

Dialogic[®] Brooktrout[®] SR140 Fax Software with Mitel 3300 MXe Controller

Installation and Configuration Integration Note

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62 **1. Scope** 63

64 This document is intended as a general guide for configuring a basic installation of the Mitel 3300 MXe Gateway 65 for use with Dialogic[®] Brooktrout[®] SR140 Fax over IP (FoIP) software platform. The interoperability includes SIP 66 call control and T.38/T.30 media.

68 This document is not intended to be comprehensive and thus does not replace the manufacturer's detailed 69 configuration documentation. Users of this document should already have a general knowledge of how to install 70 and configure the Mitel 3300 MXe Controller.

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The sample configuration shown and/or referred in the subsequent sections was used for lab validation testing by Dialogic. Therefore, it is possible and even likely that the example configuration will not match the exact configuration and 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. The Mitel 3300 MXe will be denoted herein as Mitel 3300 or 3300 MXe, or some other form thereof.

2. Configuration Details

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

88 89 2.1 Mitel 3300 MXe Controller Gateway 90

Vendor	Mitel
Model	3300 MXe
Software Version	9.0.3.15
PSTN Device	Dialogic [®] Brooktrout [®] TR1034 Fax Board
Protocol from Gateway to PSTN	T1 PRI ISDN
IP Device	Dialogic [®] Brooktrout [®] SR140
Additional Notes	Same firmware is used on 3300 CXi, CX, and MXe devices. Softswitch (call manager) option also available for overall network solution.
	T.38 licenses must be loaded.
\mathbb{Z}	DSP module must be present.

91 Dialogic[®] Brooktrout[®] SR140 Fax Software 92 2.2

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Vendor	Dialogic
Model	Dialogic [®] Brooktrout [®] SR140 Fax Software
Software Version	Dialogic [®] Brooktrout [®] SDK 6.1.1
Protocol to Gateway	SIP
callctrl.cfg file	Default values

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Dialogic[®] Brooktrout[®] TR1034 Fax Board 2.3

Vendor	Dialogic
PSTN Device	Dialogic [®] Brooktrout [®] TR1034 BRI Fax Board
Software Version	Dialogic [®] Brooktrout [®] SDK 6.1.1
Protocol to PSTN Device	BRI ISDN
callctrl.cfg file	Default values with European Community as country code.

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2.4 **Network System Configuration**

99 100 The diagram below details the sample configuration used in connection with this document.





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- SR140 Fax Server = Fax Server including Dialogic[®] Brooktrout[®] SR140 Fax Software and • 3rd party fax application.
- TR1034 Fax Server = Fax Server including Dialogic[®] Brooktrout[®] TR1034 Fax Board and • 3rd party fax application.

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3. Prerequisites 111

T.38 over SIP Trunk is used to interconnect.

For T.38, the DSP II card must be present and the T.38 licenses must be loaded on the Mitel 3300 MXe Gateway.

T.38 licenses are referred to as "FAX over IP (T.38) Licenses". If the number of T.38 licenses programmed exceeds the available DSP resources, a DSP alarm is raised and a maintenance log is generated.

SIP trunking licenses are referred to as "SIP Trunk Licenses" for the 3300 ICP.

Reboot the system to enable the licenses.

The number of T.38 and SIP trunking licenses can be verified in the "License and Option Selection" screenshot.

License and Option Selection		
Online Licensing with the Application M	lanagement Center	
0 11	5	
Application Record ID:	86969904	
		A
Purchased Options		
IP User Licenses:	460	
ACD Agent Licenses:	150	
IP Device Licenses:	250	
Mailbox Licenses:	18	
Digital Link Licenses:	5	
Compression Licenses:	0	
FAX Over IP (T.38) Licenses:	8	
SIP Trunk Licenses:	4	P.
Analog Line Licenses:	32	
SIP User Licenses:	0	
XNET Networking:	No	
IP Networking:	Yes	
Voice Mail Networking:	No	
Advanced Voice Mail:	No	
Voice Mail Hospitality/PMS:	No	
Tenanting:	No	
MLPP:	No	
Remote Management:	No	
Hardware Identifier:	000002BBC38	
Password:	********	

4. Summary of Limitations

The Mitel 3300 MXe DSP module used in testing only supported v.17 14400 bps fax transfer on T.38 and supported up to 16 fax channels.

In Mitel's documentation, Mitel recommends disabling ECM. This is likely the default configuration for a given deployment.

5. Mitel 3300 MXe Gateway Fax Configuration

SIP trunking was used for the interconnection. See IP Endpoint Configuration section for details.

