



Avaya Solution & Interoperability Test Lab

Application Notes for Syn-Apps' SA-Announce with Avaya 9600 Series IP Deskphones – Issue 1.0

Abstract

These Application Notes describe a compliance-tested configuration consisting Avaya 9600 Series IP Deskphones and Syn-Apps SA-Announce.

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

1. Introduction

Syn-Apps' SA-Announce is an enhanced paging and mass notification solution that integrates with Avaya Aura and Avaya 9600 Series Deskphones; providing audio, text and graphic notifications across an organization. SA-Announce delivers real-time, pre-recorded or scheduled announcements to streamline critical situation communication and many operational processes.

Avaya 9600 Series IP Deskphones subscribe to Syn-Apps SA-Announce, to receive XML-based data pushed by SA-Announce. The data that is pushed by SA-Announce is in the form of Alerts. In addition, SA-Announce has the ability to send Multicast audio to Avaya 9600 Series IP Deskphones.

2. General Test Approach and Test Results

The compliance test focused on the interoperability between Avaya 9600 Series IP Deskphones and Syn-Apps SA-Announce.

DevConnect Compliance Testing is conducted jointly by Avaya and DevConnect members. The jointly-defined test plan focuses on exercising APIs and/or standards-based interfaces pertinent to the interoperability of the tested products and their functionalities. DevConnect Compliance Testing is not intended to substitute full product performance or feature testing performed by DevConnect members, nor is it to be construed as an endorsement by Avaya of the suitability or completeness of a DevConnect member's solution.

2.1. Interoperability Compliance Testing

Compliance testing focused on receiving various types of Alerts and Notifications sent by SA-Announce to Avaya 9600 Series IP Deskphones. The following Alert Types were tested during Compliance test:

- Weather Alerts
- Amber Alerts
- Emergency Alerts
- RecordNPlay Notifications

Only Avaya 9600 Series H.323 phones were included in the test. Avaya 9600 Series SIP phones were not included due to the lack of support for Multicast audio. The following models were tested:

- 9611
- 9620
- 9621
- 9630
- 9640
- 9641
- 9670

2.2. Test Results

All executed test cases were passed and all objectives were met.

2.3. Support

Syn-Apps support can be contacted in the following ways:

Phone: 866-664-6071**Email:** support@syn-apps.com

Web Form: <http://www.syn-apps.com/support/request/>

3. Reference Configuration

Figure 1 below displays a sample configuration that was tested during the compliance test.

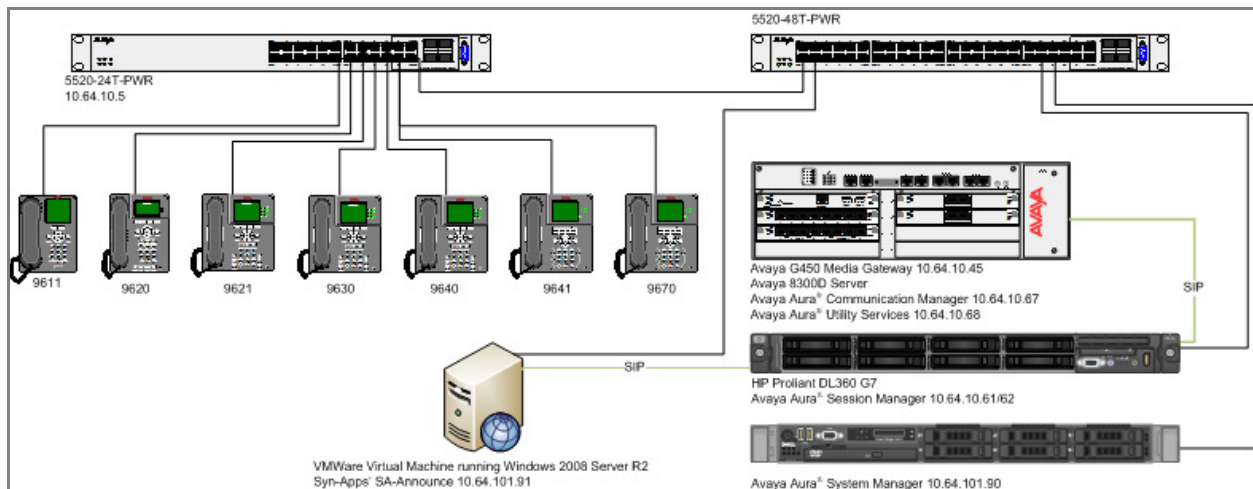


Figure 1: Reference Configuration

4. Equipment and Software Validated

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

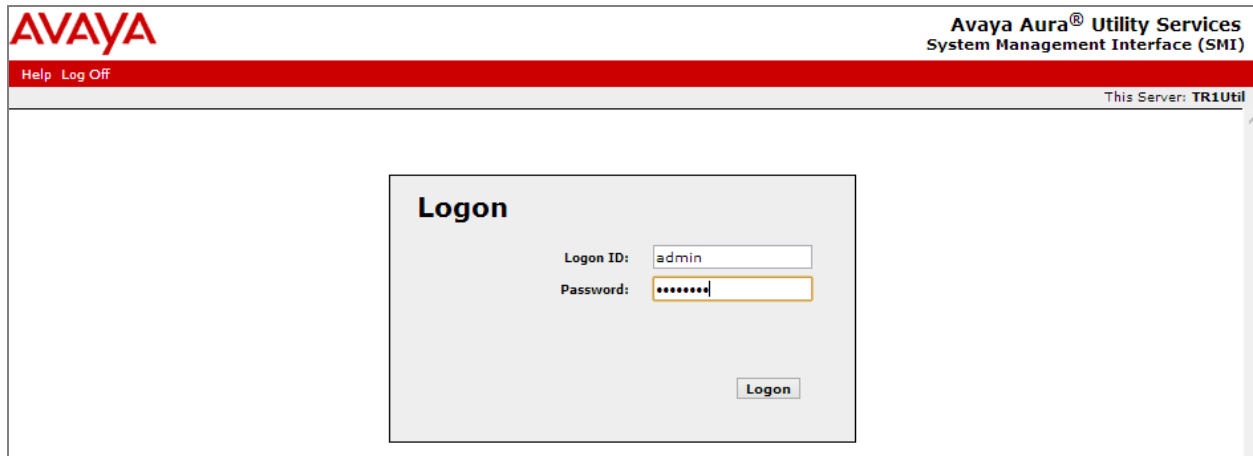
Equipment	Version
Avaya Aura® System Manager running on VMware EXSi 5.1 infrastructure	R6.2.12 Build 6.2.0.0.15669-6.2.12.408
Avaya Aura® Session Manager running on HP Proliant DL360 G7 server	6.2.3.0.622006
Avaya Aura® Communication Manager running on Avaya 8300D server	R6.2 build R016x.02.0.823.0
Avaya G450 Media Gateway	31.20.1
Avaya 96x1 Series H.323 Phones	6.2.4
Avaya 96x0 Series H.323 Phones	3.10
Avaya Aura® Utility Services	6.2.0.0.15
Syn-Apps SA-Announce running on a Windows Server2008 R2 VMWare Virtual Machine	9.0.10

5. Configure Avaya Aura® Utility Server

Avaya 9600 Series IP Deskphones settings are controlled by 46xxsettings.txt file that is downloaded and parsed each time a phone resets/reboots. In order to interact with the phones, SA-Announce has to be "trusted" by the phones and the phones need to "subscribe" to SA-Announce. Furthermore, in order for a phone to receive and display content on the screen from a remote source, it needs to be configured for WML (Wireless Mark-up Language).

