



Vcomm Solution & Interoperability Testing

Application Notes for Configuring Ingate SIParator with an Avaya S8300 for SIP Trunking without SES – Issue 1.0

Abstract

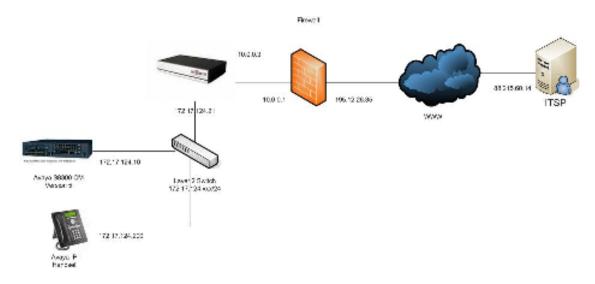
These application notes describe the configuration steps required for the Ingate SIParator version 4.6.2 SBC to successfully enable the Avaya S8300 Call Manager version 5.0 for SIP Trunking without the need for Avaya SIP Enablement Server (SES).

1. Introduction

With the introduction of software release 5.0 for the Avaya S8300 CM, it is now possible to talk directly over a sip trunk without the need for an SES server. This document will walk you through the set up of both Avaya CM and the Ingate SIParator.

The Ingate SIParator is a session border controller that handles the ingress and egress of SIP messaging between two networks (i.e LAN and WAN). It also has basic proxy and far end NAT traversal functionality that could be put to use in certain deployment situations, but are not covered by this document.

The SIParator was connected to the network in a DMZ –LAN scenario:



Please refer to the Ingate SIParator documentation for other deployment scenarios.

The IP addresses used in this test were:

Avaya S8300 172.17.124.10

Ingate SIParator LAN 172.17.124.21 DMZ 10.0.0.3 WAN 195.12.26.35

ITSP 88.215.60.14

2. Equipment and Software Validated

The following equipment and software were used for the testing of this scenario:

Equipment	Software Version				
Avaya S8300 Call Manager	5.0				
Avaya IP Handset					
Ingate SIParator	4.6.2				
Gamma SIP Trunk	IPDC V3				

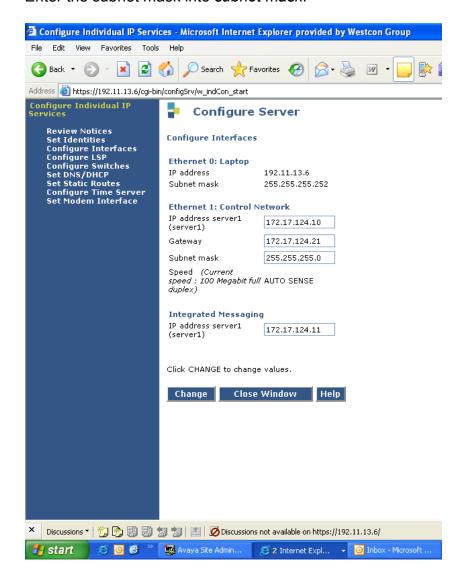
3. Configuring the Avaya S8300

For a detailed description of how to set up the Avaya S8300 please refer to the Avaya's own manual. This guide will show you the necessary changes to this default configuration needed to make this scenario work.

3.1 Avaya Network Settings

Access the Avaya S8300 network settings. Select Configure Server → Configure Interfaces.

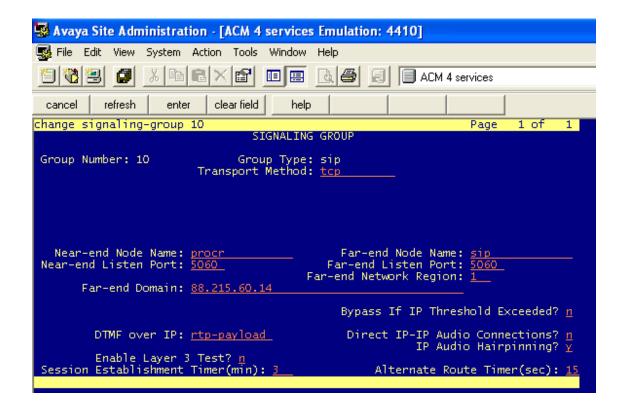
Enter the IP address of the Avaya into *IP address server 1*. Enter the IP address of the Ingate SIParator into *Gateway*. Enter the subnet mask into *subnet mask*.