5.1 Fax Configuration

This Fax Configuration form allows you to define the settings for FAX communication over the IP network.



tem Administration								Change
System Options	Inter-Zo	ne Fax Profile						
Class of Service Options A Class of Service Options A Class of Restriction Group Controller Registry Configu independent Account Code Default Account Code Defin	Maximum F High Speed Low Speed Error Correc Override No	ax Rate: Redundancy: Redundancy: ction Mode (ECM): on-Standard Facilities (N	SF)		14400 (V.17 3 8 Disabled Disabled	, 14400bps)		_
Station Service Assignmen	Previous	Page 1 of 7 🌒 Nex	t G	o to:	-	value:		Go 🔺
SMDR Options Assignmer				Change C	Change Page	Change	All	Clear
Online Services Configurati	htra-Zo	ne Fax Profiles						
Traffic Options Assignmen Application Logical Port Ass	Profile	Maximum Fax Rate	High Speed Redundancy	Low Speed Redundancy	Error Correction Mode	NSF Override	NSF Vendor Code Value	NSF Coun Code Value
Fax Configuration	B 1	-		-		-	-	-
Automatic Route Selection (AR Distribution (AC	2	14400 (V.17, 14400bps)	3	8	Enabled	Disabled		
Talende can biotrouider (No	D 3	14400 (V.17, 14400bps)	0	3	Enabled	Disabled		
	1 4							
	5							
	Eba							

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• Inter-zone FAX profile: defines the FAX settings between different zones in the network. There is only one Inter-zone FAX profile; it applies to all inter-zone FAX communication. It defaults to V.29, 7200bps. It defines the settings for FAX Relay (T.38) FAX communication. The SR140 Fax Server uses inter-zone FAX profile.

· Intra-zone FAX profile: define the FAX settings within each zone in the network.

- Profile 1 defines the settings for G.711 pass through communication.
 - Profile 2 to 64 define the settings for FAX Relay (T.38) FAX communication.
 - All zones default to G.711 pass through communication (Profile 1).

Two new Profiles were created for T.38 fax, Inter-Zone and Intra-Zone.

For the Inter-Zone Fax Profile, select 14,400 (v.17, 14400bps) maximum Fax Rate and disable Error Correction
 Mode (ECM). Save profile.

Inter-Zone Fax Profile

Maximum Fax Rate: High Speed Redundancy: Low Speed Redundancy:	14400 (V.17, 14400bps) V 0 V 3 V	
Error Correction Mode (ECM):	 ● Disabled ● Enabled 	
Override Non-Standard Facilities (NSF) Vendor Code Value: Country Code Value:	0 [0 - 65535] 0 [0 - 65535]	
Label:	Inter-zone	



Delnter-Zo	one Fax Profile							
Maximum High Spee Low Spee Error Corr	Fax Rate: ed Redundancy: ed Redundancy: ection Mode (ECM):				14400 (V.17, 1 0 3 Disabled	4400bps)		
Override I	Non-Standard Facilities (NSF)			Disabled			
Label:					Inter-zone			
Previous	Page 1 of 7 🌒 No	ext	Go	o to:		✓ value	:	Go
				Change	Change Page	Ch	ange All	Clear
EIntra-Zo	one Fax Profiles							
Profile	Maximum Fax Rate	High Speed Redundancy	Low Speed Redundancy	Error Correction Mode	NSF Override	NSF Vendor Code Value	NSF Country Code Value	Label
1	-	-	-	-	-	-	-	G.711
2	14400 (V.17, 14400bps)	0	3	Disabled	Disabled			T.38
3								
4								
5								
6								
7								
8								
9								
10								

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For the Intra-Zone Fax Profile, select 14,400 (v.17, 14400bps) maximum Fax Rate and disable Error Correction Mode (ECM). Note: ECM off is Mitel's default. Save profile.

Intra-Zone Fax Profiles		
Profile: Maximum Fax Rate: High Speed Redundancy: Low Speed Redundancy:	2 14400 (V.17, 14400bps) V 0 V 3 V	1
Error Correction Mode:	 ⊙ Disabled ○ Enabled 	ć
Override Non-Standard Facilities (N	SF)	
Vendor Code Value:	0 [0 - 65535]	
Country Code Value:	0 [0 - 65535]	
Label:	T.38	



5.2 Zone Assignment By default, all zones are set to Intra-zone FAX Profile 1. Based on your network diagram, assign the Intra-zone FAX Profiles to the Zone IDs of the zones.

