



Dialogic® Brooktrout® SR140 Fax Software with ShoreTel Gateway

Installation and Configuration Integration Note

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1. Scope

This document is intended as a general guide for configuring a basic installation of the **ShoreTel Gateway** for use with Dialogic® Brooktrout® SR140 Fax over IP (FoIP) software platform. The interoperability includes SIP call control and T.38/T.30 media.

This document is not intended to be comprehensive and thus does not replace the manufacturer's detailed configuration documentation. Users of this document should already have a general knowledge of how to install and configure the **ShoreTel Gateway**.

The sample configuration shown and/or referred in the subsequent sections was used for lab validation testing by Dialogic. Therefore, it is quite possible that the sample configuration will not match an exact configuration or versions that would be present in a deployed environment. However, the sample configuration does provide a possible starting point to work with the equipment vendor for configuring your device. Please consult the appropriate manufacturer's documentation for details on setting up your specific end user configuration.

For ease of reference, the Dialogic® Brooktrout® SR140 Fax Software and Dialogic® Brooktrout® TR1034 Fax Boards will sometimes be denoted herein, respectively, as SR140 and TR1034. All references to the SDK herein refer to the Dialogic® Brooktrout® Fax Products SDK.

2. Configuration Details

The following systems were used for the sample configuration described in the document.

2.1 Gateway

Vendor	<i>ShoreTel</i>
Model	<i>ShoreGear 220T1A Switch</i>
Software Version	<i>Controlled Release for Release 10 build 15.6.4207.0</i>
IP Device	<i>Dialogic® Brooktrout® SR140 Fax Software</i>
Protocol to Dialogic® Brooktrout® SR140 Fax Software	<i>SIP</i>
PSTN Device	<i>Dialogic® Brooktrout® DMG 2120</i>
Protocol to PSTN Device	<i>T1</i>
Additional Notes	<i>N/A</i>

2.2 Dialogic® Brooktrout® SR140 Fax Software

Vendor	Dialogic
Model	Dialogic® Brooktrout® SR140 Fax Software
Software Version	SDK 6.2.3
Protocol to Gateway or Call Manager	SIP
callctrl.cfg file	All defaults

2.3 Network System Configuration

The diagram below details the sample configuration used in connection with this document.

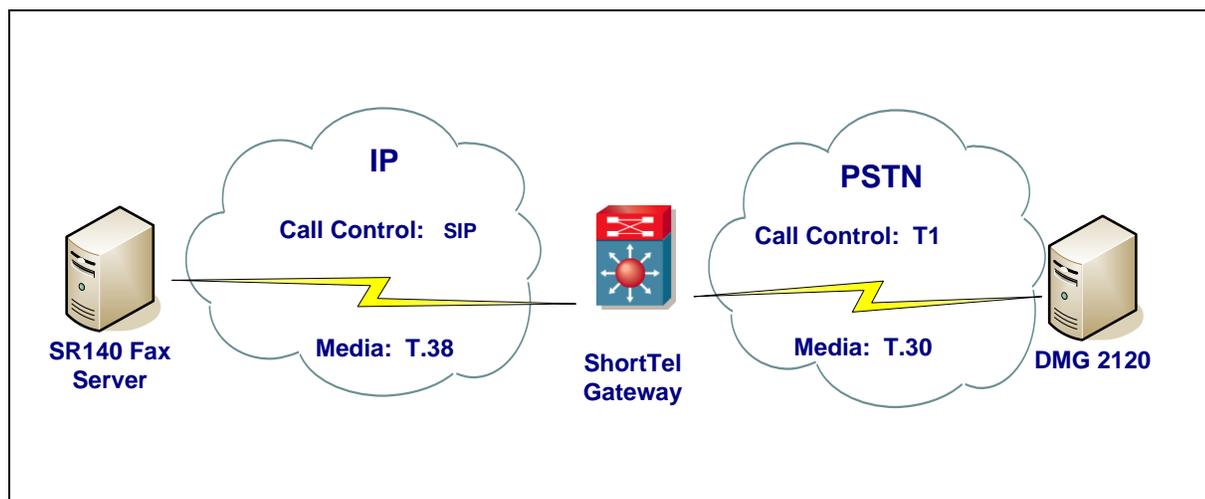


Diagram Notes:

- SR140 Fax Server = Fax Server including Dialogic® Brooktrout® SR140 Fax Software and third party fax application.