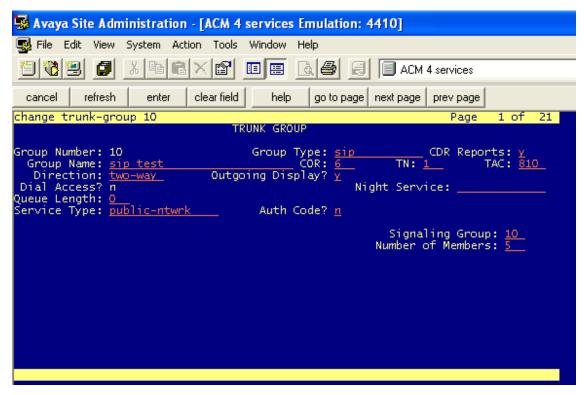


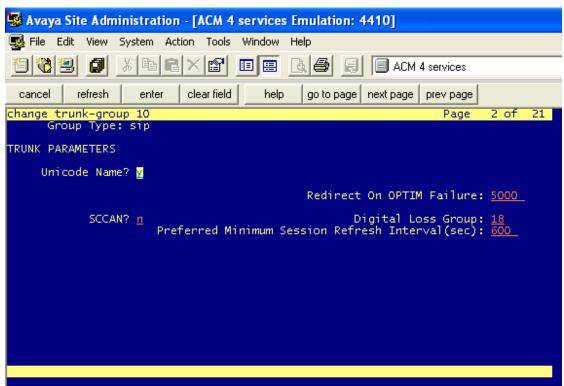
Written by Andy Glover ©2009 Vcomm Uk Ltd
Testing conducted by Andy Glover (Vcomm) and Dean Jones (Westcon)

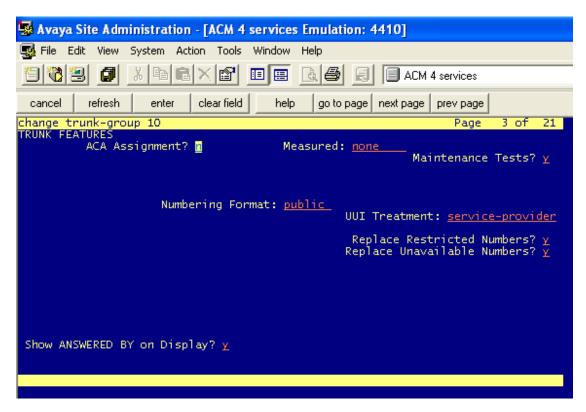
3.2 Avaya Setup Screens

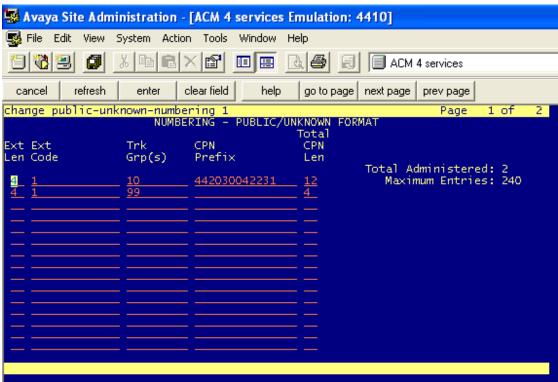
The following screens show the specific changes needed to the basic programming for SIP trunking to work.