JZone Assignment		
Zone ID: Intra-zone Compression:	1 () No	() Yes
Intra-zone Fax Profile: Label:	1	
		Save Cancel
Dizone Assignment		
Zone ID: Intra-zone Compression: Intra-zone Fax Profile:	2 No	◯ Yes
Laber.		
		Save Cancel

6. Deployment Details

6.1 Network Addresses

Device #	Device Description	Device IP Address
1	Mitel 3300 MXe	10.1.0.2
2	Dialogic [®] Brooktrout [®] SR140 Fax Software	10.1.0.9

7. IP Endpoint Configuration

The screenshot below shows the Network Element Assignment for the Mitel 3300.

Type: FQDN or IP Address: Local: Version: Zone: SIP Peer: SIP Peer Specific SIP Peer Transport: SIP Peer Port: Fxternal SIP Proxy FQDN or IP Address:	Other N 10.2.2.154 False 2 V
FQDN or IP Address: Local: Version: Zone: SIP Peer: SIP Peer Specific SIP Peer Transport: SIP Peer Port: Fxternal SIP Proxy FQDN or IP Address:	10.2.2.154 False 2
Local: Version: Zone: SIP Peer: SIP Peer Specific SIP Peer Transport: SIP Peer Port: Fxternal SIP Proxy EQDN or IP Address:	False 2 V
Zone: SIP Peer: SIP Peer Specific SIP Peer Transport: SIP Peer Port: External SIP Proxy EQDN or IP Address:	2
SIP Peer: SIP Peer Specific SIP Peer Transport: SIP Peer Port: External SIP Proxy EQDN or IP Address:	
SIP Peer Specific SIP Peer Transport: SIP Peer Port: External SIP Proxy FODN or IP Address:	
SIP Peer Transport: SIP Peer Port: External SIP Proxy EQDN or IP Address:	
SIP Peer Port: External SIP Proxy EQDN or IP Address:	×
External SIP Proxy EQDN or IP Address:	0
External on Trong Table of In Madrooot	
External SIP Proxy Transport:	~
External SIP Proxy Port:	0
SIP Registrar FQDN or IP Address:	
SIP Registrar Transport:	~
SIP Registrar Port:	0

For the test configuration with the Dialogic[®] Brooktrout[®] SR140 Fax Server endpoint, the following values were used:

- Element Name: "fax"
- Type: Other
- IP address: 10.1.0.9
- SIP Peer: checked
- SIP Transport: UDP
- SIP Port: 5060

Configuration was saved.

This resulted in the following:



In the screenshot below, the test configuration is shown as Trunk Service Number 25.

Coloction								
All forms (alphabetical)				Change	Change Page	Chang	e All 🛛 🗍 Clear	
Trunk Circuit Descriptor Assign	Previous Page 3	of 15 🌒 Nex	t	Go to:		value:	Go	
Trunk Circuit Descriptor Assign	Trunk Service #	ssignment						
Trunk Circuit Descriptor Assign Trunk Circuit Descriptor Assign Trunk Circuit Descriptor Assign	Trunk Service Number	Release Link Trunk	Class of Service	Class of Restriction	Baud Rate	Intercept Number	Trunk Label	
Trunk Circuit Descriptor Assign	E 21	No	1	1	300	1		
Trunk Group Assignment	E 22	No	1	1	300	1		
Trunk Service Assignment	E 23	No	1	1	300	1		
🖹 Unit Configuration Display	24	No	1	1	300	1		
URI/Number Translation	25	No	25	4	9600	1	lfax	
User Authorization Profiles	26	No	1	1	300	1		
User Configuration	27	No	1	1	300	1	-	
Voice Mail Options Assignmen	28	No	1	1	300	1		
Voice Mail Port Assignment	Ehm	Ma	4	4	200	4		
E Voice Mail Port Capacity E Voice Mail Prompt Language A	Trunk Service Assi	gnment						
Voice Quality Statistics	Trunk Service Number: 25 Release Link Trunk: No							
	Class of Restriction	:				4		
	Baud Rate: 9600 Intercept Number: 1 Non-dial In Trunks Answer Point - Day:							
About System Administration	Non-dial In Trunks	Answer Point - Ni	gnt 1:					
1								

This full SIP Peer Profile is shown below.