3. Prerequisites

- Dialogic Brooktrout SR140 running SDK 6.2.3 or later.
- ShoreTel Controlled Release for Release 10 build 15.6.4207.0 or later.

4. Summary of Limitations

- Enabling redundancy for T.38 causes data errors when using the ShoreTel Gateway. For interoperability, the SR140 requires redundancy for both image and control to be set to 0.
- The ShoreTel Gateway sends a SIP session refresh if it is enabled on the ShoreTel Gateway even if the SR140 is configured not to use SIP session refresh. This will cause the SR140 to drop the call if a SIP refresh is received. For interoperability, the SR140 must enable SIP or the ShoreTel Gateway must disable SIP session refresh by using the ShoreWare Director, select Call Control then Options, there you can completely disable the session timer, set the session interval and the refresher.
- The ShoreTel Gateway does not support V.17 for T.38. The maximum protocol supported is V.29 (9600 bits/sec).

5. Gateway Setup Notes

5.1 Network Addresses

The following table lists the IP addresses and their descriptions used in subsequent sections.

Device #	Device Make, Model, and Description	Device IP Address
1	ShoreTel	10.242.203.99
2	SR140	10.242.202.14

5.2 Router Configuration

Configuring the ShoreTel equipment was performed using the ShoreTel ShoreWare Director. This web interface allows the user to administrate the ShoreTel equipment using a web browser.

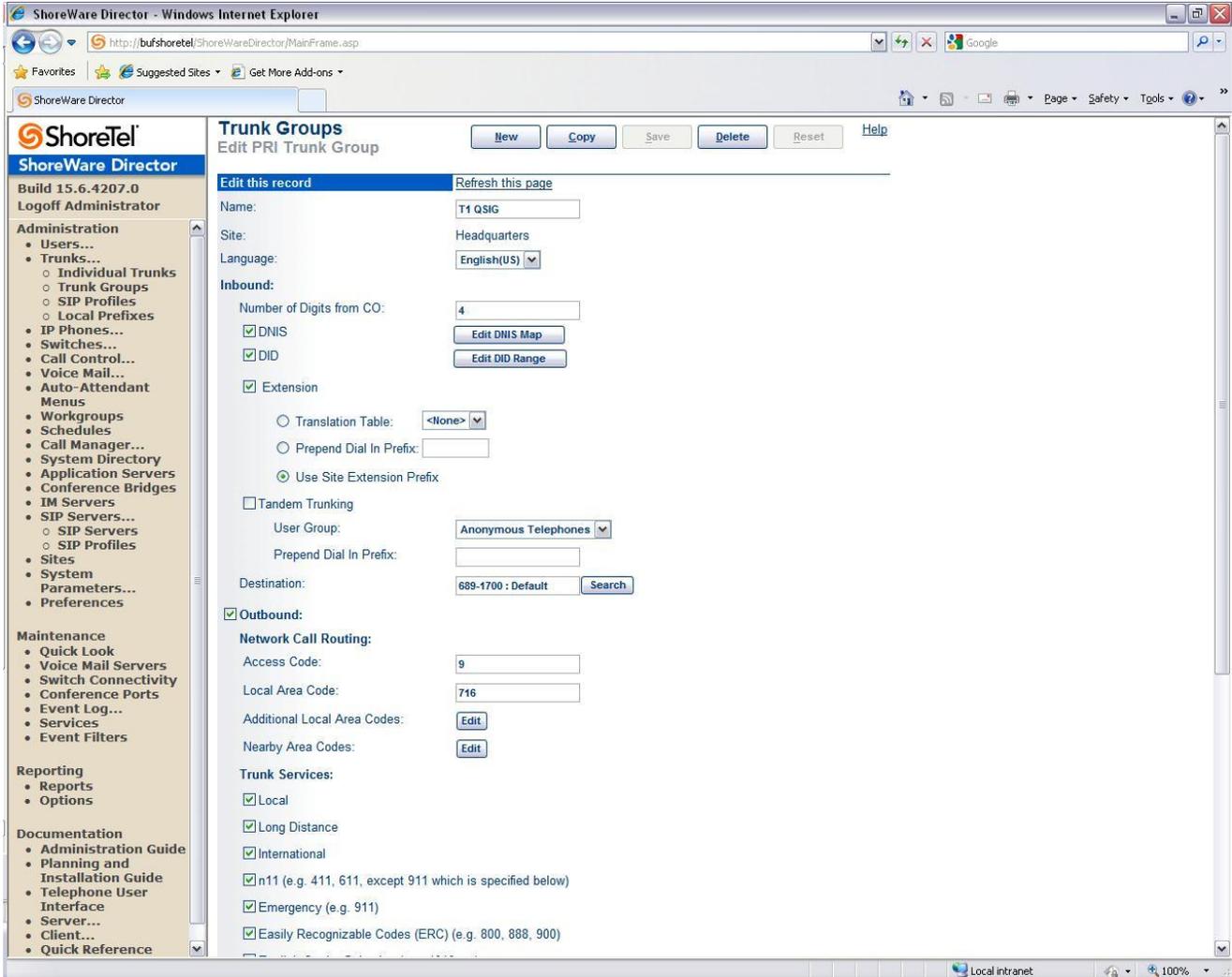
Using the web browser, the host name for the SR140 was added as a SIP Server. The protocol for the SR140 SIP Server must be set to UDP via the Protocol pulldown.