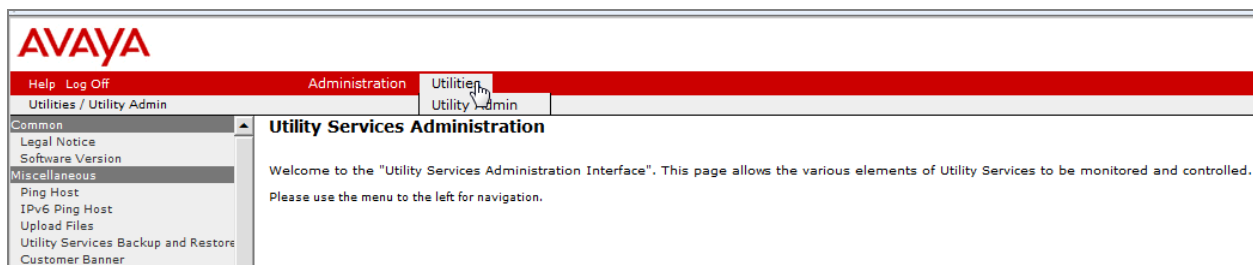
In the Interoperability Lab, Avaya Aura® Utility Server is used to manage Avaya 9600 Series IP Deskphones. All changes to the 46xxsetting.txt are made via Avaya Aura® Utility Services System Management Interface (SMI). Utility Service SMI can be reached via a web browser, <http://<ip-address>/admin.html>, where ip-address is the IP Address of Avaya Aura® Utility Services.

On the logon page, log in using appropriate credentials.



The screenshot shows the Avaya Aura® Utility Services System Management Interface (SMI) Logon page. The page has a red header with the Avaya logo on the left and the text "Avaya Aura® Utility Services System Management Interface (SMI)" on the right. Below the header, there is a red bar with "Help" and "Log Off" links. On the right side of the page, it says "This Server: TR1Util". The main content area is a light gray box with the title "Logon". Inside this box, there are two input fields: "Logon ID:" with the value "admin" and "Password:" with a masked password "*****". Below these fields is a "Logon" button.

Once logged in, navigate to **Utilities → Utilities Admin.**



The screenshot shows the Avaya Aura® Utility Services Administration page. The page has a red header with the Avaya logo on the left and the text "Avaya Aura® Utility Services Administration" on the right. Below the header, there is a red bar with "Help" and "Log Off" links. On the right side of the page, it says "This Server: TR1Util". The main content area is a light gray box with the title "Utility Services Administration". Inside this box, there is a navigation menu on the left with the following items: "Common", "Legal Notice", "Software Version", "Miscellaneous", "Ping Host", "IPv6 Ping Host", "Upload Files", "Utility Services Backup and Restore", and "Customer Banner". The "Miscellaneous" item is selected. The main content area displays the text: "Welcome to the 'Utility Services Administration Interface'. This page allows the various elements of Utility Services to be monitored and controlled. Please use the menu to the left for navigation."

Then, navigate to **IP Phone Tools** → **IP Phone Settings Editor**.

AVAYA

Avaya Aura® Utility Services
System Management Interface (SMI)

Help Log OffAdministrationUtilities

This Server: TR1Util

Common
Legal Notice
Software Version
Miscellaneous
Ping Host
IPv6 Ping Host
Upload Files
Utility Services Backup and Restore
Customer Banner
Firewall Rules
Firewall (IPv4)
Firewall (IPv6)
IP Phone Tools
ADVD Settings Editor
IP Phone Settings Editor
IP Phone Backup and Restore
IP Phone Custom File Upload
IP Phone Firmware Manager
Configure CM Login
Display Stations
Display Server Firmware
Manage Phone Firmware
Schedule Phone File Download
DHCP Manager
DHCP Server Status
Activate/Deactivate DHCP
DHCP IP Address Pools
Show DHCP Leases

Utility Services Administration

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On the **IP Phone Settings Editor**, Select **Proceed with Selected Values**.

AVAYA

Avaya Aura® Utility Services
System Management Interface (SMI)

Help Log OffAdministrationUtilities

This Server: TR1Util

Common
Legal Notice
Software Version
Miscellaneous
Ping Host
IPv6 Ping Host
Upload Files
Utility Services Backup and Restore
Customer Banner
Firewall Rules
Firewall (IPv4)
Firewall (IPv6)
IP Phone Tools
ADVD Settings Editor
IP Phone Settings Editor
IP Phone Backup and Restore
IP Phone Custom File Upload
IP Phone Firmware Manager
Configure CM Login
Display Stations
Display Server Firmware
Manage Phone Firmware
Schedule Phone File Download
DHCP Manager
DHCP Server Status
Activate/Deactivate DHCP
DHCP IP Address Pools
Show DHCP Leases
DHCP Server Log

IP Phone Settings Editor

This page allows you to check and edit the 46xxsettings.txt file

Please select display options

☒ Display file comments
☐ Display only active options

Please select a settings file to edit

☒

(URL to this server's settings file is http://10.64.10.68/46xxsettings.txt)

☐ Upload IP phone settings or xml file No file chosen

☐ Check for latest xml file via web

Enable the following options by selecting the check box in front of the options and setting their fields as follows:

- **TPSLIST**: Set it to the IP Address of SA-Announce server
- **SUBSCRIBELIST**: Set it to the following URL
 - http://<ip-address>/SA-Announce/PhoneServices/AvayaPhoneRegistration.aspx

Note: ip-address is the IP Address of SA-Announce server

- **WMLHOME**: Configure with a suitable URL as a home page
- **PUSHCAL**: Set it to 22222


Below is a screen capture of configuring TPSLIST, other options are not shown.

File Server		46xx H.323 R2.1 and later
Call Detail Recording		16xx H.323 R1.0 and later
Phone Firmware Manager		
System Database		
MyPhone	<input checked="" type="checkbox"/>	TPSLIST 10.64.101.91
TFTP Server		
Call Detail Record Tools		

Once done, select **Save New Setting File** at the bottom of the page.

Call Detail Record Tools		END
CDR Reports		END OF CONFIGURATION FILE
CDR Backups		
CDR Archive		
CDR E-Mails		
		<input type="button" value="Re-evaluate Settings"/> <input type="button" value="Save New Settings File"/>
		Lines read in 6308

Verify the values of the options above on the **Output Screen**, and select **Save 46xxsetting.txt file to this server**.

		Avaya Aura® Utility Services System Management Interface (SMI)
Help Log Off		
Administration Utilities		
This Server: TR10til		
Common Legal Notice Software Version Miscellaneous Ping Host IPv6 Ping Host Upload Files Utility Services Backup and Restore Customer Banner Firewall Rules	IP Phone Settings Editor This page allows you to check and edit the 46xxsettings.txt file <div style="text-align: center;"> Output Screen Click to save to this server <input type="button" value="Save 46xxsettings.txt file to this server"/> </div>	

On the next page, select **Continue**.

