



Avaya Solution & Interoperability Test Lab

Application Notes for InfoTalk-Speaker with Avaya Interactive Response and Avaya Communication Manager - Issue 1.0

Abstract

These Application Notes describe the configuration steps required for InfoTalk-Speaker to successfully interoperate with Avaya Interactive Response (IR) and Avaya Communication Manager. VoiceXML applications running on the Avaya IR platform utilize the text-to-speech (TTS) features of InfoTalk-Speaker using the Media Resource Control Protocol (MRCP) Version 1. Information in these Application Notes has been obtained through compliance testing and additional technical discussions. Testing was conducted via the *DeveloperConnection* Program at the Avaya Solution and Interoperability Test Lab.

1. Introduction

These Application Notes describe the compliance-tested configurations utilizing Avaya IR 1.3, Avaya Communication Manager 3.0.1 and InfoTalk-Speaker 2.0. VoiceXML applications running on the Avaya IR 1.3 platform utilize the text-to-speech (TTS) features of InfoTalk-Speaker 2.0 using the Media Resource Control Protocol (MRCP) Version 1.

InfoTalk-Speaker is a TTS solution; it brings more human-like voices to market. It speaks out computer and on-line texts into a natural, rhythmical and pleasant voice, with a quality superior to the mechanical sounds of legacy technologies.

InfoTalk-Speaker completely revolutionizes the voice application horizon. There is no longer the need to ask a human agent to create voice recordings repeatedly, which is cumbersome, unreliable and impractical. Instead, InfoTalk-Speaker automatically scans computer and on-line texts and speaks in a natural voice.

MRCP is an emerging, open standard for speech interfaces that supports the interoperability of vendor systems. MRCP addresses the communications between interactive voice response systems like the Avaya IR 1.3 and specialized speech engines such as the InfoTalk-Speaker 2.0 TTS engine. MRCP Version 1 uses the Real Time Streaming Protocol (RTSP) to establish connections from an MRCP client application to an MRCP server. All MRCP commands are then tunneled via RTSP Announce messages between the MRCP client and server. Audio data (synthesized speech delivered from a TTS engine) is then carried over a Real-time Transport Protocol (RTP) connection.

InfoTalk-Speaker is a software solution running both the InfoTalk-Speaker TTS engine and the InfoTalk MRCP Server 1.0 application on a Microsoft Windows 2000 or 2003 Server or Windows 2000 or XP Professional machine.

Figure 1 illustrates the configuration used to verify InfoTalk-Speaker 2.0 solution. The InfoTalk-Speaker 2.0 and InfoTalk MRCP Server 1.0 software were installed on a Windows XP Professional with Service Pack 2. VoiceXML scripts that used the TTS engine were installed on Avaya IR 1.3. The S8500B Media Server and G650 Media Gateway interfaced with the Avaya IR using the Voice over IP (VoIP) feature on Avaya IR. With VoIP, transmission to the switch is achieved without digital interfaces (T1/E1). Instead, all transmissions occur over the packet network using the network interface card (NIC) on the Avaya IR system. Avaya IP phones were used to place calls to the Avaya IR to run the VoiceXML scripts. The scripts would use the TTS engine to play synthesized prompts and verify DMTF presses and barge-in attempts.