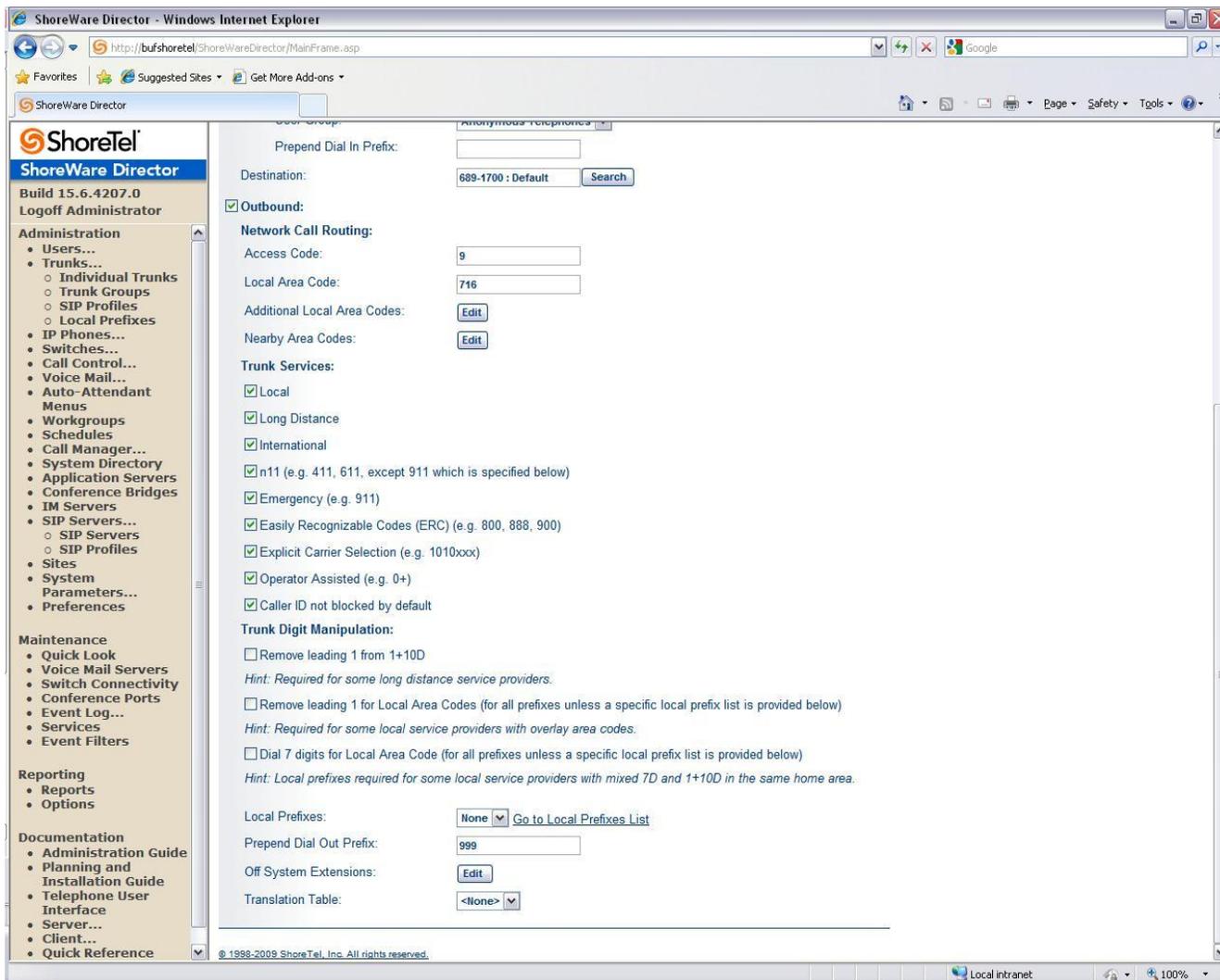
The screenshot displays the ShoreWare Director web interface in Internet Explorer. The main window shows the 'SIP Servers' configuration page with a table of existing servers. A modal dialog box titled 'SIP Server Info' is open, showing the configuration for a new server named 'SR140'.