AVAYA		Avaya Aura® Utility Services System Management Interface (SMI)	
Help Log Off		Administration Utilities	
		This Server: TR1Util	
Common	IP Phone Settings Editor		
Legal Notice			
Software Version			
Miscellaneous	This page allows you to check and edit the 46xxsettings.txt file		
Ping Host	Settings file saved to system		
IPv6 Ping Host	IP phones must be rebooted to acquire new settings		
Upload Files			
Utility Services Backup and Restore			
Customer Banner	<input type="button" value="Continue"/> <input type="button" value="Help"/>		
Firewall Rules			

Reboot the Avaya 9600 Series IP Deskphones to update the settings.

6. Configure Avaya Aura® Communication Manager

Avaya Aura® Communication Manager allows for routing calls to SA-Announce via Avaya Aura® Session Manager using SIP trunks. In order for Avaya 9600 Series IP Deskphones to be able to dial a number to activate a SA-Announce group, a SIP trunk must be created that communicates with Session Manager. Another SIP trunk will be created on Session Manager to communicate to SA-Announce. The following information allows for a SIP connection between the Communication Manager and Session Manager.

6.1. Configure IP Network Region

Use the **change ip-network-region *n*** command to configure a network region, where ***n*** is an existing network region. Configure this network region as follows:

- Set **Location** to **1**
- Set **Codec Set** to **1**
- Set **Intra-region IP-IP Direct Audio** to **yes**
- Set **Inter-region IP-IP Direct Audio** to **yes**
- Enter and **Authoritative Domain**, e.g. **avaya.com**

```
change ip-network-region 1                                     Page 1 of 20
                                                              IP NETWORK REGION
    Region: 1
    Location: 1          Authoritative Domain: avaya.com
    Name:
    MEDIA PARAMETERS          Intra-region IP-IP Direct Audio: yes
    Codec Set: 1            Inter-region IP-IP Direct Audio: yes
    UDP Port Min: 2048          IP Audio Hairpinning? n
    UDP Port Max: 3329
    DIFFSERV/TOS PARAMETERS
    Call Control PHB Value: 46
    Audio PHB Value: 46
    Video PHB Value: 26
    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 DESKPHONES          RSVP Enabled? n
    H.323 Link Bounce Recovery? y
    Idle Traffic Interval (sec): 20
    Keep-Alive Interval (sec): 5
```

6.2. Administer IP Codec Set

Use the **change ip-codec-set *n*** command to configure IP codec set, where *n* is an existing codec set number. Configure this codec set as follows, on **Page 1**:

- Set **Audio Codec 1** to **G.711MU**

change ip-codec-set 1				Page 1 of 2
IP Codec Set				
Codec Set: 1				
Audio Codec	Silence Suppression	Frames Per Pkt	Packet Size (ms)	
1: G.711MU	n	2	20	
2: G.711A	n	2	20	
3: G.729AB	n	2	20	
4:				
5:				
6:				
7:				

6.3. Administer IP Node Names

Use the **change node-names ip** command to add an entry for Session Manager. For compliance testing, **sm** and **10.64.10.62** entry was added.

change node-names ip		Page 1 of 2
IP NODE NAMES		
Name	IP Address	
default	0.0.0.0	
msgsrvr	192.168.62.28	
procr	192.168.62.28	
procr6	::	
sm	10.64.10.62	

6.4. Administer SIP Signaling Group

Use the **add signaling-group *n*** command to add a new signaling group, where *n* is an available signaling group number. Configure this signaling group as follows:

- Set **Group Type** to **sip**
- Set **Near-end Node Name** to **procr**
- Set **Far-end Node Name** to the configured Session Manager in **Section 6.3**, i.e. **sm**
- Set **Far-end Network region** to the configured region in **Section 6.1**, i.e. **1**
- Enter a **Far-end Domain**, e.g. **avaya.com**

add signaling-group 10		Page 1 of 2
SIGNALING GROUP		
Group Number: 1	Group Type: sip	
IMS Enabled? n	Transport Method: tls	
Q-SIP? n		
IP Video? n	Enforce SIPS URI for SRTP? y	
Peer Detection Enabled? y	Peer Server: Others	
Near-end Node Name: procr	Far-end Node Name: sm	
Near-end Listen Port: 5061	Far-end Listen Port: 5061	
Far-end Network Region: 1		
Far-end Domain: avaya.com		
Incoming Dialog Loopbacks: eliminate	Bypass If IP Threshold Exceeded? n	
DTMF over IP: rtp-payload	RFC 3389 Comfort Noise? n	
Session Establishment Timer(min): 3	Direct IP-IP Audio Connections? y	
Enable Layer 3 Test? y	IP Audio Hairpinning? n	
H.323 Station Outgoing Direct Media? n	Initial IP-IP Direct Media? n	
	Alternate Route Timer(sec): 6	

6.5. Administer SIP Trunk Group

Use the **add trunk-group *n*** command to add a trunk group, where *n* is an available trunk group number. Configure this trunk group as follows, on **Page 1**:

- Set **Group Type** to **sip**
- Enter a **Group Name**
- Enter a valid **TAC**, e.g. *010
- Set **Service Type** to **tie**
- Enter **Signaling Group** value to the signaling group configured in **Section 6.4**, i.e. 10
- Enter a desired number in **Number of Members** field

add trunk-group 10		Page 1 of 21	
TRUNK GROUP			
Group Number: 1	Group Type: sip	CDR Reports: y	
Group Name: Session Manager	COR: 1	TN: 1	TAC: *010
Direction: two-way	Outgoing Display? n	Night Service:	
Dial Access? n			
Queue Length: 0			
Service Type: tie	Auth Code? n		
	Member Assignment Method: auto		
	Signaling Group: 10		
	Number of Members: 25		

On **Page 3**:

- Set **Number Format** to **private**

add trunk-group 10		Page 3 of 21	
TRUNK FEATURES			
ACA Assignment? n	Measured: none	Maintenance Tests? y	
Numbering Format: private			
UI Treatment: service-provider			
Replace Restricted Numbers? n			
Replace Unavailable Numbers? n			

6.6. Administer Route Pattern

Use the **change route-pattern *n*** command to configure a route pattern, where *n* is an available route patterns. Configure this route pattern as follows:

- Type a name in **Pattern Name** field
- For line 1, set **Grp No** to the trunk group configured in **Section 6.5**, i.e. 10
- For line 1, set **FRL** to 0

change route-pattern 10												Page	1	of	3
Pattern Number: 1												Pattern Name: Voice and Fax			
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:	10	0										n	user		
2:											n	user			

6.7. Administer AAR Analysis

Use the **change aar analysis *n*** command to configure routing for extensions starting with *n*. For compliance testing, extensions starting with 26 were used to route calls to SA-Announce:

- Set **Dialed String** to starting digits of extensions that will be used, e.g. 26
- Set **Min** and **Max** to 5 for 5 digit extensions
- Set **Route Pattern** to pattern configured in **Section 6.6**, i.e. 10
- Set **Call Type** to **aar**

change aar analysis 26							Page	1	of	2
AAR DIGIT ANALYSIS TABLE										
Location: all							Percent Full: 1			
	Dialed	Total		Route	Call	Node	ANI			
	String	Min	Max	Pattern	Type	Num	Reqd			
26		5	5	10	aar		n			
27		5	5	21	aar		n			
275		5	5	10	aar		n			
29		5	5	10	aar		n			
3		7	7	254	aar		n			
4		5	5	2	aar		n			
45000		5	5	30	aar		n			
5		5	5	32	aar		n			

6.8. Administer Private Numbering

Use the **change private-numbering 1** command to define the calling party number to send to Session Manager.