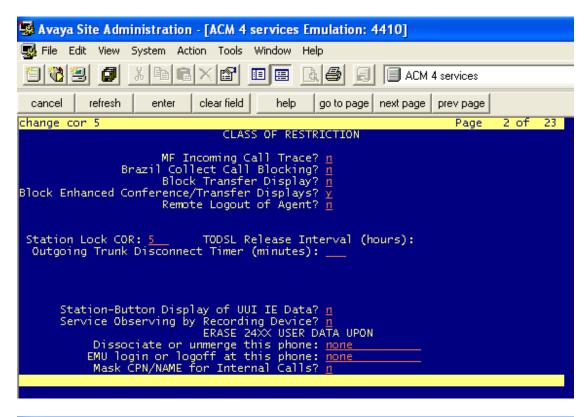


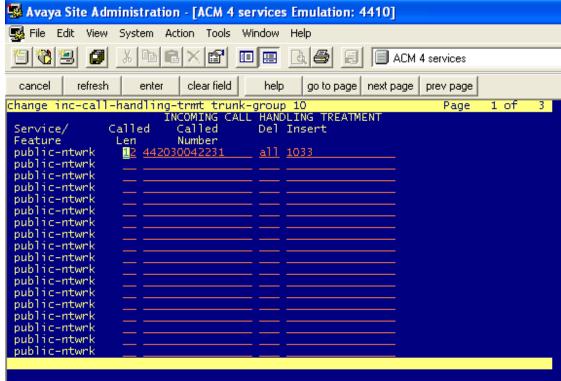












3.3 Points to Note

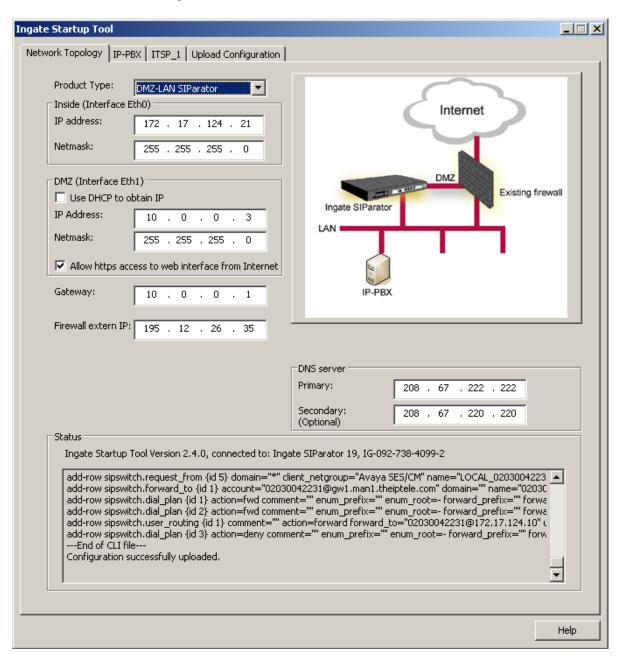
- 1. The Ingate SIParator is used as the Gateway off of the host network
- 2. The Far-End-Domain is that off the ITSP not the Ingate
- 3. Direct IP-to-IP Audio is set to NO
- 4. Hairpinning is set to YES

4. Setting up the Ingate SIParator

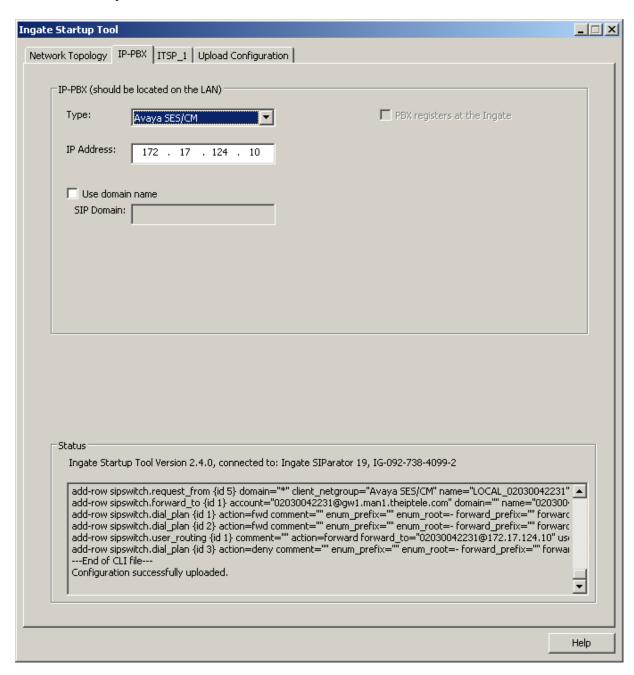
4.1 Basic Configuration

Download and install the Ingate startup tool from www.ingate.com. Once installed run the program and connect to the Ingate.

Enter the network settings.