Name	Extension	Site	Host	Override Default Port	SIP Profile	Protocol	Voice Mail Enabled	Fax Enabled
My Desk	689-1801	Headquarters	10.242.207.164		Microsoft Exchange	TCP	Yes	Yes
PIMG203.76	689-1710	Headquarters	10.242.203.76		Microsoft Exchange	TCP	Yes	Yes
SR140	689-1706	Headquarters	10.242.202.14		Microsoft Exchange	UDP	Yes	Yes
TIMG203.77	689-1709	Headquarters	10.242.203.77		Microsoft Exchange	TCP	Yes	Yes
TIMG203.99	689-1701	Headquarters	10.242.203.99		Microsoft Exchange	UDP	No	Yes

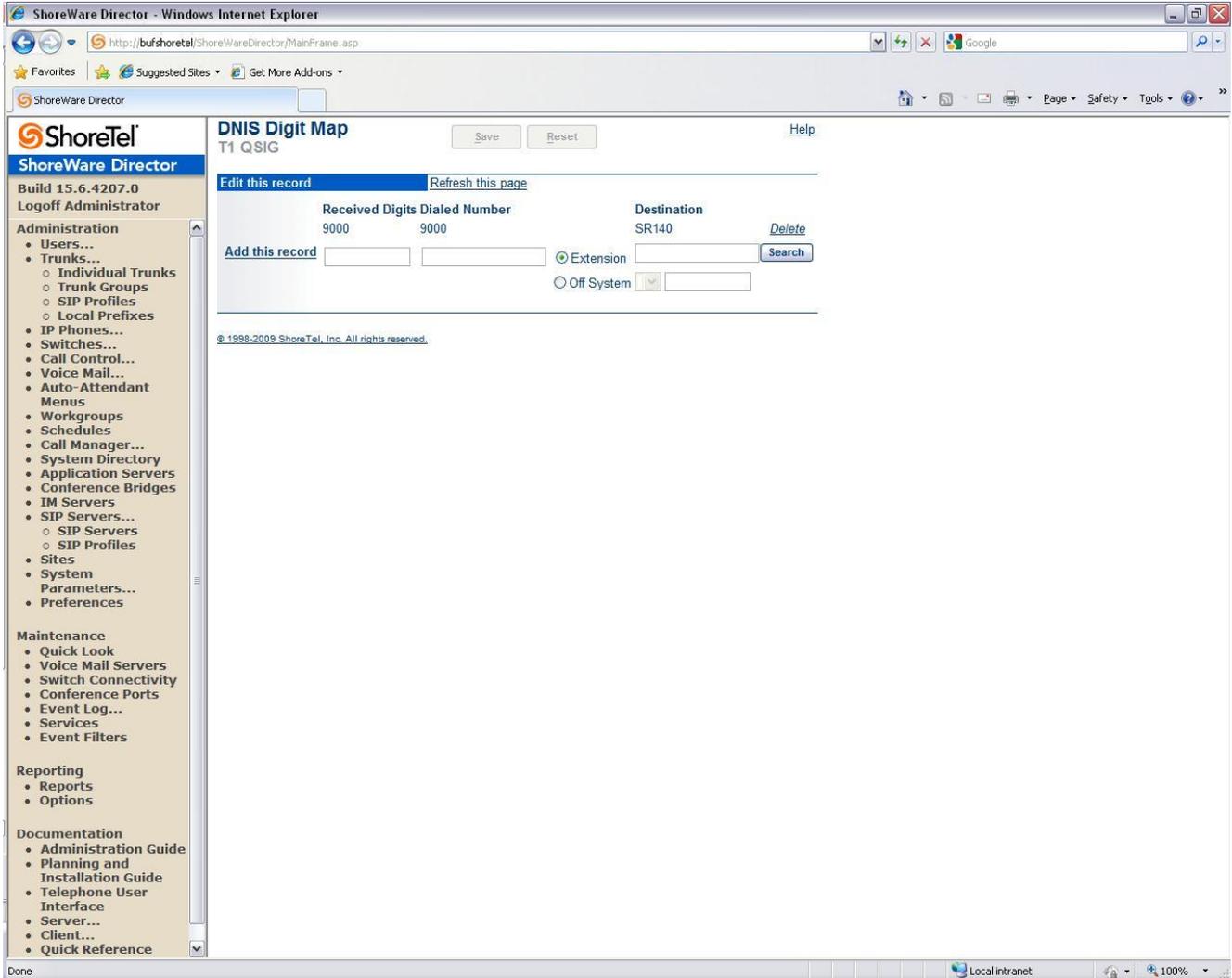
Name:	SR140
Site:	Headquarters
Protocol:	UDP
Host (Name / Address / Domain):	10.242.202.14
Override Default Port:	
<input checked="" type="checkbox"/> Allow External Voice Mail for Extension-Only User	
<input checked="" type="checkbox"/> Allow Fax Redirect to This Server	
Extension:	689-1706
Assigned User Group:	IP Telephones
SIP Profile:	Microsoft Exchange
Digest Authentication:	<None>
User ID:	
Password:	

The next two screenshots were included as a reference and include the settings used in the sample test configuration for the PRI Trunk Group between the ShoreTel Gateway and the DMG 2120.





To the configure ShoreTel to route calls from the SIP Trunk to the SR140, a DNIS Digit Map was created to direct calls from the SIP Trunk that match the 'Dialed Number' to the SR140 location.