Configure private numbering as follows:

- During the compliance test, extensions starting with 2 and were 5 digits long were used; calls from these extension were made to SA-Announce via trunk group configured in **Section 6.5**.

change private-numbering 1					Page 1 of 2
NUMBERING - PRIVATE FORMAT					
Ext	Ext	Trk	Private	Total	
Len	Code	Grp(s)	Prefix	Len	
5	2	10		5	Total Administered: 1
5	5			5	Maximum Entries: 540

6.9. Administer Dial Plan Analysis and Stations

Administration of Dial Plan Analysis and Avaya Stations/Extensions in Communication Manager is not shown in this document. Please refer to document [1] in reference section of this document.

7. Configure Avaya Aura® Session Manager

Access the Session Manager Administration web interface by entering <https://<ip-address>/SMGR> URL in a web browser, where <ip-address> is the IP address of System Manager. Log in using appropriate credentials.

AVAYA Avaya Aura® System Manager 6.3

Home / Log On

Log On

Recommended access to System Manager is via FQDN.

[Go to central login for Single Sign-On](#)

If IP address access is your only option, then note that authentication will fail in the following cases:

- First time login with "admin" account
- Expired/Reset passwords

Use the "Change Password" hyperlink on this page to change the password manually, and then login.

User ID:

Password:

Supported Browsers: Internet Explorer 8.x, 9.x or 10.x or Firefox 15.0, 16.0 or 17.0. [Change Password](#)

Once logged in, the dashboard is displayed.

AVAYA Avaya Aura® System Manager 6.3 Last Logged on at July 25, 2013 9:29 AM Help | About | Change Password | Log out

Users	Elements	Services
Administrators Manage Administrative Users	Communication Manager Manage Communication Manager 5.2 and higher elements	Backup and Restore Backup and restore System Manager database
Directory Synchronization Synchronize users with the enterprise directory	Communication Server 1000 Manage Communication Server 1000 elements	Bulk Import and Export Manage Bulk Import and Export of Users, User Global Settings, Roles, Elements and others
Groups & Roles Manage groups, roles and assign roles to users	Conferencing Manage Conferencing Multimedia Server objects	Configurations Manage system wide configurations
User Management Manage users, shared user resources and provision users	IP Office Manage IP Office elements	Events Manage alarms, view and harvest logs
	Meeting Exchange Manage Meeting Exchange and Avaya Aura Conferencing 6.0 elements	Geographic Redundancy Manage Geographic Redundancy
	Messaging Manage Avaya Aura Messaging, Communication Manager Messaging, and Modular Messaging	Inventory Manage, discover, and navigate to elements
	Presence Presence	Licenses View and configure licenses
	Routing Session Manager Routing Administration	Replication Track data replication nodes, repair replication nodes
	Session Manager Session Manager Administration, Status, Maintenance and Performance Management	Scheduler Schedule, track, cancel, update and delete jobs
		Security Manage Security Certificates

7.1. Add SIP Domain

Navigate to **Home → Elements → Routing → Domains**, click on **New** button (not shown) and configure as follows:

- In **Name** field type in a domain (authoritative domain used in **Section 6.1**) i.e. avaya.com
- Set **Type** to **sip**

Click **Commit** to save changes.

AVAYA Avaya Aura® System Manager 6.3

Last Logged on at July 25, 2013 9:29 AM
Help | About | Change Password | Log out

Routing x Home

Home / Elements / Routing / Domains

Domain Management

1 Item | Refresh Filter: Enable

Name	Type	Notes
* avaya.com	sip	

Commit Cancel

7.2. Add Location

Navigate to **Home → Elements → Routing → Location**, click on **New** button (not shown) and configure as follows:

Under **General**:

- Type in a descriptive **Name**

Under **Location Pattern** click on **Add** (not shown):

- Type in an **IP Address Pattern**, e.g. 10.64.10.*

Note: During compliance test, IP addresses in 10.64.10.* and 10.64.101.* were used. Also, the “*” refers to wildcard indicating a range.

Click **Commit** to save changes. Screen shot shown on next page.

Location Details
Commit
Cancel

General

* Name:

Notes:

Dial Plan Transparency in Survivable Mode

Enabled: ☐

Listed Directory Number:

Associated CM SIP Entity:

Overall Managed Bandwidth

Managed Bandwidth Units:

Total Bandwidth:

Multimedia Bandwidth:

Audio Calls Can Take Multimedia Bandwidth: ☒

Per-Call Bandwidth Parameters

Maximum Multimedia Bandwidth (Intra-Location): Kbit/Sec

Maximum Multimedia Bandwidth (Inter-Location): Kbit/Sec

* Minimum Multimedia Bandwidth: Kbit/Sec

* Default Audio Bandwidth:

Alarm Threshold

Overall Alarm Threshold: %

Multimedia Alarm Threshold: %

* Latency before Overall Alarm Trigger: Minutes

* Latency before Multimedia Alarm Trigger: Minutes

Location Pattern

Add Remove

2 Items Refresh
Filter: Enable

<input type="checkbox"/>	IP Address Pattern	Notes
<input type="checkbox"/>	* 10.64.10.*	<input type="text"/>
<input type="checkbox"/>	* 10.64.101.*	<input type="text"/>

Select : All, None

7.3. Add SIP Entity – Communication Manager

Each SIP device that communicates with the Session Manager over a SIP trunk, requires a SIP Entity configuration. Add Communication Manager as a SIP Entity. Navigate to **Home → Elements → Routing → SIP Entities**, click on **New** (no shown) and configure as follows:

- Type in a descriptive name in **Name** field
- Type in the FQDN or IP address of the Communication Manager in **FQDN or IP Address** field.
- Set **Type** to **CM**
- Set **Location** to the location configured in **Section 7.2**

Click **Commit** to save changes.

Note: It is assumed that SIP Entity for Session Manager has been already configured.

The screenshot shows the 'SIP Entity Details' configuration page. At the top right, there are 'Commit' and 'Cancel' buttons and a 'Help ?' link. The page is divided into two sections: 'General' and 'SIP Link Monitoring'. In the 'General' section, the following fields are visible: 'Name' (cm-tr1), 'FQDN or IP Address' (10.64.10.67), 'Type' (Session Manager), 'Notes' (empty), 'Location' (Test Room 1), 'Outbound Proxy' (empty), 'Time Zone' (America/Fortaleza), and 'Credential name' (empty). The 'SIP Link Monitoring' section contains a single dropdown menu labeled 'SIP Link Monitoring' set to 'Use Session Manager Configuration'.

