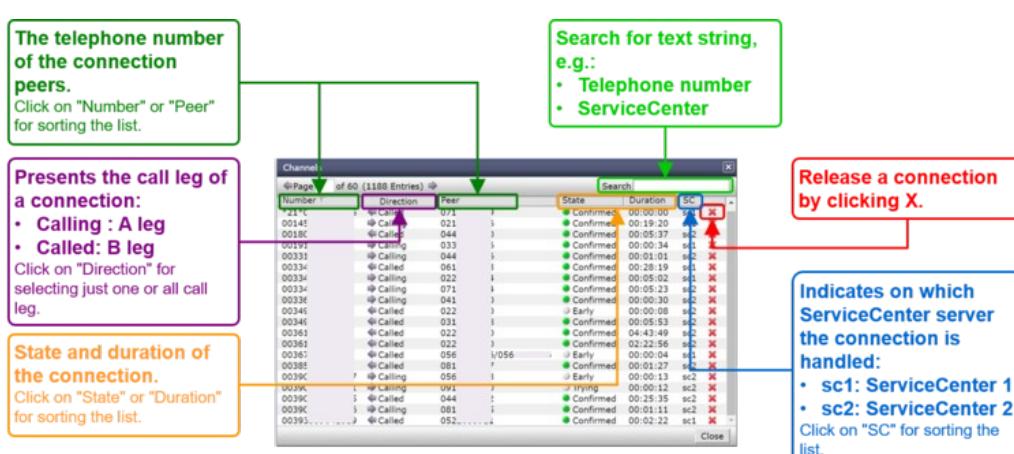
Navigate to "Channels"

ConfigCenter:

→ Menu "Channels"

Interpretation of "Channels" Information

Display of "Channels":



The ConfigCenter System Utilization

The "System Utilization" gives a statistical overview of the VoIP Switch resource utilization:

- ◊ Number of accounts
- ◊ Number of addresses (telephone numbers)
- ◊ Number of registrations
- ◊ etc

Note On most VoIP Switches the "System Utilization" display is not available for the supporters and operators.

Navigate to "System Utilization"

ConfigCenter:

- Menu "System"
- Menu "Utilization"

Interpretation of the "System Utilization" Information

The "System Utilization" provides the numbers of used resources:

