



Application Notes for Configuring NovaLink NovaAlert with Avaya Communication Manager H.323 – Issue 1.0

Abstract

These Application Notes describe the compliance testing of the NovaLink NovaAlert alarm system connected to Avaya Communication Manager via an H.323 link. These Application Notes contain an extensive description of the configurations for both NovaAlert and Avaya Communication Manager.

Information in these Application Notes has been obtained through *DeveloperConnection* compliance testing and additional technical discussions. Testing was conducted via the *DeveloperConnection* Program at the Avaya Solution and Interoperability Test Lab.

1. Introduction

The purpose of this document is to describe the compliance testing done with NovaAlert and Avaya Communication Manager, including a description of the configuration of each, a description of the tests that were performed, and a summary of the results of those tests.

The NovaAlert is a PC-resident application which is used in a health care, hotel or industrial environment for alerting, messaging or information services. NovaAlert can react to external alarm stimuli which indicate the existence of an emergency situation by informing affected persons of the situation.

NovaAlert supports multiple interfaces, including the H.323 trunk described in these Application Notes.

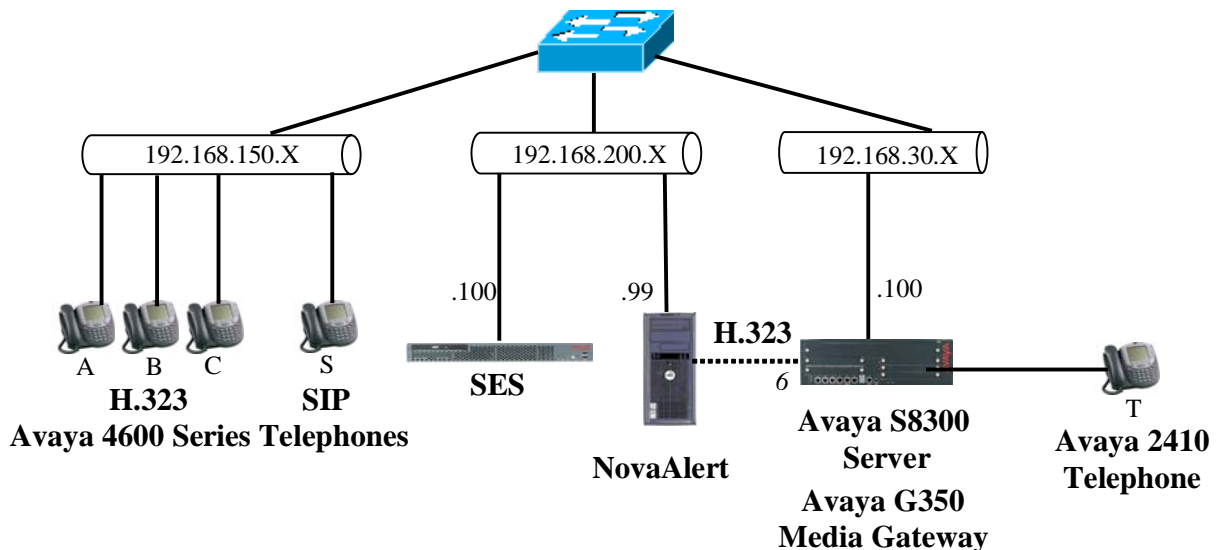


Figure 1: NovaAlert Test Configuration

The H.323 trunk connecting Avaya Communication Manager to the NovaAlert server was configured as trunk group 6, as shown in the diagram. The function of each of the components in **Figure 1** is as follows:

- The NovaAlert server signals alarm events to Avaya IP Telephones attached to Avaya Communication Manager via the H.323 trunk.
- Avaya Communication Manager runs on the Avaya S8300 Server and communicates with the NovaAlert server and Avaya Telephones via the Avaya G350 Media Gateway.
- The Avaya SIP Enablement Services (SES) server is the interface between Avaya Communication Manager and Avaya SIP Telephones.

2. Equipment and Software Validated

The following equipment and software were used for the sample configuration provided:

Equipment	Software Version
Avaya S8300 Server	Avaya Communications Manager 4.0 (R014x.00.0.730.5) Service Pack 00.0.730.5-13566
Avaya SIP Enablement Services Server	SES-3.1.2.0-309.0
Avaya G350 Media Gateway	26.31.0
MM712AP DCP	HW05 FW008
Avaya 4600 series H.323 stations	2.8
Avaya 4600 series SIP stations	2.2.2
NovaLink NovaAlert	7.5 SP 1A
Microsoft Windows Server 2003 SE	SP2

Table 1: Version Numbers of Equipment and Software

3. Configuration

The following table contains the extensions that are used for testing. The capital letter designations correspond to the telephones shown in **Figure 1**.

Extension	Designation
3000136	A
3000134	B
3000133	C
3000115	S
3000001	T
6000000	NovaAlert via H.323
6000001	NovaAlert number for direct alarm initiation

Table 2: Extensions Used for Testing

3.1. Configure Avaya Communication Manager

The configuration and verification operations illustrated in this section were all performed using the Avaya Communication Manager SAT terminal via SSH port 5022.

The information provided in this section describes the configuration of Avaya Communication Manager for this solution. For all other provisioning information such as installation and configuration, please refer to the product documentation in reference [1].

The configuration operations described in this section can be summarized as follows:

- Verify that the licenses allocated to the system are sufficient to support the required configuration.
- Configure the dial plan and call routing required for the NovaAlert configuration.
- Configure the H.323 interface that is used to connect to the NovaAlert server.
- Configure the telephone stations that are to be used for testing.
- Configure Avaya Communication Manager as required to interface to the Avaya SIP Enablement Services server.

3.1.1. Verify system-parameters customer-options

Use the **display system-parameters customer-options** command to verify that Avaya Communication Manager is licensed to meet the minimum requirements to interoperate with the NovaAlert server. Those items shown in bold indicate required values or minimum capacity requirements. If these are not met in the configuration, please contact an Avaya representative for further assistance.

On page 1 of this form, verify that the “Maximum Off-PBX Telephones – OPS” is sufficient for the number of Avaya SIP Telephones to be used.

display system-parameters customer-options		Page 1 of 10
OPTIONAL FEATURES		
G3 Version: V14		
Location: 2	RFA System ID (SID): 1	
Platform: 13	RFA Module ID (MID): 1	
		USED
Platform Maximum Ports:	900	76
Maximum Stations:	450	7
Maximum XMOBILE Stations:	0	0
Maximum Off-PBX Telephones - EC500:	0	0
Maximum Off-PBX Telephones - OPS:	5	2
Maximum Off-PBX Telephones - PBFMC:	0	0
Maximum Off-PBX Telephones - PVFMC:	0	0
Maximum Off-PBX Telephones - SCCAN:	0	0

Figure 2: System-Parameters Customers-Options Form, Page 1

On page 2, the value configured for “Maximum Concurrently Registered IP Stations” must be sufficient to support the total number of IP stations used.

The number “Maximum Administered SIP Trunks” must be sufficient to support the maximum number of members assigned to all SIP trunks.

display system-parameters customer-options		Page 2 of 10
OPTIONAL FEATURES		
IP PORT CAPACITIES		USED
Maximum Administered H.323 Trunks: 30		5
Maximum Concurrently Registered IP Stations: 10		3
Maximum Administered Remote Office Trunks: 0		0
Maximum Concurrently Registered Remote Office Stations: 0		0
Maximum Concurrently Registered IP eCons: 0		0
Max Concur Registered Unauthenticated H.323 Stations: 0		0
Maximum Video Capable H.323 Stations: 0		0
Maximum Video Capable IP Softphones: 0		0
Maximum Administered SIP Trunks: 10		3
Maximum Number of DS1 Boards with Echo Cancellation: 0		0
Maximum TN2501 VAL Boards: 0		0
Maximum Media Gateway VAL Sources: 0		0
Maximum TN2602 Boards with 80 VoIP Channels: 0		0
Maximum TN2602 Boards with 320 VoIP Channels: 0		0
Maximum Number of Expanded Meet-me Conference Ports: 0		0

Figure 3: System-Parameters Customers-Options Form, Page 2

On page 3 of this form, the “Cvg Of Calls Redirected Off-net” parameter must be set to “y” to allow redirection of calls to NovaAlert.

display system-parameters customer-options		Page 3 of 10
OPTIONAL FEATURES		
Abbreviated Dialing Enhanced List? n	Audible Message Waiting? n	
Access Security Gateway (ASG)? n	Authorization Codes? n	
Analog Trunk Incoming Call ID? n	CAS Branch? n	
A/D Grp/Sys List Dialing Start at 01? n	CAS Main? n	
Answer Supervision by Call Classifier? n	Change COR by FAC? n	
ARS? y	Computer Telephony Adjunct Links? n	
ARS/AAR Partitioning? y	Cvg Of Calls Redirected Off-net? y	
ARS/AAR Dialing without FAC? y	DCS (Basic)? n	
ASAI Link Core Capabilities? n	DCS Call Coverage? n	
ASAI Link Plus Capabilities? n	DCS with Rerouting? n	
Async. Transfer Mode (ATM) PNC? n		
Async. Transfer Mode (ATM) Trunking? n	Digital Loss Plan Modification? n	
ATM WAN Spare Processor? n	DS1 MSP? n	
ATMS? n	DS1 Echo Cancellation? n	
Attendant Vectoring? n		

Figure 4: System-Parameters Customers-Options Form, Page 3

On page 4, the parameters must be set as shown in **Table 3**.

Parameter	Required Setting	Comment
IP Stations	y	This is required so that IP stations can be configured.
Enhanced EC500	y	This is required to enable the allocation of off-PBX SIP telephones.
IP Trunks	y	This is required to allow the allocation of the H.323 trunks to be attached to NovaAlert.

Table 3: System-Parameters Customers-Options Form, Page 4

display system-parameters customer-options		Page 4 of 10
OPTIONAL FEATURES		
Emergency Access to Attendant? y	IP Stations? y	
Enable 'dadmin' Login? y		
Enhanced Conferencing? n	ISDN Feature Plus? n	
Enhanced EC500? y	ISDN Network Call Redirection? n	
Enterprise Survivable Server? n	ISDN-BRI Trunks? y	
Enterprise Wide Licensing? n	ISDN-PRI? y	
ESS Administration? n	Local Survivable Processor? n	
Extended Cvg/Fwd Admin? n	Malicious Call Trace? n	
External Device Alarm Admin? n	Media Encryption Over IP? n	
Five Port Networks Max Per MCC? n	Mode Code for Centralized Voice Mail? n	
Flexible Billing? n		
Forced Entry of Account Codes? n	Multifrequency Signaling? y	
Global Call Classification? n	Multimedia Call Handling (Basic)? n	
Hospitality (Basic)? y	Multimedia Call Handling (Enhanced)? n	
Hospitality (G3V3 Enhancements)? n		
IP Trunks? y		
IP Attendant Consoles? n		

Figure 5: System-Parameters Customers-Options Form, Page 4

On page 8, the “Value-Added (VALU)?” parameter must be set to “y” to enable QSIG features required by NovaAlert.

display system-parameters customer-options		Page 8 of 10
QSIG OPTIONAL FEATURES		
Basic Call Setup? y		
Basic Supplementary Services? y		
Centralized Attendant? n		
Interworking with DCS? n		
Supplementary Services with Rerouting? y		
Transfer into QSIG Voice Mail? n		
Value-Added (VALU)? y		

Figure 6: System-Parameters Customers-Options Form, Page 8

3.1.2. Configure Node Names

Use the **change node-names ip** command to configure the IP addresses of the NovaAlert and the Avaya SES servers.

change node-names ip		Page 1 of 2
		IP NODE NAMES
Name	IP Address	
default	0.0.0.0	
NovaAlert	192.168.200.99	
procr	192.168.30.100	
ses	192.168.200.100	

Figure 7: Node-Names IP Form

3.1.3. Configure H.323 Interface to the NovaAlert Server

Use the **add trunk-group** command to configure the Trunk Group to the NovaAlert Server. Assign values for this command as shown in **Table 4**.

Parameter	Usage
Group Type	Specify the Group Type as “isdn”
Group Name	Select an appropriate name to identify the device.
TAC	Specify a trunk access code that can be used to provide dial access to the trunk group.
Carrier Medium	Specify a Carrier Medium of “H.323” will be used for this trunk group.
Dial Access	Allow dial access to the trunk by dialing the trunk access code.
Service Type	Designate the trunk as a “tie” line to a peer system.
Member Assignment Method *	Specify “auto” assignment.
Signaling Group *	Specify signaling group 6.
Number of Members *	Specify a sufficient number of members to accommodate expected traffic. A value of “5” was used for these tests.
Supplementary Service Protocol	Specify a Supplementary Service Protocol of “b” for QSIG.
Format (page 2)	Specify “unk-unk” to use unknown dialing plan for calls in both directions.
Disconnect Supervision Out	Specify “y” to allow trunk-to-trunk transfers.
Send Name	Specify “y” so that the name of the caller is sent for outgoing calls.
Send Calling Number	Specify “y” so that the number of the caller is sent for outgoing calls.
Format (page 3)	Specify “unknown” to use unknown dialing plan for calls in both directions.
Send Called/Busy/Connected Number	Specify “y” so that the number of the connected party is sent to the caller.
QSIG Value-Added	Specify “y” so that NovaAlert can use the QSIG Value-Added extensions.

Table 4: Trunk-Group Parameters for H.323 Interface

*** Perform a “change trunk-group 6” command after adding Signaling group 6 to define Member Assignment Method, Signaling Group, and Number of Members field.**

add trunk-group 6		Page 1 of 21
TRUNK GROUP		
Group Number: 6	Group Type: isdn	CDR Reports: y
Group Name: NovaLink	COR: 1	TN: 1 TAC: *06
Direction: two-way	Outgoing Display? n	Carrier Medium: H.323
Dial Access? y	Busy Threshold: 255	Night Service:
Queue Length: 0	Auth Code? n	
Service Type: tie	Member Assignment Method: manual	
	Signaling Group:	
	Number of Members:	

Figure 8: Trunk-Group Form for H.323 Interface, Page 1

add trunk-group 6		Page 2 of 21
Group Type: isdn		
TRUNK PARAMETERS		
Codeset to Send Display: 6	Codeset to Send National IEs: 6	
	Charge Advice: none	
Supplementary Service Protocol: b	Digit Handling (in/out): overlap/overlap	
Digit Treatment:	Digits:	
	Digital Loss Group: 18	
Incoming Calling Number - Delete:	Insert:	Format: unk-unk
Disconnect Supervision - In? y Out? y		
Answer Supervision Timeout: 0		

Figure 9: Trunk-Group Form for H.323 Interface, Page 2

add trunk-group 6		Page 3 of 21
TRUNK FEATURES		
ACA Assignment? n	Measured: none	
	Internal Alert? n	Maintenance Tests? y
	Data Restriction? n	NCA-TSC Trunk Member:
	Send Name: y	Send Calling Number: y
Used for DCS? n	Hop Dgt? n	Send EMU Visitor CPN? n
Suppress # Outpulsing? n	Format: unknown	
	UUI IE Treatment: service-provider	
	Replace Restricted Numbers? n	
	Replace Unavailable Numbers? n	
	Send Called/Busy/Connected Number: y	
	Hold/Unhold Notifications? y	
Send UUI IE? n	Modify Tandem Calling Number? n	
Send UCID? n		
Send Codeset 6/7 LAI IE? y		

Figure 10: Trunk-Group Form for H.323 Interface, Page 3

<pre> add trunk-group 6 QSIG TRUNK GROUP OPTIONS TSC Method for Auto Callback: drop-if-possible Diversion by Reroute? y Path Replacement? y Path Replacement with Retention? n Path Replacement Method: better-route SBS? n Display Forwarding Party Name? y Character Set for QSIG Name: eurofont QSIG Value-Added? y Encoding Method: proprietary </pre>	Page 4 of 21
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Figure 11: Trunk-Group Form for H.323 Interface, Page 4

Use the **add signaling-group** command to allocate a signaling group for this trunk group.

Parameter	Usage
Group Type	Specify “h.323”.
Max number of NCA TSC	Assign a sufficient number of Non-Call Associated Temporary Signaling Connections to accommodate the anticipated maximum number of concurrent MWI messages which can be sent by NovaAlert to control the message waiting lamps of Avaya Telephones. A value of “4” was used for these tests.
Trunk Group for NCA TSC	Specify “6” as the Trunk Group to be used Non-Call Associated Temporary Signaling Connection.
Trunk Group for Channel Selection	Specify “6” as the Trunk Group to be used for channel selection.
TSC Supplementary Service Protocol	Specify “b” to designate use of the QSIG protocol.
Far-end Node Name	Specify “NovaAlert” as the far end node name.

Table 5: Signaling-Group Parameters for H.323 Interface

```

add signaling-group 6                                     Page 1 of 5
                                SIGNALING GROUP

Group Number: 6      Group Type: h.323
Remote Office? n      Max number of NCA TSC: 4
SBS? n                Max number of CA TSC: 0
IP Video? n           Trunk Group for NCA TSC: 6
Trunk Group for Channel Selection: 6
TSC Supplementary Service Protocol: b
T303 Timer(sec): 10

Near-end Node Name: procr      Far-end Node Name: NovaAlert
Near-end Listen Port: 1720     Far-end Listen Port: 1720
Far-end Network Region: 1
LRQ Required? n              Calls Share IP Signaling Connection? n
RRQ Required? n

Bypass If IP Threshold Exceeded? n
H.235 Annex H Required? n
DTMF over IP: out-of-band     Direct IP-IP Audio Connections? n
Link Loss Delay Timer(sec): 90 IP Audio Hairpinning? n
Enable Layer 3 Test? n        Interworking Message: PROgress
                                DCP/Analog Bearer Capability: 3.1kHz

```

Figure 12: Signaling-Group Form for H.323 Interface

3.1.4. Configure Dial Plan and Call Routing

Use the **change dialplan analysis** command to specify that dialed strings which begin with “3” or “6” are extensions. The extensions local to this PBX are all seven digit numbers which begin with a “3”. The extensions assigned to the NovaAlert are all seven digit numbers which begin with “6”. The dial string “*06” is used as a trunk access code to access the NovaAlert trunk.

```

change dialplan analysis                                     Page 1 of 12
                                DIAL PLAN ANALYSIS TABLE
                                Percent Full: 3

Dialed   Total   Call   Dialed   Total   Call   Dialed   Total   Call
String   Length  Type   String   Length  Type   String   Length  Type
3         7      ext
6         7      ext
*83       3      dac
*06       3      dac

```

Figure 13: Dialplan Analysis Form

Use the **change uniform-dialplan** command to designate extensions which begin with “6” and are seven digits in length to use the Automatic Alternate Routing (AAR) table.

change uniform-dialplan 0							Page 1 of 2		
UNIFORM DIAL PLAN TABLE							Percent Full: 0		
Matching			Insert			Node			
Pattern	Len	Del	Digits	Net	Conv	Num			
6	7	0		aar	n				

Figure 14: Uniform-Dialplan Form

Use the **change aar analysis** command to select routing pattern “6” for numbers which have the leading dialed string “6”, as specified in the uniform dial plan shown in **Figure 14**.

change aar analysis 0							Page	1 of	2	
AAR DIGIT ANALYSIS TABLE							Percent Full: 3			
	Dialed			Route	Call	Node				ANI
	String	Min	Max	Pattern	Type	Num				Reqd
6		7	7	6	aar					n

Figure 15: AAR Analysis Form

Use the **change route-pattern** command to route numbers using Routing Pattern 6 via Trunk Group 6.

change route-pattern 6										Page	1 of	3
Pattern Number: 2										Pattern Name: NovaAlert H.323		
SCCAN? n										Secure SIP? n		
Grp	FRL	NPA	Pfx	Hop	Toll	No.	Inserted		DCS/	IXC		
No			Mrk	Lmt	List	Del	Digits		QSIG			
							Dgts		Intw			
1:	6	0							n	user		
2:									n	user		
3:									n	user		
4:									n	user		
5:									n	user		
6:									n	user		
BCC VALUE										TSC	CA-TSC	
0	1	2	M	4	W			Request	ITC	BCIE		
										Service/Feature		PARM
										No.	Numbering	LAR
										Dgts	Format	
										Subaddress		
1:	y	y	y	y	y	n	n	rest		none		
2:	y	y	y	y	y	n	n	rest		none		
3:	y	y	y	y	y	n	n	rest		none		
4:	y	y	y	y	y	n	n	rest		none		
5:	y	y	y	y	y	n	n	rest		none		
6:	y	y	y	y	y	n	n	rest		none		

3.1.5. Configure Public-Unknown-Numbering Format

Use the **change public-unknown-numbering** command to designate how telephone numbers are to be displayed on stations that have displays. Specify that seven digit numbers starting with “6” from trunk group “6” and “3” from trunk group “83” should not be modified.

change public-unknown-numbering 7				Page 1 of 2
NUMBERING - PUBLIC/UNKNOWN FORMAT				
Ext	Ext	Trk	CPN	Total
Len	Code	Grp(s)	Prefix	CPN
				Len
7	6	6		7
7	3	83		7
				Total Administered: 2
				Maximum Entries: 240

Figure 17: Public-Unknown-Numbering Form

3.1.6. Configure Telephone Stations

Use the **add station** command to configure all of the telephones shown in **Table 2**. The settings for Avaya 2400 Telephones are the same as those required for the Avaya 4621 Telephone, except that the “Type” designation must be set to match the telephone type.

Parameter	Usage
Type	Enter the type of station that is to be configured.
Security Code	Enter a numeric security code.
Name	Enter a descriptive name for the user of the station.
BUTTON ASSIGNMENTS	Assign “send-calls” and “call-fwd” buttons to the stations, as required to test call coverage and call forwarding with NovaAlert. This not required for SIP telephones.

Table 6: Station Parameters

add station 3000136		Page 1 of 5
STATION		
Extension: 300-0136	Lock Messages? n	BCC: 0
Type: 4621	Security Code: 6310003	TN: 1
Port: S00006	Coverage Path 1:	COR: 1
Name: extn 3000136	Coverage Path 2:	COS: 1
	Hunt-to Station:	
STATION OPTIONS		
Loss Group: 19	Time of Day Lock Table:	
	Personalized Ringing Pattern: 1	
Speakerphone: 2-way	Message Lamp Ext: 300-0136	
Display Language: english	Mute Button Enabled? y	
Survivable GK Node Name:	Expansion Module? n	
Survivable COR: internal	Media Complex Ext:	
Survivable Trunk Dest? y	IP SoftPhone? n	
Customizable Labels? y		

Figure 18: Add Station Form, Page 1

add station 3000136		Page 4 of 5
STATION		
SITE DATA		
Room:		Headset? n
Jack:		Speaker? n
Cable:		Mounting: d
Floor:		Cord Length: 0
Building:		Set Color:
ABBREVIATED DIALING		
List1:	List2:	List3:
BUTTON ASSIGNMENTS		
1: call-appr	5: call-fwd	Ext:
2: call-appr	6:	
3: call-appr	7:	
4: send-calls Ext:	8:	

Figure 19: Add Station Form, Page 4

3.1.7. Configure Interface to Avaya SES and Integration for SIP Telephones

Use the **change off-pbx-telephone station-mapping** command to configure SIP telephones. Assign values for this command as shown in the following table.

Parameter	Usage
Station Extension	Enter the extension of the SIP telephone.
Application	Enter “OPS”.
Phone Number	Enter the phone number assigned to the SIP telephone.
Trunk Selection	Enter the number assigned to the SIP trunk group later in this section .
Call Limit	Enter “3” to allow the SIP telephone to do call transfers.

Table 7: Parameters for Off-PBX-Telephone Station-Mapping

change off-pbx-telephone station-mapping 3000115						Page 1 of 2
STATIONS WITH OFF-PBX TELEPHONE INTEGRATION						
Station Extension	Application	Dial Prefix	CC	Phone Number	Trunk Selection	Config Set
300-0115	OPS	-		3000115	83	1

Figure 20: Off-PBX-Telephone Form, Page 1

change off-pbx-telephone station-mapping 3000115						Page 2 of 2
STATIONS WITH OFF-PBX TELEPHONE INTEGRATION						
Station Extension	Call Limit	Mapping Mode	Calls Allowed	Bridged Calls		
300-0115	3	both	all	both		

Figure 21: Off-PBX-Telephone Form, Page 2

Use the **change feature-access-codes** command to assign feature codes required by SIP telephones, as shown in the following table:

Parameter	Usage
Call Forwarding Activation All and Deactivation	Assign unused feature access codes that are within the local dial plan to activate/deactivate call forwarding.
Send All Calls Activation and Deactivation	Assign unused feature access codes that are within the local dial plan to activate/deactivate call sending all calls to coverage.

Table 8: Parameters for the Feature Access Codes

change feature-access-codes		Page 1 of 5
FEATURE ACCESS CODE (FAC)		
Abbreviated Dialing List1 Access Code:		
Abbreviated Dialing List2 Access Code:		
Abbreviated Dialing List3 Access Code:		
Abbreviated Dial - Prgm Group List Access Code:		
Announcement Access Code:		
Answer Back Access Code:		
Attendant Access Code:		
Auto Alternate Routing (AAR) Access Code:		
Auto Route Selection (ARS) - Access Code 1:		Access Code 2:
Automatic Callback Activation:		Deactivation:
Call Forwarding Activation	Busy/DA: *75 All: *73	Deactivation: *74
Call Forwarding Enhanced Status:	Act:	Deactivation:
Call Park Access Code:		
Call Pickup Access Code:		
CAS Remote Hold/Answer Hold-Unhold Access Code:		
CDR Account Code Access Code:		
Change COR Access Code:		
Change Coverage Access Code:		
Contact Closure	Open Code:	Close Code:

Figure 22: Feature Access Codes Form, Page 1

change feature-access-codes		Page 3 of 5
FEATURE ACCESS CODE (FAC)		
Leave Word Calling Send A Message:		
Leave Word Calling Cancel A Message:		
Limit Number of Concurrent Calls Activation:		Deactivation:
Malicious Call Trace Activation:		Deactivation:
Meet-me Conference Access Code Change:		
PASTE (Display PBX data on Phone) Access Code:		
Personal Station Access (PSA) Associate Code:		Dissociate Code:
Per Call CPN Blocking Code Access Code:		
Per Call CPN Unblocking Code Access Code:		
Priority Calling Access Code:		
Program Access Code:		
Refresh Terminal Parameters Access Code:		
Remote Send All Calls Activation:		Deactivation:
Self Station Display Activation:		
Send All Calls Activation: *71		Deactivation: *72
Station Firmware Download Access Code:		

Figure 23: Feature Access Code Form, Page 3

Use the **change off-pbx-telephone feature-name-extension** command to assign extensions to features required by SIP telephones, as shown in the following table below. Note that the extensions used here are assigned to speed dial entries for SIP telephones, as shown in **Table 14**.

Parameter	Usage
Call Forward All	Assign an unused extension within the local dial plan to the “Call Forward All” feature.
Call Forward Cancel	Assign an unused extension within the local dial plan to the “Call Forward Cancel” feature.
Send All Calls	Assign an unused extension within the local dial plan to the “Send All Calls” feature.
Send All Calls Cancel	Assign an unused extension within the local dial plan to the “Send All Calls Cancel” feature.

Table 9: Parameters for Off-PBX-Telephone Feature-Name-Extension

```

change off-pbx-telephone feature-name-extensions                               Page 1 of 2
EXTENSIONS TO CALL WHICH ACTIVATE FEATURES BY NAME

Active Appearance Select:
Automatic Call Back:
Automatic Call-Back Cancel:
    Call Forward All: 300-1804
Call Forward Busy/No Answer:
    Call Forward Cancel: 300-1806
Call Park:
Call Park Answer Back:
Call Pick-Up:
Calling Number Block:
Calling Number Unblock:
Conference on Answer:
Directed Call Pick-Up:
Drop Last Added Party:
Exclusion (Toggle On/Off):
Extended Group Call Pickup:
Held Appearance Select:

```

Figure 24: Off-PBX-Telephone Feature Name Extensions Form, Page 1

```

change off-pbx-telephone feature-name-extensions                               Page 2 of 2
EXTENSIONS TO CALL WHICH ACTIVATE FEATURES BY NAME

Idle Appearance Select:
Last Number Dialed:
Malicious Call Trace:
Malicious Call Trace Cancel:
Off-Pbx Call Enable:
Off-Pbx Call Disable:
Priority Call:
    Send All Calls: 300-1825
    Send All Calls Cancel: 300-1826
Transfer On Hang-Up:
Transfer to Voice Mail:
Whisper Page Activation:

```

Figure 25: Off-PBX-Telephone Feature Name Extensions Form, Page 2

Use the **change ip-network-region** command to configure the network region used by Avaya SES. Assign values for this command as shown in the following table.

Parameter	Usage
Authoritative Domain	Enter the name assigned to Avaya SES in Figure 38 .
Name	Enter a descriptive name.

Table 10: Parameters for IP-Network-Region 1

```

change ip-network-region 1                                     Page 1 of 19
                                     IP NETWORK REGION
  Region: 1
  Location: 1          Authoritative Domain: ffm.com
    Name: FFM
MEDIA PARAMETERS
  Codec Set: 1          Intra-region IP-IP Direct Audio: yes
                        Inter-region IP-IP Direct Audio: yes
  UDP Port Min: 2048    IP Audio Hairpinning? y
  UDP Port Max: 3329
DIFFSERV/TOS PARAMETERS
  Call Control PHB Value: 46    RTCP Reporting Enabled? y
  Audio PHB Value: 46          RTCP MONITOR SERVER PARAMETERS
  Video PHB Value: 26          Use Default Server Parameters? y
802.1P/Q PARAMETERS
  Call Control 802.1p Priority: 6
  Audio 802.1p Priority: 6
  Video 802.1p Priority: 5      AUDIO RESOURCE RESERVATION PARAMETERS
H.323 IP ENDPOINTS
  H.323 Link Bounce Recovery? y    RSVP Enabled? n
  Idle Traffic Interval (sec): 20
  Keep-Alive Interval (sec): 5
  Keep-Alive Count: 5

```

Figure 26: IP-Network-Region Form

Use the **change ip-codec-set** command to specify the codec to be used for the Network Region assigned to Avaya SES. Specify that the G.711A codec is to be used.

```

change ip-codec-set 1                                         Page 1 of 2
                                     IP Codec Set

  Codec Set: 1

  Audio      Silence    Frames    Packet
  Codec      Suppression Per Pkt  Size(ms)
1: G.711A    n          2          20
2:
3:
4:
5:
6:
7:

```

Figure 27: IP-Codec-Set Form

Use the **add signaling-group** command to configure the Signaling Group parameters for the SIP trunk group. Assign values for this command as shown in the following table.

Parameter	Usage
Group Type	Enter the Group Type as “sip”.
Far-end Node Name	Enter node name assigned to the Avaya SES in Figure 7 .
Far-end Domain	Enter the domain name configured for SES in Figure 38 .

Table 11: Signaling-Group Parameters for SIP Interface

```

add signaling-group 83                                     Page 1 of 1
                                SIGNALING GROUP

Group Number: 83          Group Type: sip
                        Transport Method: tls

Near-end Node Name: procr          Far-end Node Name: ses
Near-end Listen Port: 5061        Far-end Listen Port: 5061
                        Far-end Network Region:
Far-end Domain: ffm.com

                                Bypass If IP Threshold Exceeded? n

DTMF over IP: rtp-payload          Direct IP-IP Audio Connections? y
                                IP Audio Hairpinning? y

Enable Layer 3 Test? n
Session Establishment Timer(min): 3

```

Figure 28: Avaya SES Signaling-Group Form

Use the **add trunk-group** command to configure the SIP interface to Avaya SES. Assign values for this command as shown in the following table.

Parameter	Usage
Group Type	Specify the Group Type as “sip”
Group Name	Select an appropriate name to identify the device.
TAC	Specify a trunk access code which can be used to provide dial access to the trunk group.
Service Type	Designate the trunk as a “tie” line to a peer system.
Signaling Group	Enter the number assigned to the SIP signaling group show in Figure 28 .
Number of Members	Specify sufficient number of members to support the maximum simultaneous connections required.

Table 12: Trunk-Group Parameters for the SIP Interface

add trunk-group 83		Page 1 of 21	
TRUNK GROUP			
Group Number: 83	Group Type: sip	CDR Reports: y	
Group Name: SIP	COR: 1	TN: 1	TAC: *83
Direction: two-way	Outgoing Display? n	Night Service:	
Dial Access? n			
Queue Length: 0			
Service Type: tie	Auth Code? n		
		Signaling Group: 83	
		Number of Members: 5	

Figure 29: Trunk-Group Screen Form

3.2. Configure Avaya IP Telephones

Configure the **46xxsettings.txt** text file to be used by Avaya IP Telephones. The parameters that are required to be configured in this file are shown in the following table. This is a “flat” ASCII file that must reside in the directory of the TFTP server accessible by the Avaya IP Telephones. Avaya IP Telephones must be configured so that the “FileSv” parameter is set to the address of the TFTP server that contains this configuration file, which is re-read each time the phone is restarted.

Parameter	Usage
MWISVR	The value “SES_IP_address” indicates that Avaya SIP telephones should register with the Avaya SES server to receive message waiting events.
SIPDOMAIN	Enter the name of the SIP domain.
ENHDIALSTAT	Set this parameter to “0” to indicate that enhanced dialing is not required.

Table 13: Parameters for Telephone Setting File

SET MWISVR	"SES_IP_address"
SET SIPDOMAIN	"ffm.com"
SET ENHDIALSTAT	0

Figure 30: Telephone Settings File Content

In addition to these settings, Avaya SIP Telephones must be configured manually to add speed dial entries to activate/deactivate Call Forwarding and Send All Calls features, by assigning the extensions that were assigned to the features shown in the following table to speed dial entries. These extensions are those that were assigned to using the **Off-Pbx-Telephone Feature-Name-Extensions** command described in **Table 9**.

Parameter	Extension	Usage
CallFwd On	3001804	Activate Call Forwarding
CallFwd Off	3001806	Deactivate Call Forwarding
SendAll On	3001825	Activate Send All Calls
SendAll Off	3001826	Deactivate Send All Calls

Table 14: Speed Dial Entry Assignments for Avaya SIP Telephones

3.3. Configure Avaya SIP Enablement Services

Avaya SES is needed in this configuration only if Avaya SIP IP Telephones are used. Log in to the Avaya SES Web-based Integrated Management tool by selecting the IP address of the Avaya SES server followed by “/admin” from the Web browser. After entering the login ID and password, select “Launch Administration Web Interface”.

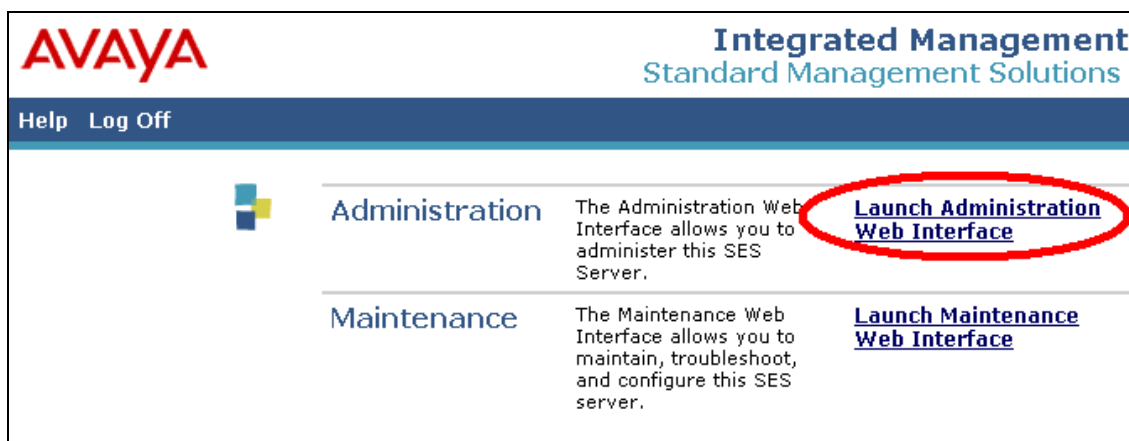


Figure 31: SES Initial Greeting Screen

From the top-level management screen, click “Manage Hosts” followed by “Add Host”.

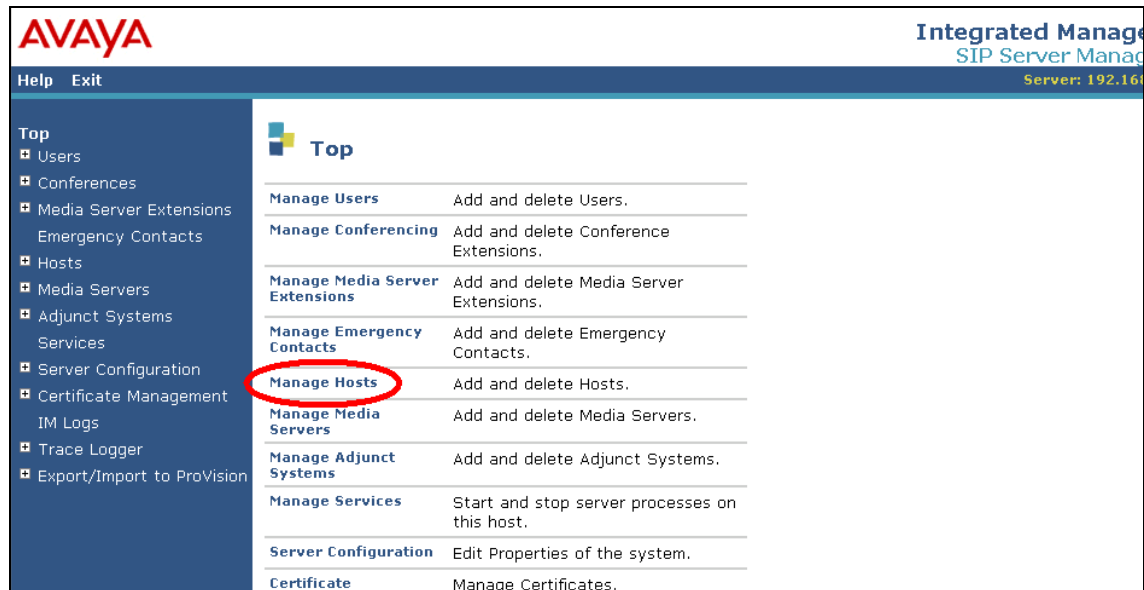
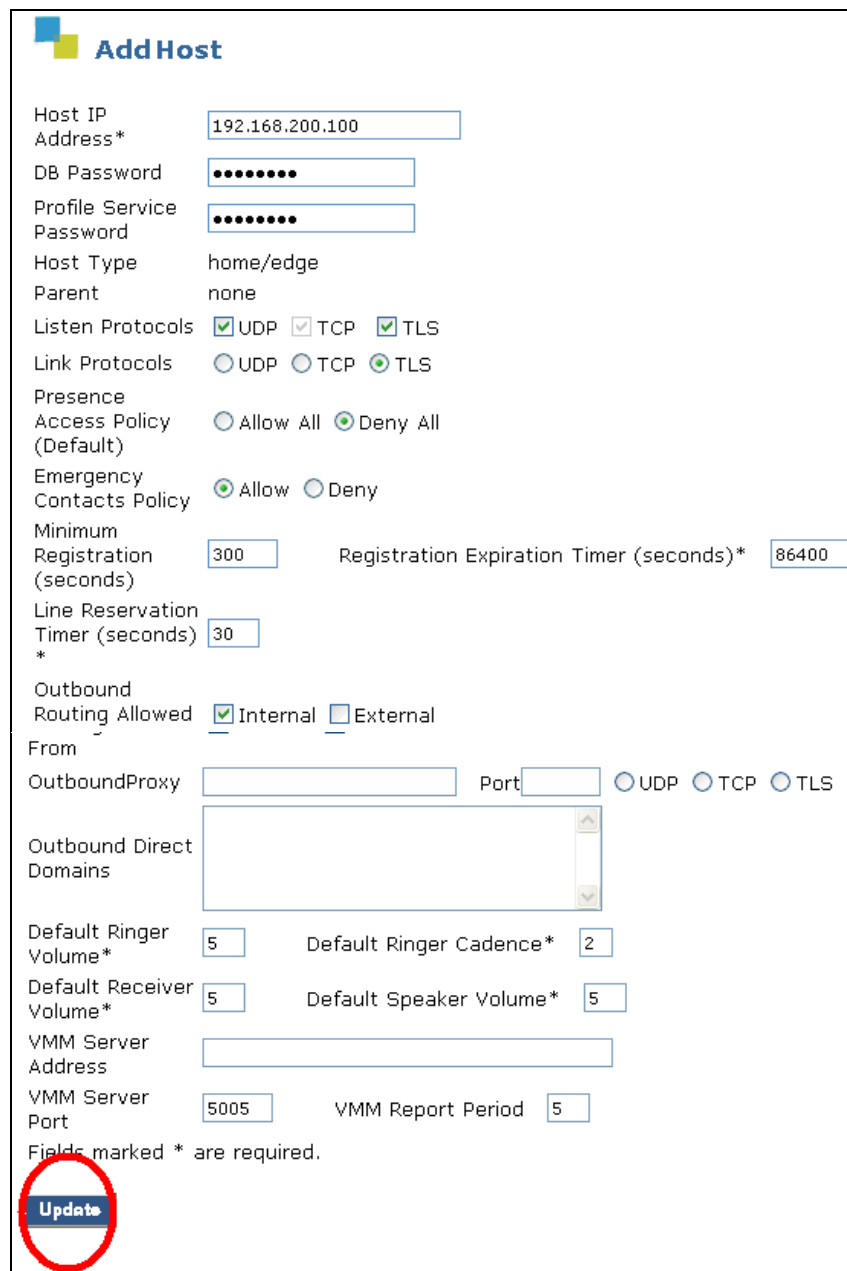


Figure 32: Host Management Selection from Top-Level Administration Screen

Enter the IP address of the Avaya SES Server, a database password, and a Profile Service Password that were allocated to the Avaya SES server when it was installed. Leave the other field assigned to their respective default values. Select the “Update” button.



Add Host

Host IP Address*

DB Password

Profile Service Password

Host Type

Parent

Listen Protocols ☒ UDP ☒ TCP ☒ TLS

Link Protocols ☐ UDP ☐ TCP ☒ TLS

Presence Access Policy (Default) ☐ Allow All ☒ Deny All

Emergency Contacts Policy ☒ Allow ☐ Deny

Minimum Registration (seconds) Registration Expiration Timer (seconds)*

Line Reservation Timer (seconds)*

Outbound Routing Allowed ☒ Internal ☐ External

From OutboundProxy Port ☐ UDP ☐ TCP ☐ TLS

Outbound Direct Domains

Default Ringer Volume* Default Ringer Cadence*

Default Receiver Volume* Default Speaker Volume*

VMM Server Address

VMM Server Port VMM Report Period

Fields marked * are required.

Update

Figure 33: Avaya SES “Add Host” Screen

From the top-level management screen, select “Manage Media Servers”.

The screenshot displays the Avaya management interface. At the top is the Avaya logo. Below it is a navigation bar with 'Help' and 'Exit' links. The main area is divided into two columns. The left column is a dark blue sidebar with a 'Top' header and a list of menu items, each preceded by a plus icon. The right column is white and contains a 'Top' header with a small icon, followed by a table of management options. The 'Manage Media Servers' option in the table is circled in red.

Top	
Manage Users	Add and delete Users.
Manage Conferencing	Add and delete Conference Extensions.
Manage Media Server Extensions	Add and delete Media Server Extensions.
Manage Emergency Contacts	Add and delete Emergency Contacts.
Manage Hosts	Add and delete Hosts.
Manage Media Servers	Add and delete Media Servers.
Manage Adjunct Systems	Add and delete Adjunct Systems.
Manage Services	Start and stop server processes on this host.
Server Configuration	Edit Properties of the system.
Certificate Management	Manage Certificates.
IM Logs	Download IM Logs.
Trace Logger	Manage SIP Trace Logs.
Export Import to ProVision	Export and import data using ProVision on this host.

Figure 34: Media Server Management Selection from Top-Level Administration Screen

Assign a meaningful name to the “Media Server Interface Name”. Select the IP address of the Avaya SES server as the “Host”. Enter the address of the Avaya S8300 Server as the SIP Trunk IP Address. Select the “Add” button when these parameters have been entered.

Add Media Server Interface

Media Server Interface Name*

Host

SIP Trunk

SIP Trunk Link Type ☐ TCP ☒ TLS

SIP Trunk IP Address*

Media Server

Media Server Admin Address (see Help)

Media Server Admin Login

Media Server Admin Password

Media Server Admin Password Confirm

Fields marked * are required.

Add

Figure 35: Avaya SES Add Media Server Interface Screen

From the top-level management screen, select “Server Configuration”.

The screenshot displays the Avaya Top-Level Administration interface. On the left is a dark blue sidebar with a 'Top' section containing a list of management options: Users, Conferences, Media Server Extensions (with sub-items Emergency Contacts and Hosts), Media Servers (with sub-items List and Add), Adjunct Systems (with sub-item Services), Server Configuration, Certificate Management (with sub-item IM Logs), Trace Logger, and Export/Import to ProVision. The 'Server Configuration' option is highlighted with a red oval. The main content area on the right, titled 'Top', contains a table of management functions.

Top	
Manage Users	Add and delete Users.
Manage Conferencing	Add and delete Conference Extensions.
Manage Media Server Extensions	Add and delete Media Server Extensions.
Manage Emergency Contacts	Add and delete Emergency Contacts.
Manage Hosts	Add and delete Hosts.
Manage Media Servers	Add and delete Media Servers.
Manage Adjunct Systems	Add and delete Adjunct Systems.
Manage Services	Start and stop server processes on this host.
Server Configuration	Edit Properties of the system.
Certificate Management	Manage Certificates.
IM Logs	Download IM Logs.
Trace Logger	Manage SIP Trace Logs.
Export Import to ProVision	Export and import data using ProVision on this host.

Figure 36: Server Configuration Selection from Top-Level Administration Screen

From the Server Configuration screen, select “System Properties”.

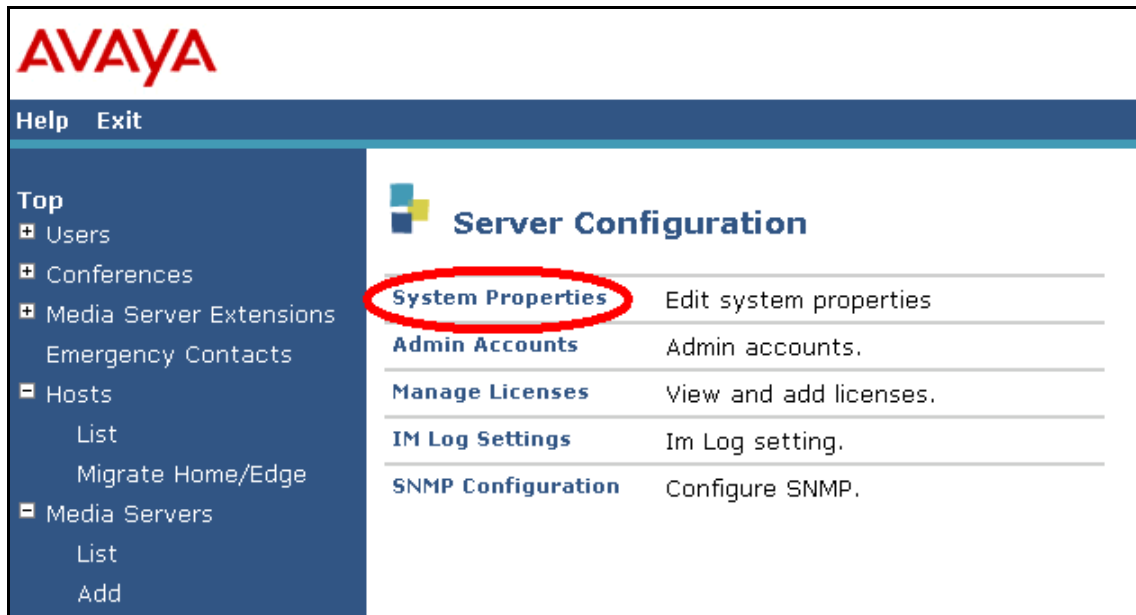
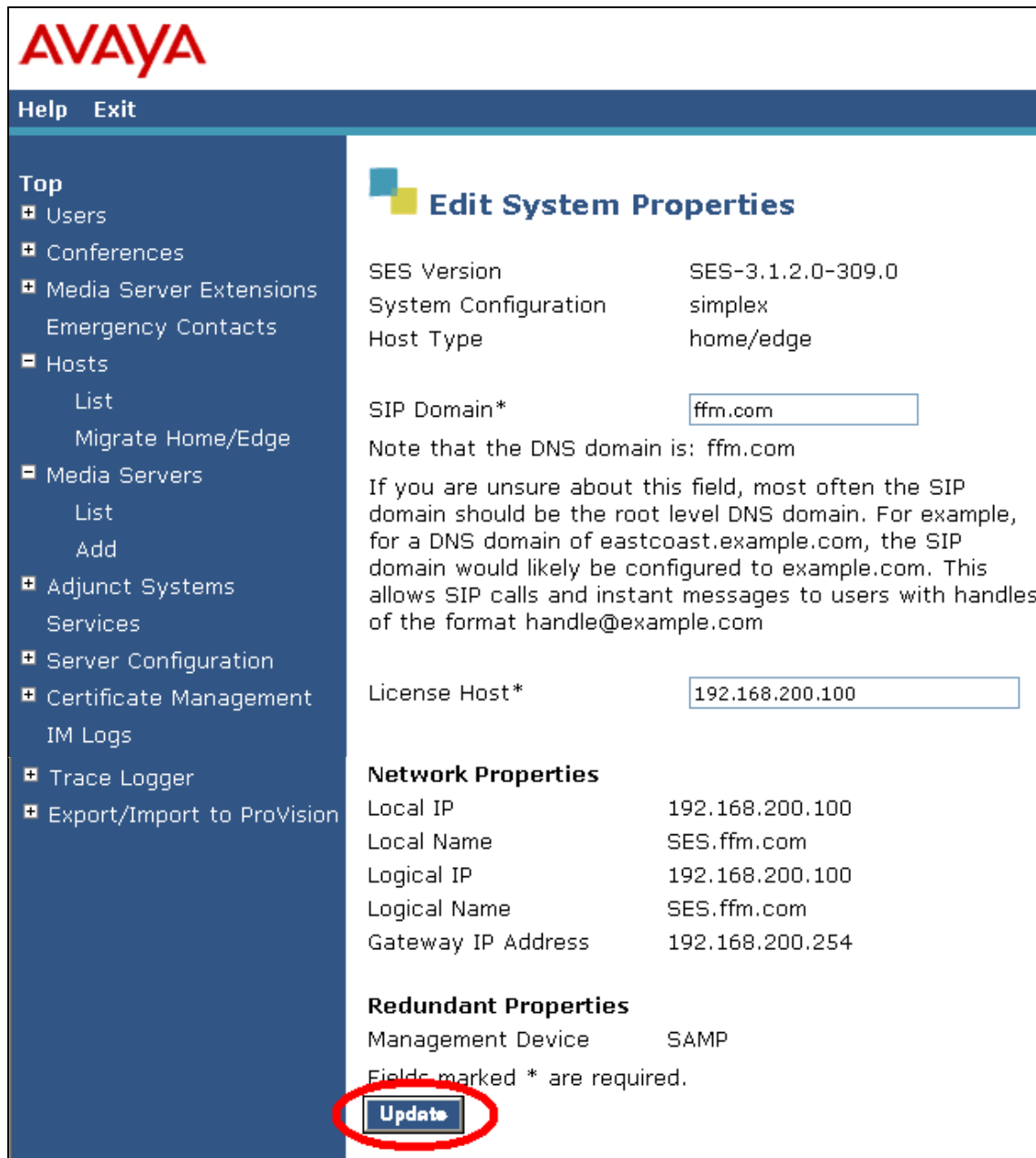


Figure 37: System Properties Selection from Server Configuration Screen

Enter the name to be assigned to the “SIP Domain”. This must be the same name as is assigned in **Figure 26** and **Figure 28**. Select the “Update” button.



The screenshot shows the Avaya SES Edit System Properties screen. On the left is a dark blue sidebar with a navigation menu. The main content area has a white background with a title bar and several sections for system configuration. The 'SIP Domain*' field is highlighted with a red circle, and the 'Update' button at the bottom is also circled in red.

AVAYA

Help Exit

Top

- Users
- Conferences
- Media Server Extensions
 - Emergency Contacts
- Hosts
 - List
 - Migrate Home/Edge
- Media Servers
 - List
 - Add
- Adjunct Systems
 - Services
- Server Configuration
- Certificate Management
 - IM Logs
- Trace Logger
- Export/Import to ProVision

Edit System Properties

SES Version SES-3.1.2.0-309.0

System Configuration simplex

Host Type home/edge

SIP Domain*

Note that the DNS domain is: ffm.com

If you are unsure about this field, most often the SIP domain should be the root level DNS domain. For example, for a DNS domain of eastcoast.example.com, the SIP domain would likely be configured to example.com. This allows SIP calls and instant messages to users with handles of the format handle@example.com

License Host*

Network Properties

Local IP 192.168.200.100

Local Name SES.ffm.com

Logical IP 192.168.200.100

Logical Name SES.ffm.com

Gateway IP Address 192.168.200.254

Redundant Properties

Management Device SAMP

Fields marked * are required.

Update

Figure 38: Avaya SES Edit System Properties Screen

From the top-level management screen, select “Manage Users”.

The screenshot shows the Avaya Integrated Management SIP Server Management interface. The top header includes the Avaya logo, the title 'Integrated Management SIP Server Management', and the server IP '192.168.200.100'. A left sidebar contains a 'Top' menu with expandable items: Users, Conferences, Media Server Extensions, Emergency Contacts, Hosts, Media Servers, Adjunct Systems, Services, Server Configuration, Certificate Management, IM Logs, Trace Logger, and Export/Import to ProVision. The main content area, titled 'Top', lists management functions in a table. The 'Manage Users' entry is circled in red.

Top	
Manage Users	Add and delete Users.
Manage Conferencing	Add and delete Conference Extensions.
Manage Media Server Extensions	Add and delete Media Server Extensions.
Manage Emergency Contacts	Add and delete Emergency Contacts.
Manage Hosts	Add and delete Hosts.
Manage Media Servers	Add and delete Media Servers.
Manage Adjunct Systems	Add and delete Adjunct Systems.
Manage Services	Start and stop server processes on this host.
Server Configuration	Edit Properties of the system.
Certificate Management	Manage Certificates.
IM Logs	Download IM Logs.
Trace Logger	Manage SIP Trace Logs.
Export Import to ProVision	Export and import data using ProVision on this host.

Figure 39: User Management Selection from Top-Level Administration Screen

Select “Add User”.

The screenshot shows the Avaya Integrated Management SIP Server Management interface. The top header includes the Avaya logo, the title 'Integrated Management SIP Server Management', and the server IP '192.168.200.100'. A left-hand navigation menu lists various system components, with 'Users' being the active section. The main content area, titled 'User Administration', contains a table of administrative actions. The 'Add User' link in this table is highlighted with a red circle.

User Administration	
List Users	List all users.
Add User	Add a new user.
Search Users	Search for users.
Edit User Profile	Edit a user by user id.
Delete User	Delete a user by user id.
Update Password	Change a password by user id.
Edit Default User Profile	Edit the default user profile.
Registered Users	Search for registered and provisioned users.

Figure 40: Avaya SES User Administration Screen

Enter the extension of the user to be added as the “Primary Handle”. This is the same extension that was configured in **Section 3.1.7**. Enter a password and first/last name of the user, check the “Add Media Server Extension” box, and click “Add”.

AVAYA Integrated Management
SIP Server Management
Server: 192.168.200.100

Help Exit

Top

- Users
- Conferences
- Media Server Extensions
- Emergency Contacts
- Hosts
- Media Servers
- Adjunct Systems
- Services
- Server Configuration
- Certificate Management
- IM Logs
- Trace Logger
- Export/Import to ProVision

Add User

Primary Handle*

User ID

Password*

Confirm Password*

Host*

First Name*

Last Name*

Address 1

Address 2

Office

City

State

Country

Zip

Add Media Server Extension ☒

Fields marked * are required.

Add

Figure 41: Avaya SES “Add User” Screen

Enter the Media Server Extension for the User ID 3000115 (the extension of the Avaya SIP telephone). Select the Media Server (refer to **Figure 42**) and drop down box and click “Add” to continue.

The screenshot shows the Avaya Integrated Management SIP Server Management interface. The top header includes the Avaya logo, the title 'Integrated Management SIP Server Management', and the server address 'Server: 192.168.200.100'. A navigation menu on the left lists various options under 'Top', including 'Users', 'Conferences', 'Media Server Extensions', 'Emergency Contacts', 'Hosts', 'Media Servers', 'Adjunct Systems', 'Services', 'Server Configuration', 'Certificate Management', 'IM Logs', 'Trace Logger', and 'Export/Import to ProVision'. The main content area is titled 'Add Media Server Extension'. It contains two input fields: 'Extension*' with the value '3000115' and 'Media Server' with a dropdown menu showing 'G350'. Below these fields is a note: 'Fields marked * are required.' At the bottom of the form is a button labeled 'Add', which is circled in red.

Figure 42: Avaya SES Add Media Server Extension Screen

3.4. Configure NovaAlert

3.4.1. Configuration file NovaAlert.ini

The NovaAlert.ini configuration file is a “flat” ASCII file that can be edited with a text editor. This file is contained in the main installation directory on the NovaAlert server (e.g. C:\Program Files\NovaAlert). The values within this file must be set as shown in **Figure 43**. The values for those items shown in bold may vary, depending on the configuration of external components. The values to be used for these entries are described in the following table.

Parameter	Usage
DefaultCallingParty	This is the default extension to be used for calls made by NovaAlert. A value of “6000000” was used for these tests.
LocalUserName	This is the user name to be used for calls made by NovaAlert. A value of “6000000” was used for these tests.
H323_Gateway	This is the IP address that was the Avaya Communication Manager “procr” entry in the Node-Names form shown in Figure 7 .
H323_ListenerConfig	This is the IP port at which NovaAlert listens for connections. This must be the same value that is used as the “Far-end Listen Port” in Figure 12 . The default H.323 signaling port “1720” is used for these tests.

Table 15: Parameters for Telephone Setting File

```
[CallInfo]
CardDriver=3
Interface=9
GewählteNummer=1
AufschaltenAktiv=0
CallingPartyAktiv=1
DefaultCallingParty=6000000
CNIPAktiv=1
DefaultLocalName=NovaAlert
WatchdogDTMF=

[VoIP]
DriverPref=2
LocalUserName=6000000
SilenceDetection=0
MinJitterBuffer=50
MaxJitterBuffer=250
H323_Gateway=192.168.30.100
H323_UseGateKeeper=0
H323_GateKeeperAddress=
H323_GateKeeperZone=
H323_GateKeeperPwd=
H323_UseFastStart=0
H323_UseH245Tunneling=0
H323_ListenerConfig=*:1720
H323_Alias=
```

Figure 43: NovaAlert.ini Configuration File Content

3.4.2. Configure Users

Use the Windows “Start” button to select the program NovaAlert, and select the “User master data” icon from the left frame.

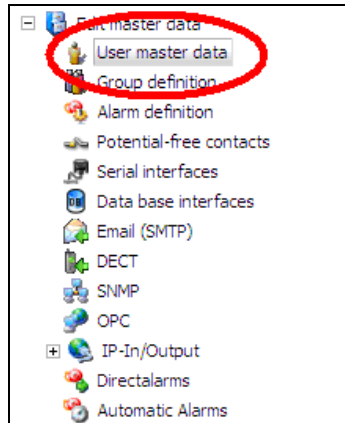


Figure 44: User Master Data Configuration Selection from NovaAlert Top-Level Screen

When the “Person definitions” screen appears, click the “New person” button.

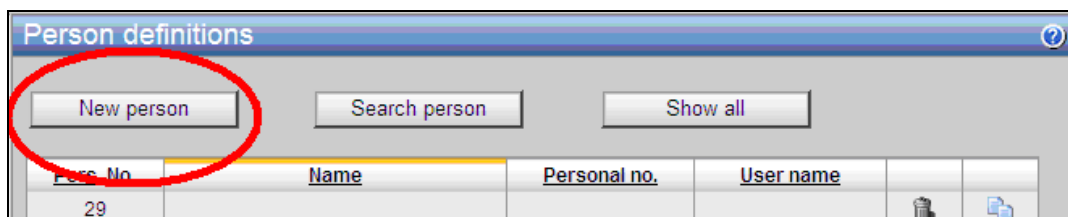


Figure 45: New Person Selection from NovaAlert User Screen

In the “Personal details” tab, enter the name and a PIN code to be assigned to the user. This PIN code will be used by the user when an authorization sequence is required.

Edit person Back ?

No.: 37 Name: Apparat 3000001

Personal details Telephone numbers Authorization Notes

Name: Apparat 3000001 ☐ Deactivated

Add. information: PIN code: 1234

Name of street: Personal ID:

ZIP/Town/City:

Lingua: German

Figure 46: NovaAlert Edit Personal Details Screen

Select the “Telephone numbers” tab, enter the telephone number to be assigned to the user, and click the “Save changes” button.

Edit person Back ?

No.: Name:

Personal details **Telephone numbers** Authorization Notes

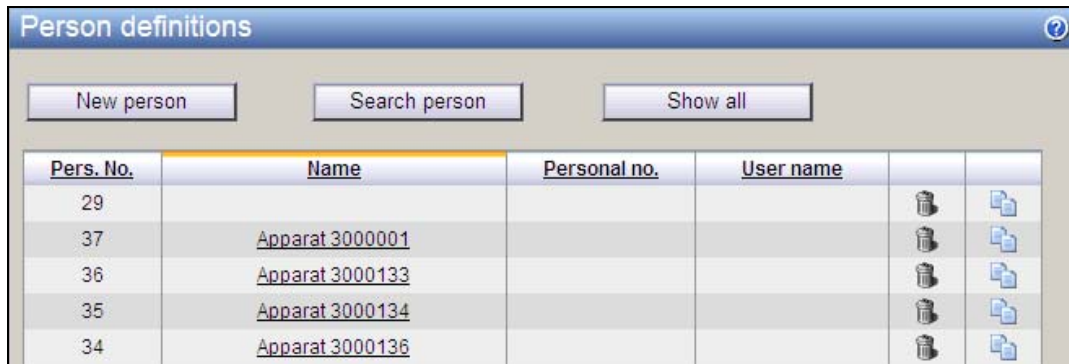
	On-call duty		On-call duty
Office 1:	<input type="text" value="3000001"/> <input checked="" type="checkbox"/>	Office 2:	<input type="text"/> <input checked="" type="checkbox"/>
Home 1:	<input type="text"/> <input checked="" type="checkbox"/>	Home 2:	<input type="text"/> <input checked="" type="checkbox"/>
Mobile 1:	<input type="text"/> <input checked="" type="checkbox"/>	Mobile 2:	<input type="text"/> <input checked="" type="checkbox"/>
SMS GSM 1:	<input type="text"/> <input checked="" type="checkbox"/>	SMS GSM 2:	<input type="text"/> <input checked="" type="checkbox"/>
DECT 1:	<input type="text"/> <input checked="" type="checkbox"/>	DECT 2:	<input type="text"/> <input checked="" type="checkbox"/>
Fax 1:	<input type="text"/> <input checked="" type="checkbox"/>	Fax 2:	<input type="text"/> <input checked="" type="checkbox"/>
Serial 1:	<input type="text"/> <input checked="" type="checkbox"/>	Serial 2:	<input type="text"/> <input checked="" type="checkbox"/>
Pager 1:	<input type="text"/> <input type="text" value="Tone call"/> <input checked="" type="checkbox"/>		
Pager 2:	<input type="text"/> <input type="text" value="Tone call"/> <input checked="" type="checkbox"/>		
E-Mail:	<input type="text"/> <input checked="" type="checkbox"/>		
PC-Name/IP:	<input type="text"/> <input checked="" type="checkbox"/>		

Save changes **Discard**

Figure 47: NovaAlert Edit Personal Telephone Numbers Screen

Repeat this for the other extensions which are used for testing.

The newly configured users are now listed in the “Person definitions” screen.



The screenshot shows the 'Person definitions' window with a title bar, a help icon, and three buttons: 'New person', 'Search person', and 'Show all'. Below these is a table with columns: 'Pers. No.', 'Name', 'Personal no.', 'User name', and two icons (trash and copy). The table contains five rows of data.

Pers. No.	Name	Personal no.	User name		
29					
37	Apparat 3000001				
36	Apparat 3000133				
35	Apparat 3000134				
34	Apparat 3000136				

Figure 48: NovaAlert Personal User Display Screen

3.4.3. Configure Alarms

Use the Windows “Start” button to select the “Alarm definition” icon from the left frame.

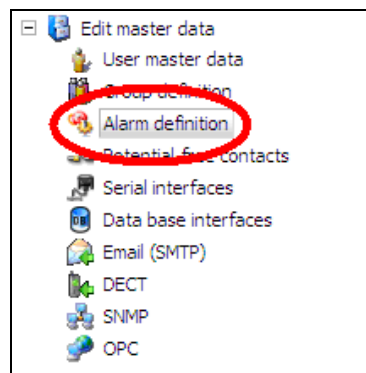


Figure 49: Alarm Definition Configuration Selection from NovaAlert Top-Level Screen

When the “Alarm definition” screen appears, click the “New Alarm” button.

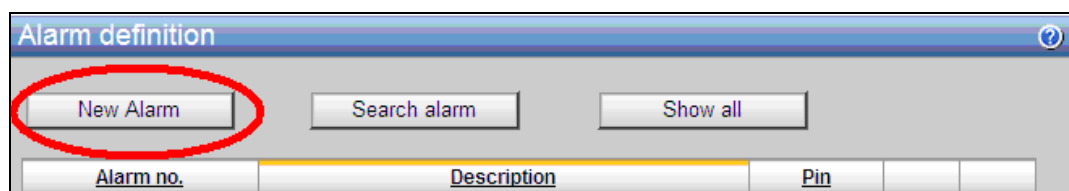


Figure 50: NovaAlert New Alarm Selection Screen

Configure the “General” alarm definition tab with the information shown in the following table.

Parameter	Usage
Description	Enter a name to be assigned to the alarm.
Pin code for trigger	Enter the PIN code to be used for alarm recipient verification
Priority	Select “Höchste Priorität” (highest priority) from the drop-down box.
Group call	Select “Group Call” from the drop-down box.
Nbr. Of pers. To be contac	Select “Alle” (all) from the drop-down box.
Based on person	Check this box.
Display on reports	Check this box.
Select contact group	Select the “Compile individual alert list” radio button.

Table 16: NovaAlert General Alarm Configuration Parameters

Figure 51: NovaAlert New Alarm Definition Screen

Configure the “Messages” alarm definition tab with the information shown in the following table and click the “Save changes” button.

Parameter	Column	Usage
Phone display	Alarm messages	Enter the text message which is to be shown on the Avaya telephone display. The length of this message should not exceed the maximum calling party name text length which can be displayed by Avaya telephones, which is 15 characters for telephones used for these tests.
	Event text	Select “Yes” from the drop-down box.
	Call type	Select “Dauer” from the drop-down box.
	copy	Select this check box.
Phone TTS		Select the check box which is positioned to the left of the “Alarm messages” text box.
	Alarm messages	Enter the text message which is converted to speech and announced to the party receiving the alarm.
	Event text	Select “Yes” from the drop-down box.

Table 17: NovaAlert Alarm Messages Configuration Parameters

The screenshot shows the 'Messages' tab in the NovaAlert configuration interface. The 'Alarm messages' column is populated with 'Testalarm'. The 'Event text' dropdown is set to 'Yes', and the 'Call type' dropdown is set to 'Dauer'. The 'copy' checkbox is checked. The 'Phone display' field is empty. The 'Phone TTS' checkbox is checked. The 'Calling Party' field is empty. The 'Alarm mess. before pin' and 'Alarm mess. before conference' checkboxes are unchecked. The 'Fix before indiv. voice mess.' checkbox is unchecked. The 'Save changes' button is circled in red.

Figure 52: NovaAlert Alarm Message Definition Screen

3.4.3.1 Configure Direct Alarms

The Direct Alarm function maps a specific Called Party Number to an alarm, so that when this number is dialed, the caller records a message (optional) and the alarm is triggered. The recorded message is played on the alarmed stations.

Use the Windows “Start” button to select the “Directalarms” icon from the left frame.

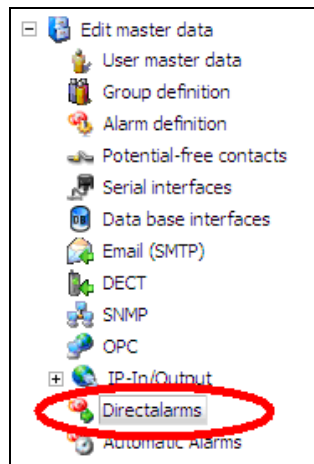


Figure 53: Direct Alarm Configuration Selection from NovaAlert Top-Level Screen

When the “Direct alarm” screen appears, click the “New Alarm” button.

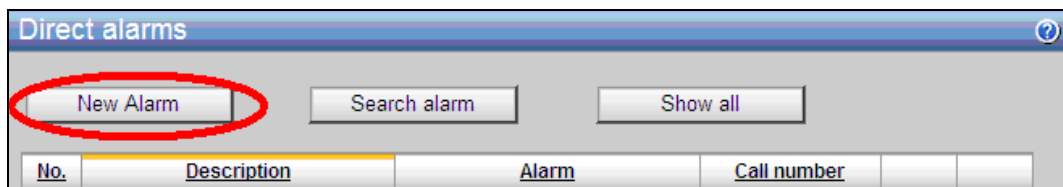


Figure 54: NovaAlert New Direct Alarm Selection Screen

Configure the “General” alarm definition tab with the information shown in the following table, and click the “Save changes” button.

Parameter	Usage
Description	Enter a name to be assigned to the alarm.
Initiating call number	Enter the telephone number which is to be used by NovaAlert to make the alarm call.
PIN Code	Enter a PIN code that needs to get input to trigger the alarm. Leave empty if none is required.
Alarm no.	Select one of the previously configured alarms from the drop-down box.
Alarm text	Input an alarm text to display on the alarmed stations (as Calling Party Name). Leave empty for the default alarm text.
Recording	Check this box to allow the recording of an alarm message per call.
Min. recording time	Enter the minimum recording time, in seconds.

Table 18: NovaAlert General Direct Alarm Configuration Parameters

Figure 55: NovaAlert Direct Alarm Configuration Screen

4. Interoperability Compliance Testing

The interoperability compliance tests included feature and serviceability testing.

The feature testing focused on testing scenarios that involve interaction between the NovaLink server and Avaya products, including various sequences involving the following:

- Verification of the correct delivery of alarm voice messages
- Verification of the correct display of alarm text messages
- Verification of the ability of NovaAlert to recognize DTMF tones.
- Verification of the ability of Avaya Telephones to correctly log unanswered alarms.

The serviceability testing focused on verifying that the NovaLink product components can recover from interruption to interface connections which can occur during routine maintenance activities. Each of these units was also tested for recovery from unexpected power interruption.

4.1. General Test Approach

The test method employed can be described as follows:

- The individual features of the NovaAlert were tested by manually generating alarms from the NovaAlert console and manually making calls from Avaya Telephones.
- NovaAlert server robustness was tested by verifying its ability to recover from interruptions to its external connections via the LAN between the NovaAlert and the network.
- Verifying the ability to recover from power interruptions to the NovaAlert server further tested its robustness.

All testing was performed manually. The tests were all functional in nature, and no performance testing was done.

4.2. Test Results

The following was observed during testing:

- It is not possible for NovaAlert to detect that an Avaya 4600 Series H.323 phone is disconnected, as Avaya Communication Manager does not report this status to the caller.

5. Verification Steps

The following steps can be performed to verify the basic operation of the various system components:

- Verify that Avaya Communication Manager and the NovaAlert server can ping each other. The “ping” command can be executed from the NovaAlert server by executing the “cmd” component via the run facility from the Windows “Start” control and entering “ping” followed by the IP address to which the ping message is to be sent. The “ping” command can be executed from Avaya Communication Manager via an SSH login session.

- Verify that the Avaya IP Telephones can call each other.
- From the Avaya Communication Manager SAT terminal, use the “status trunk” command to verify that the ports for the trunk connected to NovaAlert are in the “in-service/idle” state.
- Verify that each of the Avaya Telephones can call the extension allocated to NovaAlert to participate in an incoming conference.
- Verify that it is possible for NovaAlert to call each of the Avaya IP Telephones to participate in an outgoing conference.
- Verify that it is possible to navigate the NovaAlert voice menu from each of the Avaya Telephones by calling the NovaAlert extension, and entering key sequences in response to prompting requests from NovaAlert.
- Verify the ability of Avaya Telephones to correctly log unanswered calls by initiating an unanswered alarm call from NovaAlert to each of the Avaya Telephones, verifying the name and number in the log of the telephone, and subsequently dialing the caller from the telephone log.
- From the Avaya SES Maintenance Web Interface, select the “Status Summary” screen and verify that the server is in “Active” mode, no alarms are being generated, the “Server Hardware” is “okay”, and that server “Processes” are “okay”.
- Verify that it is possible to place calls between SIP and H.323 Telephones.

6. Support

Technical support from NovaLink can be obtained through the following:

NovaLink GmbH
 Businessstower
 Zuercherstrasse 310
 8500 Frauenfeld
 Switzerland
helpdesk@novalink.ch
 Phone: +41 52 762 66 77
 Fax: +41 52 762 66 99

7. Conclusion

These Application Notes describe the conformance testing of the NovaAlert with Avaya Communication Manager. The various features of the NovaAlert that involve its telephone interface were tested. A detailed description of the configuration required for both the Avaya and the NovaLink equipment is documented within these Application Notes. NovaAlert passed all of the tests performed, which included both functional and robustness tests.

8. Additional References

- [1] *Administrator Guide for Avaya Communication Manager*, February 2007, Issue 3, Document Number 03-300509
- [2] *Feature Description and Implementation for Avaya Communication Manager*, February 2007, Issue 5, Document Number 555-245-205
- [3] *Installing and Administering SIP Enablement Services*, March 2007, Issue 2.1, Document Number 03-600768
- [4] *NovaAlert 7.5 manual*, May 2007

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