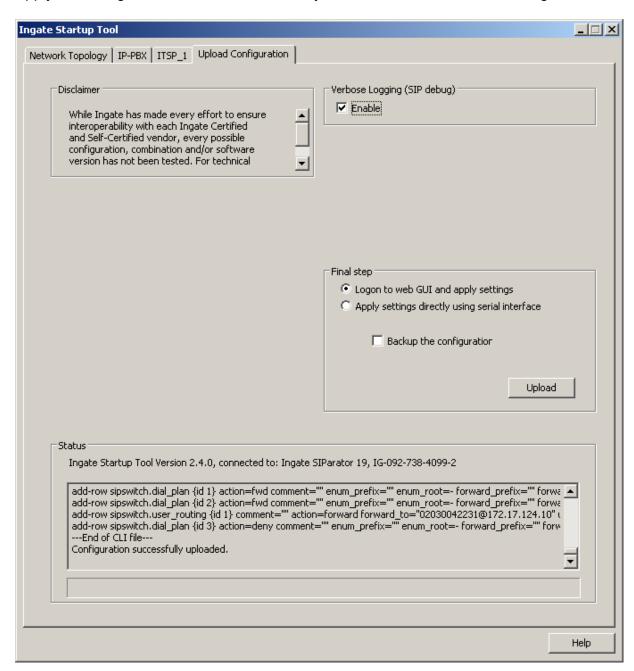
Select the Avaya SES/CM and enter it's IP Address



Select an ITSP, entering the details provided by them for the SIP Trunk.

ingate Startup Tool	
Network Topology IP-PBX ITSP_1 Upload Configuration	
Name: Generic ITSP	DID (start of range) 0001146950084 User name): 1
Provider address IP Address: 0 , 0 , 0 , 0 Use domain name	Account information: Use account Authentication name: (same as DID if blank) Increment authentication name for ranges
Prefix to match and remove from inbound calls Prefix:	Domain: gw1.man1.theiptele.c Password:
Prefix to add to outbound calls Prefix: Forward 3xx messages Family Enable	PBX local numbers (advanced) Local number(start of range, use same as DID if local numbers are not used): Password (only used if PBX registers at the Ingate): PBX registers at the Ingate
Ingate Startup Tool Version 2.4.0, connected to: Ingate SIPara add-row sipswitch.request_from {id 5} domain="*" client_netge add-row sipswitch.forward_to {id 1} account="02030042231@ add-row sipswitch.dial_plan {id 1} action=fwd comment="" enu add-row sipswitch.dial_plan {id 2} action=fwd comment="" enu add-row sipswitch.user_routing {id 1} comment="" action=forw add-row sipswitch.dial_plan {id 3} action=deny comment="" enu add-row sipswitch.dial_plan {id 3} action=deny comment="" enu-reconfiguration successfully uploaded.	roup="Avaya SES/CM" name="LOCAL_02030042231" gw1.man1.theiptele.com" domain="" name="020300- m_prefix="" enum_root=- forward_prefix="" forwarc m_prefix="" enum_root=- forward_prefix="" forwarc vard forward to="02030042231@172.17.124.10" us
	Help

Apply the configuration. This will then take you to the web interface of the Ingate.