6. Dialogic® Brooktrout® SR140 Fax Software Setup Notes

The Installation and Configuration Guides used to setup the SR140 is available from the site:

<http://www.dialogic.com/manuals/brooktrout/default.htm>

The SR140 callctrl.cfg file used in the sample test configuration is shown below for reference. The changes from the default settings are highlighted in yellow.

```
l3l4_trace=verbose
l4l3_trace=verbose
api_trace=verbose
internal_trace=verbose
host_module_trace=verbose
ip_stack_trace=warning
# Most of the time a path should be used for this file name.
trace_file=test_0004_ecc.log
max_trace_files=1
max_trace_file_size=10
[host_module.1]
module_library=brktsip.dll
enabled=true
[host_module.1/t38parameters]
t38_fax_rate_management=transferredTCF
fax_transport_protocol=t38_only
t38_fax_udp_ec=t38UDPRedundancy
rtp_ced_enable=true
t38_max_bit_rate=14400
t38_fax_version=0
media_passthrough_timeout_inbound=1000
media_passthrough_timeout_outbound=4000
media_renegotiate_delay_inbound=1000
media_renegotiate_delay_outbound=-1
t38_fax_fill_bit_removal=false
t38_fax_transcoding_jbig=false
t38_fax_transcoding_mmr=false
t38_t30_fastnotify=false
t38_type_of_service=0
t38_UDPTL_redundancy_depth_control=0
t38_UDPTL_redundancy_depth_image=0
[host_module.1/rtp]
rtp_frame_duration=20
rtp_jitter_buffer_depth=100
rtp_codec=pcmu pcma
rtp_silence_control=inband
rtp_type_of_service=0
rtp_voice_frame_replacement=0
[host_module.1/parameters]
sip_max_sessions=256
sip_default_gateway=0.0.0.0:0
sip_proxy_server1=
sip_proxy_server2=
sip_proxy_server3=
sip_proxy_server4=
sip_registration_server1=
sip_registration_server1_aor=
sip_registration_server1_username=
sip_registration_server1_password=
sip_registration_server1_expires=3600
sip_registration_server2=
sip_registration_server2_aor=
sip_registration_server2_username=
sip_registration_server2_password=
sip_registration_server2_expires=3600
sip_registration_server3=
```

```
sip_registration_server3_aor=  
sip_registration_server3_username=  
sip_registration_server3_password=  
sip_registration_server3_expires=3600  
sip_registration_server4=  
sip_registration_server4_aor=  
sip_registration_server4_username=  
sip_registration_server4_password=  
sip_registration_server4_expires=3600  
sip_registration_interval=60  
sip_Max-Forwards=70  
sip_From=Anonymous <sip:no_from_info@anonymous.invalid>  
sip_Contact=0.0.0.0:0  
sip_username=-  
sip_session_name=no_session_name  
sip_session_description=  
sip_description_URI=  
sip_email=  
sip_phone=  
sip_Route=  
sip_session_timer_session_expires=1800  
sip_session_timer_minse=-1  
sip_session_timer_refresh_method=0  
sip_ip_interface=  
sip_ip_interface_port=5060  
sip_redirect_as_calling_party=0  
sip_redirect_as_called_party=0  
sip_user_agent=Brktsip/6.2.0B5 (Dialogic)  
[module.41]  
model=SR140  
virtual=1  
exists=1  
vb_firm=C:\interop kit SDK620 v1.2\fdtool-6.2.0\bin\bostvb.dll  
channels=120  
[module.41/ethernet.1]  
ip_interface={7D57B541-A7F4-4674-9B2B-29AAE2E3A9A2}:0  
media_port_min=56000  
media_port_max=57000  
[module.41/host_cc.1]  
host_module=1  
number_of_channels=120
```

7. Frequently Asked Questions

- *"I'm configured as near as possible to this the sample configuration described in this document, but calls are still not successful; what is my next step?"*
 - ➔ Provide this document to your gateway support.
 - ➔ Ensure T.38 is enabled on the gateway.
 - ➔ Confirm that basic network access is possible by pinging the gateway.
- *"How do I obtain Wireshark traces?"*
 - ➔ The traces can be viewed using the Wireshark network analyzer program, which can be freely downloaded from <http://www.wireshark.org>.
 - ➔ To view the call flow in Wireshark, open the desired network trace file and select "Statistics->VoIP Calls" from the drop down menu. Then highlight the call and click on the "Graph" button.