SIP Entity Details		Commit	Cancel	Help ?
General				
* Name:	cm-tr1			
* FQDN or IP Address:	10.64.10.67			
Type:	Session Manager			
Notes:				
Location:	Test Room 1			
Outbound Proxy:				
Time Zone:	America/Fortaleza			
Credential name:				
SIP Link Monitoring				
SIP Link Monitoring:	Use Session Manager Configuration			

7.4. Add Adaptation

Adaptations are used to manipulate digits (via Digit Conversion) and SIP URIs (Via Module and Egress Parameters) for incoming and outgoing calls. Navigate to **Home → Elements → Routing → Adaptation**, click **New** (not shown) and configure as follows:

- Type in a descriptive name in **Adaptation Name** field
- Type in **DigitConversionAdapter** in **New Module Name** field
- In the **Module Parameter** field type in the following:
 - iodstd=avaya.com odstdd=ip-address fromto=true

Note: ip-address is the IP Address of SA-Announce.

Note: Module Parameters used during this test performed the following function.

- **fromto:** Modifies From and To headers of a SIP message.
- **idstd** (overrideDestinationDomain): Replaces the domain in Request-URI, To header, Refer-To header, and Notify/message-summary body with the given value for egress only. Egress refers to call routing out of Session Manager.
- **iodstd** (ingressOverrideDestinationDomain) : Replaces the domain in Request-URI, To header, and Notify/message-summary body with the given value for ingress only. Ingress refers to calls arriving into Session Manager.

Click **Commit** to save the changes.

Adaptation Details

Commit Cancel

Help ?

General

* Adaptation name: sa-tr1

New module name: DigitConversionAdapter

Module parameter: iodstd=avaya.com odstdd=10.64

Egress URI Parameters:

Notes:

Digit Conversion for Incoming Calls to SM

Add Remove

0 Items Refresh

Filter: Enable

	Matching Pattern	Min	Max	Phone Context	Delete Digits	Insert Digits	Address to modify	Adaptation Data	Notes
--	------------------	-----	-----	---------------	---------------	---------------	-------------------	-----------------	-------

Digit Conversion for Outgoing Calls from SM

Add Remove

0 Items Refresh

Filter: Enable

	Matching Pattern	Min	Max	Phone Context	Delete Digits	Insert Digits	Address to modify	Adaptation Data	Notes
--	------------------	-----	-----	---------------	---------------	---------------	-------------------	-----------------	-------

Commit Cancel

7.5. Add SIP Entity – SA-Announce

Add SA-Announce as a SIP Entity. Navigate to **Home → Elements → Routing → SIP Entities**, click on **New** (no shown) and configure as follows:

- Type in a descriptive name in **Name** field
- Type in the FQDN or IP address of the SA-Announce server in **FQDN or IP Address** field
- Set **Type** to **SIP Trunk**
- Set **Adaptation** to the one configured in previous section
- Set **Location** to the location configured in **Section 7.2**

Click **Commit** to save the changes.

SIP Entity DetailsCommitCancelHelp ?

General

*** Name:** sa-tr1

*** FQDN or IP Address:** 10.64.101.91

Type: SIP Trunk

Notes: SA-Announce

Adaptation: sa-tr1

Location: Test Room 1

Time Zone: America/Denver

Override Port & Transport with DNS SRV: ☐

*** SIP Timer B/F (in seconds):** 4

Credential name:

Call Detail Recording: egress

Loop Detection

Loop Detection Mode: Off

SIP Link Monitoring

SIP Link Monitoring: Use Session Manager Configuration

7.6. Add Entity Link – Communication Manager

A SIP Trunk between a Session Manager and another SIP entity is described by an Entity Link. Navigate to **Home → Elements → Routing → Entity Links**, click on **New** (not shown) and configure as follows:

- Type in a descriptive name in **Name** field
- Set **SIP Entity 1** to the name of Session Manager SIP Entity
- Set **SIP Entity 2** to Communication Manager SIP Entity configured in **Section 7.3**

Click **Commit** to save changes.

Entity Links Commit Cancel

1 Item | Refresh Filter: Enable

<input type="checkbox"/>	Name	SIP Entity 1	Protocol	Port	SIP Entity 2	Port	Connection Policy	Deny New Service	N
<input type="checkbox"/>	* Communication Ma	* asm-tr1 ▼	TLS ▼	* 5061	* cm-tr1 ▼	* 5061	trusted ▼	<input type="checkbox"/>	

< >

Select : All, None

7.7. Add Entity Link – SA-Announce

Navigate to **Home → Elements → Routing → Entity Links**, click on **New** (not shown) and configure as follows:

- Type in a descriptive name in **Name** field
- Set **SIP Entity 1** to the name of Session Manager SIP Entity
- Set **Protocol** to **TCP**
- Set **SIP Entity 2** to Communication Manager SIP Entity configured in **Section 7.5**

Note: SA-Announce only supports **TCP**.

Click **Commit** to save changes.

7.8. Add Time Ranges

Navigate to **Home → Elements → Routing → Time Ranges**, click on **New** (now shown) and configure as follows:

- Type in a descriptive name in **Name** field

Click **Commit** to save changes.

7.9. Add Routing Policy

Session Manager uses the data configured in the Routing Policy to find the best match against a number or address of the called party configured in a Dial Pattern. Dial pattern is configured once a Routing Policy is added. Navigate to **Home → Elements → Routing → Routing Policies**, click on **New** (not shown) and configure as follows:

- Type in a descriptive name in **Name** field
- Under **SIP Entity as Destination**, click on **Select** (not shown):
 - Select SA-Announce SIP entity added in **Section 7.5**
- Under **Time of Day**, click on **Add** (not shown):
 - Select time range added in previous step

Click **Commit** to save changes.

Routing Policy DetailsCommitCancel

General

*** Name:**

Disabled: ☐

*** Retries:**

Notes:

SIP Entity as Destination
Select

Name	FQDN or IP Address	Type	Notes
sa-tr1	10.64.101.91	SIP Trunk	SA-Announce

Time of Day
Add Remove View Gaps/Overlaps

1 Item | RefreshFilter: Enable

<input type="checkbox"/>	Ranking ▲	Name	Mon	Tue	Wed	Thu	Fri	Sat	Sun	Start Time	End Time	Notes
<input type="checkbox"/>	<input type="text" value="0"/>	24/7	✓	✓	✓	✓	✓	✓	✓	00:00	23:59	

Select : All, None

7.10. Add Dial Patterns

Navigate to **Home → Elements → Routing → Dial Patterns**, click on **New** (not shown) and configure as follows:

Under **General**:

- Set **Pattern** to prefix of dialed number
- Set **Min** to minimum length of dialed number
- Set **Max** to maximum length of dialed number

During the compliance test, called numbers starting with digits 26 and 5 digits long were routed to SA-Announce.

Under **Originating Locations and Routing Policies**:

- Click **Add** and select originating location and SA-Announce routing policy as configured in **Section 7.9**

Click **Commit** to save changes.

Dial Pattern Details

CommitCancelHelp ?

General

* Pattern: 26

* Min: 5

* Max: 5

Emergency Call: ☐

Emergency Priority: 1

Emergency Type:

SIP Domain: -ALL-

Notes:

Originating Locations and Routing Policies

AddRemove

1 Item | RefreshFilter: Enable

<input type="checkbox"/>	Originating Location Name	Originating Location Notes	Routing Policy Name	Rank	Routing Policy Disabled	Routing Policy Destination	Routing Policy Notes
<input type="checkbox"/>	Test Room 1		sa-tr1	0	<input type="checkbox"/>	sa-tr1	

Select : All, None

8. Configure SA-Announce

SA-Announce is installed on a Windows Server platform. Windows Server 2008 R2 was used during compliance-testing. For other Windows Server configurations, please reference the SA-Announce Avaya User Guide. A link to the document has been provided in **Section 11**.