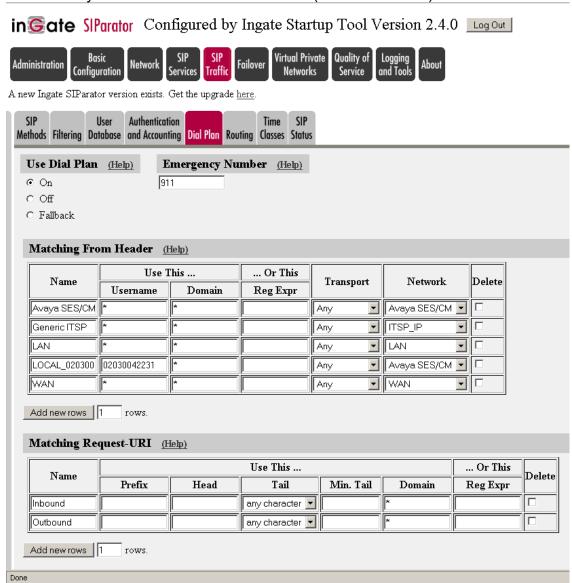


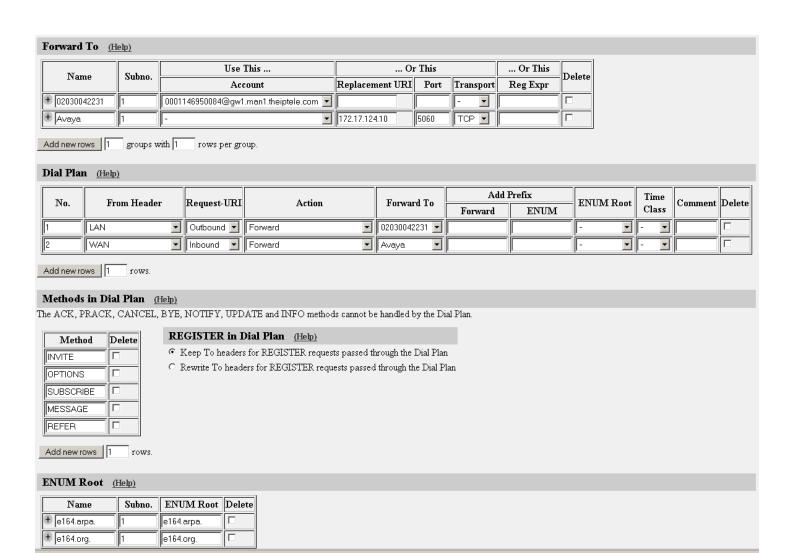
Firstly log in to the web interface, then go to Administration \rightarrow Apply Settings \rightarrow Save Configuration.

Go to Sip Traffic → Dial Plan. Change the transport type under Matching From Header to 'Any'

Add an entry in Matching Request-URI for Inbound calls to suit the inbound number range.

Add an entry into Dial Plan for Inbound calls (see screen shot).





Under Sip Traffic → Routing, change the SIP Routing order as shown.

Administration Retwork SIP SIP SIP Traffic Failover Virtual Private Quality of Services and Tools About

A new Ingate SIParator version exists. Get the upgrade here.

· Changes have been made to the preliminary configuration, but have not been applied.

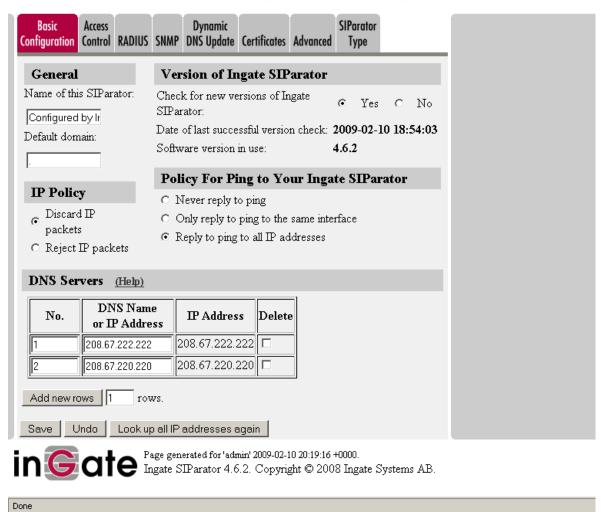
SIP Methods	Filtering	User Database	Authentication and Accounting	Dial Plan	Routing	Time Classes	SIP Status					
DNS	DNS Override For SIP Requests (Help)											
		Relay To										
Dom	- 11	VS Name P Addres	Peerhha HIII	Port	ansport	Priorit	Weigh	Delete				
Addin	Add new rows 1 groups with 1 rows per group.											
em												
	SIP Routing Order (Help) Class 3xx Message Processing (Help) © Forward all											
N	o. R	outing Fu	ınction			t-						
1	Di	Dial Plan DINS Override C Follow redirects										
2	D:											
3		ocal Regist	trar									
<u> </u>												
Stati	c Regis	strations	s (Help)									
Requ	ıests To	User Us	Also Forwar		Delete							
Add n	new rows	1 gr	roups with 1	rows	per group	o.						
Loca	d REF	ER Han	dling (Help)	<u>)</u>								
	Always	handle RI	EFER locally			□ Fo	r clients r	ot support	ting REFE	R		
						□ Fo	r cliente r	int gunnari	tina renlac	ec		
Done												

Next, under Basic Configuration select Reply to ping to all IP addresses.



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Now go to Administration \rightarrow Apply Settings \rightarrow Save Configuration. You should now be able to successfully make inbound and outbound calls.

5. Interoperability Testing

Features tested using this configuration were:

- Inbound calls
- Outbound calls
- Call hold
- Call transfer
- Conference

Further features of the Avaya were not tested but should work. For further assistance in this please contact Vcomm UK Ltd or your Ingate partner.

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