Str Deer Profile SP Peer Profile Label: Fax: Hennock Element: Address Type: Calling Label: Address Type: Creat Account Humanian Address Type: Address Type: Creat Account Humanian Address Type: Creat Account Humanian Address Type: Address Type: Creat Account Humanian Control Creation Trank Service: Trank Service	2	🔊 Webpage Dialog			×
SP Per Profile Label: Memori Elamolt: Registration low Name: Address Type: CrgON Politie Bestination Server: Claine D Debate (Collis) Registration Collis: Soution (Three) Registration Collis: Politie Marinem Simulaneous Calls: Soution (Three) Registration Collis: Registration Col	,	SIP Peer Profile			
SP Few Network lemenes: Few Network lemenes: Fey Section floormation Registration there frames: Outcourd Proxy Server: 10.10.2 Policies 10.10.2 Trans Kerstone: 10.10.2 Maximum Standanewound Calls: 10.10.2 Section Time: 10.10.2 Datable Relation Standanewound Calls: 10.10.2 Section Time: 10.10.2 Datable Relation Proximant Responses 10.10.2 Tomb Mail Propriation Response 10.10.2 Tomb Mail Propriation Response 10.10.2 Tomb Response Response 10.10.2 Tomb Response Response 10.10.2 <td< td=""><td>N</td><td></td><td></td><td></td><td>-</td></td<>	N				-
Nework Extended: Loo Address Type: Collocation Devices Address Type: Collocation Proxy Server:		SIP Peer Profile Label:	FAX		
Address Type: Address Type: Address Type: Address Type: Address Type: Address Type: C PODI Paint CPI: Baint CPI: Bain	y	Network Element:	fax1 💌		
Address Address Address Address Address Address Address Calling Line D Balant CPI: Balant CPI:<	Ē	Local Account Information			A
Address type: C PGCN C	L	Registration User Name:			
Outhound Proxy Server: Calling Line 10 Betaint CPU: Restriction: Policia Trans Service: Basic Minimulaneous Callis: Section Timer: Zone: Tamis Section Timer: Diable Reliable Provisional Response: Provent the collision Handling: Pia Conting Locse Routing Indicator: Pia Conting Locse Routing Indicator: Pia Conting Locse Routing Indicator: Provent the collision Handling: Pia Conting SDP in Initial Invien: Provent the collision Handling: Pia Route Calling Ter Heads: Route Calling Ter Keist	-1	Address Type:	O FQDN	(● IP Address:	
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Beskut CPI: Restriction: Policies Trunk Service: Trunk Service: Trunk Service: Taruk Service: <td>6</td> <td>Calling Line ID</td> <td></td> <td></td> <td></td>	6	Calling Line ID			
Restriction: Image: Connect Restriction: Maximum Simultaneous Calls: 4 Session Time: 0 Zone: 1 Maximum Simultaneous Calls: 4 Session Time: 0 Natt Keepalive: 1 Maximum Simultaneous Calls: 4 Datable Mile Proprietary SDP: 0 Max Respansion: No Use Passified Identify Header: No Use Restricted Character Set For Authentication: No Suppress Use of SDP factors No Consent Linking Proving Indicator: No Suppress Use of SDP factors No Suppress Use of SDP factors No Use Passored Identify Header: No Use Passored Header No andgoing Calls: No Use To Address In Fram Header on Outgoing Calls: No Use Pass SDP in Initial Invite: No Provide SDP To Initial Invite: No Restriction: No C Yes Header Condents Incore Header No C Yes Restriction Incore Collision Header: No C Yes No C Yes	L	Default CPN:			
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Pinties Turk Service: 25 Interconnect Restriction: 1 Maximum Simutaneous Calle: 4 Session Time: 0 Some: 1 Maximum Simutaneous Calle: 4 Maximum Simutaneous Calle: 0 SMDR Tag: 0 Maximum Simutaneous Calle: 0 We Restricted Character Sut For Authentication: 0 Diable Mitel Poprietaneous 0 FONN of P dadress: 0 Supprase Use OSD Financito Media Streamane 0 Supprase Use OSD Financiton Buddiation: 0 Supprase Use OSD Financiton Budiation: 0 Force Answer - send SDP in initial Invite 0 Prevent Header Sin Form Header on Outgoing Calle: 0 How P. Prefered Header No turgoing Calle: 0 Use Dadress In Form Header on Outgoing Calle: 0 How Diverting Party Number Pasathough: 0 0 How Divere	е				
Turk Service: 25 Interconnect Restriction: 1 Maximum Simultaneous Calls: 0 Zone: 1 Session Time: 0 Doe: 1 Solid Tag: 0 NAT Keepalive: 0 Enable Mitel Proprietary SDP: 0 Use Pasterial distribution of the session of t	e	Policies			
Interconnect Restriction: I Maximum Simultaneous Calls: I Session Time: I Jone: I SMDR Tag: I NAT Keepalive: I Enable Mitel Proprietary SDP: I Use Pastered Identity Header: IIII Disable Reliable Provisional Response: IIIIIIIIIIIIIIIIIIIIIIIIIIIIIIIIIIII	е	Trunk Service:	25		
Maximum Simultaneous Calls: Image: Construction of the second of the	1	Interconnect Restriction:	1		
Session Timer: 0 Zone: 1 MAT Keepalive: 0 Enable Mile Proprietary SDP: C No C Yes Use PAssend Identity Header: 0 No C Yes Use Assended Identity Header: 0 No C Yes Use Assended Identity Header: 0 No C Yes Use Assended Identity Header: 0 No C Yes Togote Provisional Response: No C Yes Use Asternate Destination Domain: 0 No C Yes FQDN of P Address: FNo C Yes Suppress Use of SDP Inactive Media Streams: 0 No C Yes Enable Special R-invite Collision Inanding: 0 No C Yes Force Anwey - send SDP in initial Invite message: 0 No C Yes Force Anwey - send SDP in initial Invite message: 0 No C Yes Route Call Using To Header: 0 No C Yes Private SIP Trank: 0 No C Yes Build Contact Using Request URI Address: 0 No C Yes Repeat SDP Trank: 0 No C Yes Repeat SDP Trank: 0 No C Yes Repeat SDP T	1	Maximum Simultaneous Calls:	4		
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MAT Kepalve: 0 Image: 0 Wat Kepalve: 0 Use Passeticated Identity Header: 0 Use Restricted Character Set For Authentication: 0 Use Restricted Character Set For Authentication: 0 Use Restricted Character Set For Authentication: 0 Use Alternate Destination Domain: 0 Use Alternate Destination Domain: 0 Ignore Incoming Loss Routing Indicators: 0 Suppress Use of SDP Inactive Media Streams: 0 Enable Sectination Induction: 0 Corce sending S for List Authentication: 0 Force Answer: sending S for List Authentication: 0 Force Answer: sending S for List Authentices 0.0.0.0 in SDP Messages: 0 Privents ID Tarkiter 0 0 Context Calling Party Number Restrough: 0 0 Privents ID Frank: 0 0 Cyes Privents ID Party Number Restrough: 0 0 Cyes Build Contact Using Request URI Address: 0 0 Cyes Build Contact Using Request URI Address: 0 0 Cyes Renegof	n	Zone:	1		
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For the test configuration, the SIP Peer profile was configured with the following options:

- Network Element: the selected SIP Peer Profile was associated with the previously created "fax" Network Element.
- Address Type: used the IP addresses in SIP messages
- Outbound Proxy Server: selected the Network Element previously configured for the Outbound Proxy Server
- Calling Line ID: the default CPN was applied to all calls
- Trunk Service Assignment: entered the trunk service assignment previously configured, #25
- SMDR: If Call Detail Records are required for SIP Trunking, the SMDR Tag should be configured (by default there is no SMDR and this field is left blank)
- The remaining SIP Peer Profile policy options are similar to the screen capture above.

8. Dialing Plan Overview

The following, as seen on the SIP Peer profile, provides an overview of the dialing plan used for this document.

4 last digits place a call on SIP peer (Dialogic[®] Brooktrout[®] SR140 Fax Software) 8 + phonenumber places a call on T1 ISDN to the external PSTN network



9. Call Routing Configuration

All standard PBX configuration steps with nothing specific for SIP/T.38 trunk towards the Dialogic[®] Brooktrout[®] SR140 Fax Software were used.

10. Dialogic[®] Brooktrout[®] SR140 Fax Software Setup Notes

The Installation and Configuration Guides for SDK 5.2.x, SDK 6.0.x and SDK 6.1.x are available from the site:

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

For the sample test configuration, the SR140 was configured using the default values from SDK 6.1.1 and is shown below for reference.

I3I4_trace=none I4I3_trace=none api_trace=none internal_trace=none host_module_trace=none ip_stack_trace=none # Most of the time a path should be used for this file name. trace_file= 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_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=5 t38_UDPTL_redundancy_depth_image=2 [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=0 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 [module.41] model=SR140 virtual=1 exists=1 vb_firm=C:\fdtool-6.1.1\bin\bostvb.dll channels=6 [module.41/ethernet.1] ip_interface={567CDC61-517C-4CD5-8F10-3DF5CB9CCDEC}:0 media_port_min=56000 media_port_max=57000 [module.41/host_cc.1] host module=1 number_of_channels=6

No sip_default_gateway was filled in since the IP address of the gateway was specified in the dial string in the application. The following dial string was used for the outbound calls: 80113225374152@10.1.0.2. However, when the application does not allow specifying the gateway's IP address, make sure to fill in the IP address in the sip_default_gateway field. In our test scenario, this would be: sip_default_gateway=10.1.0.2:5060