8.1. Server Requirements

The Windows Server requirements are as follows:

Server Requirement	Description
Manual Setup Requirement	These require manual setup.
Disabling Public Firewall	Firewall must be disabled or set to allow access on the required ports (see Firewall Ports).
Auto-Setup Requirement	The installer should automatically complete these.
Disabling IE Enhanced Security Configuration	IE Enhanced Security Configuration must be disabled to allow the website to function properly.
Creating ASP.NET user account	
Installing Internet Information Service (IIS)	IIS is required for the product to function.
Installing ASP.NET	ASP.NET 4.0 is required for the product to function.
Recommended	These are recommended items.
Disabling User Account Control (UAC)	SA-Announce recommends disabling UAC.

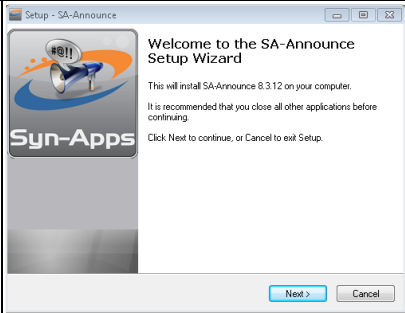
8.2. Network Requirements

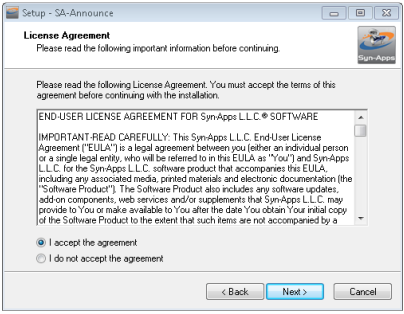
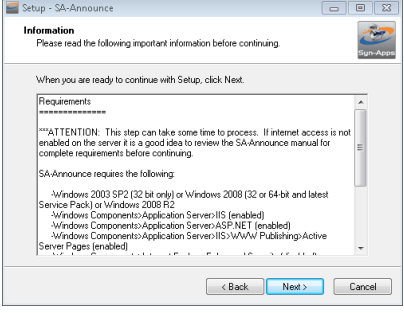
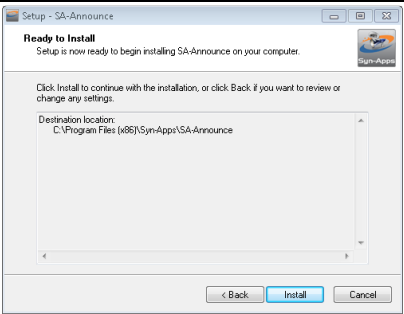
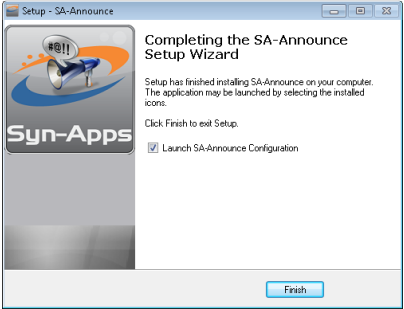
Firewall requirements/port usage:

Ports	Description
SA-Announce server to Avaya	<i>Protocol description</i>
5060	TCP – SIP – Session Initiation Protocol
SA-Announce Server to IP Phones:	<i>Protocol description</i>
80	TCP - HTTP
20480-32767	UDP - Real-Time Protocol (RTP)
20480-32767	UDP - Multicast Real-Time Protocol (RTP)
IP Phones to SA-Announce Server:	<i>Protocol description</i>
80	TCP - HTTP
20480-32767	UDP - Real-Time Protocol (RTP)
20480-32767	UDP - Multicast Real-Time Protocol (RTP)

8.3. SA-Announce Installation

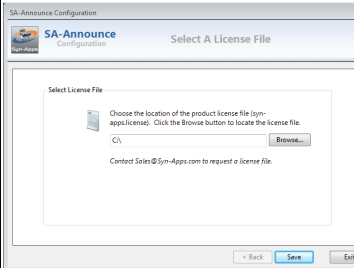
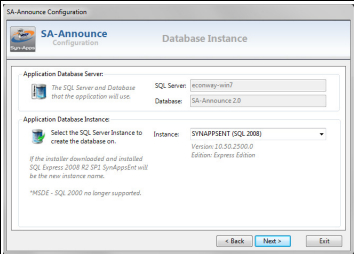
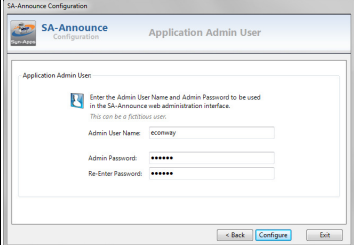
If the SA-Announce server does not have internet access, then please make sure the required components ASP.NET 4.0 and SQL Express 2008 R2 SP1 have been installed. See the SA-Announce Avaya User Guide for more information.

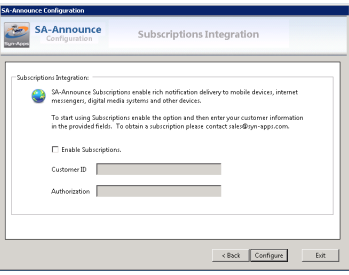
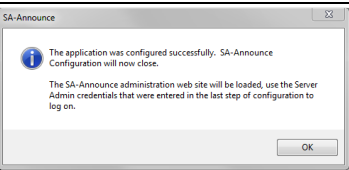
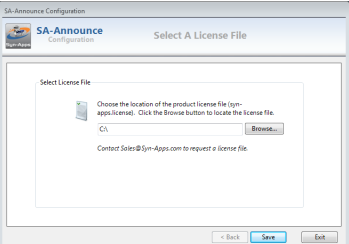
Installation Procedure	Description	Screenshot
Start the Installer	<p>Start the SA-Announce setup program. The SA-Announce setup program welcome screen should appear.</p> <p>Click Next to proceed.</p> <p>Note: Download SA-Announce here: http://www.syn-apps.com/support/downloads/</p> <p>A license file is required. Please contact sales@syn-apps.com to request one.</p>	

Installation Procedure	Description	Screenshot
Accept License Agreement	<p>The License Agreement page should appear. Read and accept the license terms by selecting clicking I accept the agreement radio button.</p> <p>Click Next to proceed.</p>	
Requirement Information	<p>Review and verify the server requirements.</p> <p>Click Next to proceed.</p> <p>*The system requirements will now be validated. Missing components will be downloaded from SA-Apps servers and installed if necessary. This could take some time depending on the system.</p>	
Ready to Install	<p>Click Install to start the product installation.</p>	
Install Complete	<p>The installation is now complete. Make sure the Launch SA-Announce Configuration checkbox is checked.</p> <p>Click Finish to begin the SA-Announce Configuration Utility.</p>	

8.4. SA-Announce Configuration Utility

The SA-Announce Configuration Utility will automatically run at the end of installation. It can also be started manually at any time from the **Start→All Programs→Syn-Apps→SA-Announce→SA-Announce Configuration** shortcut.

Config Step	Description	Screenshot
License Selection	<p>In order to use the SA-Announce software, a valid license key must be obtained. Please contact sales@syn-apps.com to obtain a license. If the license file has already obtained, click Browse to locate it.</p> <p>The license file screen will only appear if the license file, Syn-Apps.license, does not exist in the program base directory (C:\Program Files\Syn-Apps\SA-Announce\). If you encounter problems with the browser, simply place your license in the program base directory and make sure it is copied there with the precise filename Syn-Apps.license.</p> <p>Click Next to proceed.</p>	
Database Instance	<p>If SQL Server Express 2008 R2 was installed along with this installation leave the default instance name as SynAppsEnt. If an existing local SQL server instance is to be used, select that instance name from the drop-down list.</p> <p>Click Next to proceed.</p>	
Application Admin Credentials	<p>Setup the SA-Announce application administrator user account credentials. This will be the system admin user for the SA-Announce notification system. It is not a Windows or Domain account.</p> <p>Click Next to proceed.</p>	

Config Step	Description	Screenshot
Subscriptions Integration	<p>SA-Announce Subscription Services include various cloud messaging methods such as Smartphone and SMS notifications. This compliance test did not include testing of the Subscription Services. Please contact sales@syn-apps.com for more information.</p> <p>Click Configure to start the configuration process.</p>	
Configuration Complete	<p>When the application is configured you will see a success window.</p> <p>Click OK to proceed.</p>	
License Activation	<p>If the license has not been activated, the Activation Wizard will appear.</p> <p>Select Online if the SA-Announce server has access to the Internet. Click Next to activate the license.</p> <p>Select By Email if the SA-Announce server does not have access to the Internet. Click Next. Further instructions will be provided in this case.</p>	

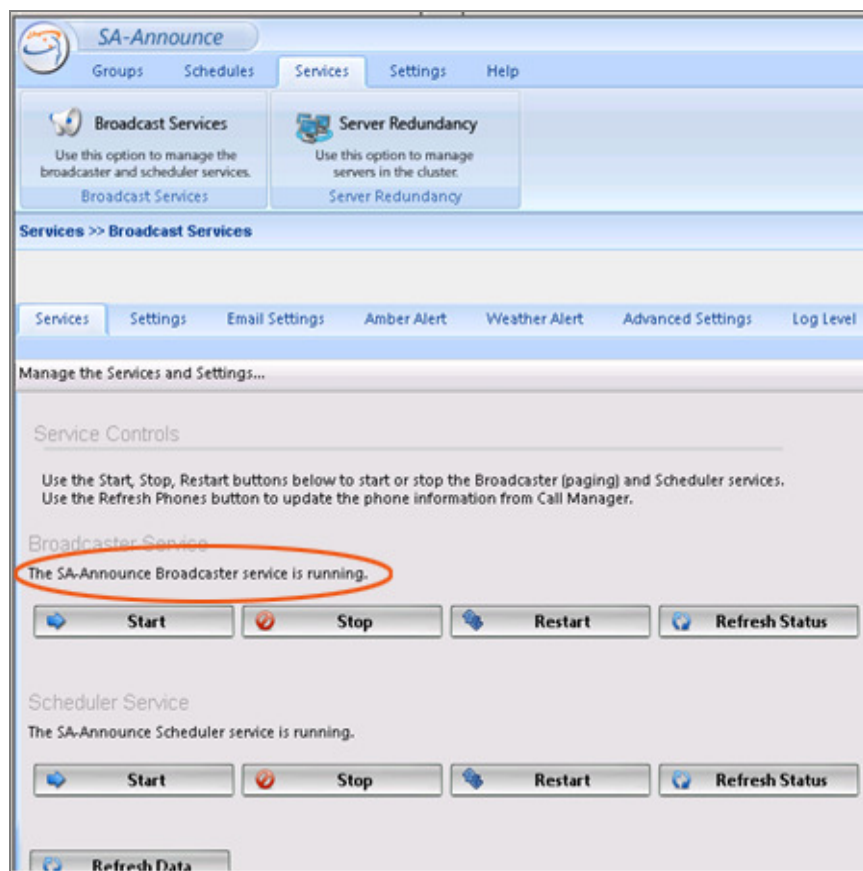
When the Configuration Wizard has completed the installation a shortcut will be placed on the desktop labeled **SA-Announce**, use this to access the SA-Announce web administration program. It is also possible to directly access the SA-Announce administration web pages from any machine on the network, by browsing to <http://<application-server-ip>/SA-Announce>, where application-server-ip address is the ip address of SA-Announce server.

The SA-Announce web page should be displayed automatically when the configuration program completes. The opening page reviews the settings necessary all SA-Announce to communicate with Avaya Communications Manager and Phones.

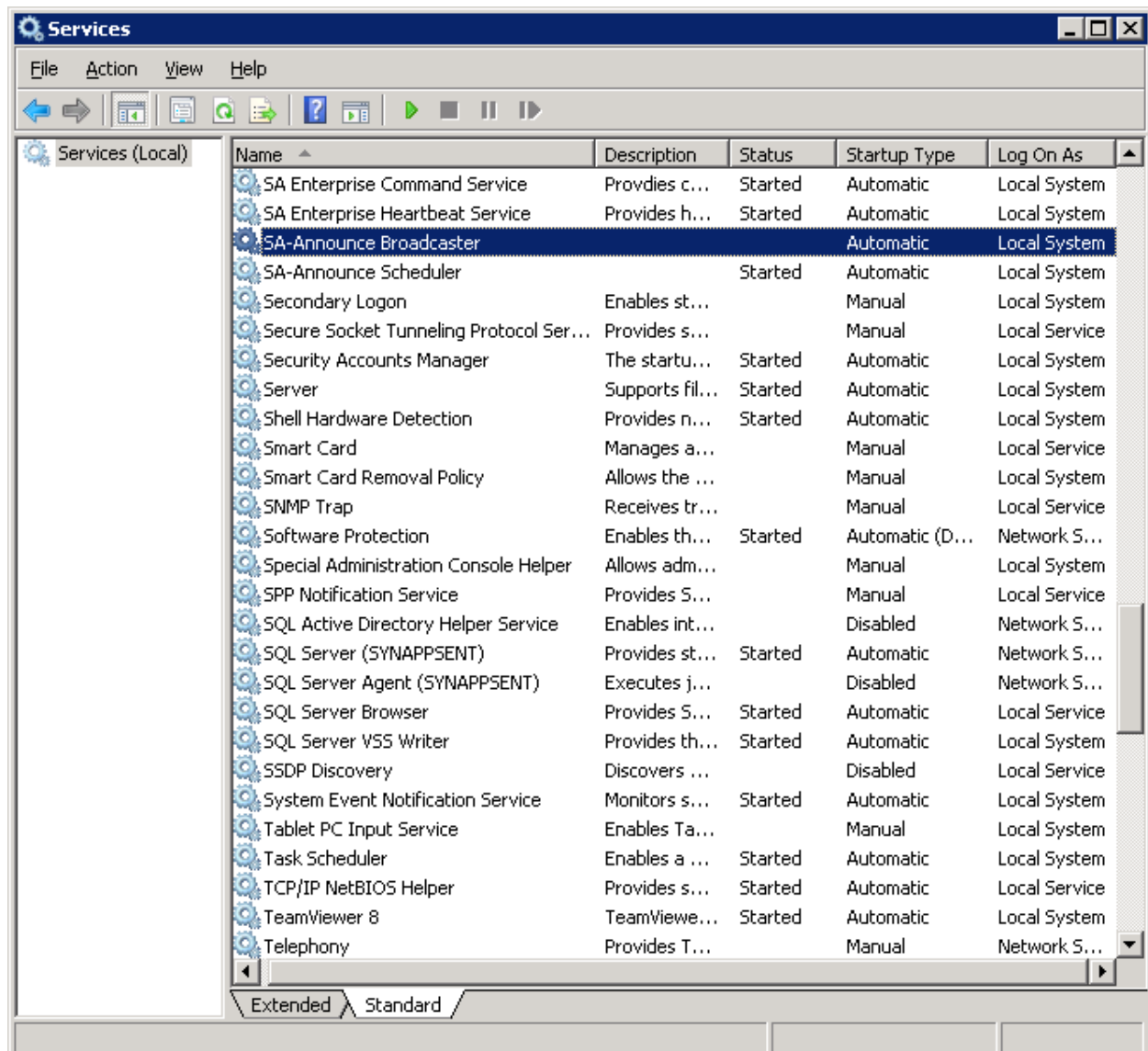
Note: When SA-Announce is installed it will automatically enable SIP traffic for port 5060 using TCP.

8.5. SA-Announce Paging Basics

Ensure the SA-Announce Broadcaster Service is running prior to testing paging. Open the **Services→Broadcast Services** page. The Services tab shows the service status and contains buttons to start, stop, and restart the service. The Broadcaster Service should be running as shown in the screen capture below.

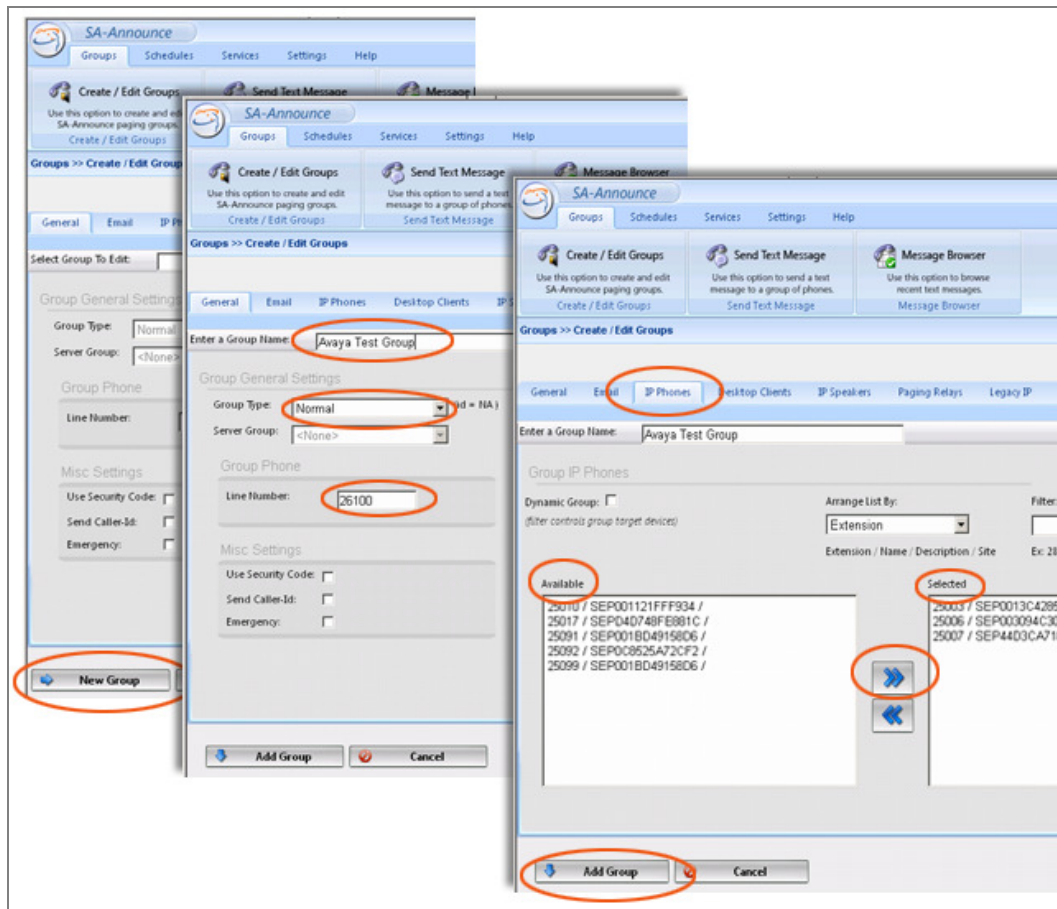


The SA-Announce services can also be controlled from the Windows Program Services Console (Services.msc). To control the SA-Announce broadcaster services, look for **SA-Announce Broadcaster** services in **Services.msc**.



To create a simple Paging Group to test, go to the **Groups→Create/Edit Groups** page.

- Click the **New Group** button at the bottom of the page
- Enter a Paging **Group name**
- Select the **Normal** as **Group Type**
- Enter a **Line Number** (Extension) for the the group
- Click the **IP Phones** tab
- Select phones from the **Available** list and move them to the Selected list by clicking the **double right arrow button**
- Click the **Add Group** button to complete the process



Once the group is saved, test the group by dialing the line number of the group (26100 in the example above). An audible tone should be heard the source phone's speaker. After hearing the tone, start speaking. The message should then be heard on the selected destination phones.

For more information on Group Types and Advanced Group configuration and testing, please reference the SA-Announce Avaya User Guide. A link to the document has been provided in Section 11.

9. Verification Steps

9.1. Avaya Aura® Session Manager

From the System Manager web page, navigate to **Session Manager → System Status → SIP Entity Monitoring**. Under the **All Monitoring SIP Entities**, select SA-Announce SIP entity that was configured in this document (not shown).

Ensure that **Conn. Status** is **UP**, and **Reason Code** is **200 OK**. This will verify that the connection between Session Manager and SA-Announce server is successful.

1 Items Refresh		Filter: Enable							
	Session Manager	SIP Entity Resolved IP	Port	Proto.	Deny	Conn. Status	Reason Code	Link Status	
<input type="radio"/>	asm-tr1	10.64.101.9	5060	TCP	FALSE	UP	200 OK	UP	

Additionally, a test call can be made to a group configured in **Section 8.5** to verify that the group is active.

10. Conclusion

Syn-apps' SA-Announce was able to successfully interoperate with Avaya 9600 Series IP Deskphones. All executed test cases were passed.

11. Additional References

Product documentation for Avaya products may be found at <http://support.avaya.com>.

- [1] Administering Avaya Aura® Communication Manager, Release 6.3, Document 03-300509, Issue 8 May 2013
- [2] Administering Avaya Aura® Session Manager, Release 6.3, June 2013

Product information for Syn-Apps SA-Announce can be found at the following URL:

[3] SA-Announce Notification System User Manual Version 8.0.0
<http://www.syn-apps.com/downloads/Install-Guides/SA-Announce%20Avaya%20User%20Guide.pdf>

Note: The document used to Install and configured SA-Announce is for Version 8.0.0. Document for Version 9.0.0 will be published at a later time. Please contact Syn-apps to obtain the latest version of the document.

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