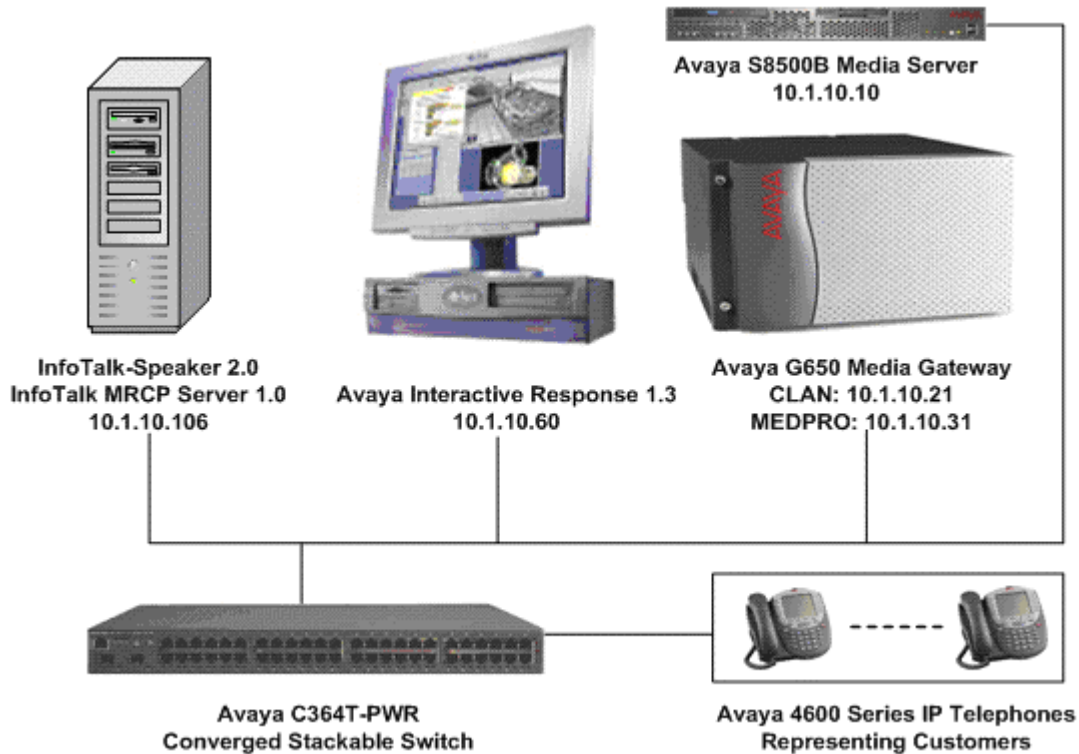


Figure 1: InfoTalk-Speaker 2.0 with Avaya IR 1.3 Configuration

2. Equipment and Software Validated

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

| Equipment | Software |
|---|--|
| Avaya Interactive Response on SunBlade 150 | 1.3 |
| Avaya S8500B Media Server | R013x.00.1.346.0 (3.0.1) |
| Avaya G650 Media Gateway <ul style="list-style-type: none"> • TN2312BP IP Server Interface • TN799DP C-LAN Interface • TN2302AP IP Media Processor | - HW07, FW022 HW01, FW016 HW20, FW107 |
| Avaya 4600 Series IP Telephones | 2.3 (4610SW) 2.3 (4620SW) 2.3 (4621SW) 2.5 (4625SW) |
| Avaya IR Designer | 5.2 |
| Avaya Dialog Designer | 3.0.18 |
| InfoTalk-Speaker | 2.0 |
| InfoTalk MRCP Server | 1.0 |
| Generic PC (Pentium 4, 2.8 GHz, 1 GB RAM) | Windows XP Professional with Service Pack 2 |

3. Configure Avaya Communication Manager

This section presents the configuration required on Avaya Communication Manager to interface with Avaya IR. This covers the following capabilities:

- Inbound calls are routed to **VDN 7910** that invokes **Vector 2**.
- **Vector 2** queues the incoming call to **Hunt Group 2** with IVR ports.
- IVR ports are configured as **Stations** (79101-79124).
- Stations associated with IVR ports automatically log into the hunt group via **Agent LoginIDs** (29101-29024).

The following configuration is performed via the System Access Terminal (SAT) on Avaya Communication Manager. It is assumed that the Avaya Media Server has already been assigned an IP address.

| Step | Description |
|------|--|
| 1. | <p>Use the display system-parameters customer-options command to check that Avaya Communication Manager has the feature license enabled for Avaya IR connectivity.</p> <p>On page 10, verify that the Limit field for IP_IR_A has a value greater than or equal to the maximum number of channels configured on Avaya IR.</p> <pre> display system-parameters customer-options Page 10 of 11 MAXIMUM IP REGISTRATIONS BY PRODUCT ID Product ID Rel. Limit Used IP_API_A : 500 0 IP_API_B : 0 0 IP_API_C : 0 0 IP_Agent : 100 0 IP_IR_A : 100 30 IP_Phone : 2400 4 IP_ROMax : 2400 0 IP_Soft : 100 0 IP_eCons : 2 0 : 0 0 : 0 0 : 0 0 : 0 0 : 0 0 : 0 0 </pre> |

| Step | Description |
|------|---|
| 2. | <p>Use the add station n command, where n is a valid extension, to configure each IVR port as a station with the Type field set to H.323. Specify the Security Code, which will be used in Section 4 Step 6 when configuring the phone numbers on IR. Repeat this configuration for each IVR port. In this configuration, 30 IVR ports were configured with an extension range of 10101 to 10130. These stations will be members of Hunt Group 110 (configured in Step 3) and will automatically log into the split via the Agent LoginIDs (configured in Step 4).</p> <pre> add station 10101 Page 1 of 3 STATION Extension: 10101 Lock Messages? n BCC: 0 Type: H.323 Security Code: 10101 TN: 1 Port: IP Coverage Path 1: COR: 1 Name: IR #1 Coverage Path 2: COS: 1 Hunt-to Station: Tests? y STATION OPTIONS Loss Group: 19 Message Waiting Indicator: none Survivable COR: internal SurvivableDTMF over IP: in-band IP Video? n </pre> |

| Step | Description |
|------|--|
| 3. | <p data-bbox="358 233 1464 373">Enter the add hunt-group n command, where n is an unused hunt group number. The IVR ports, configured as H.323 stations, will automatically log into the hunt group. Set the Group Extension field to a valid extension and set ACD, Queue and Vector to y.</p> <pre data-bbox="358 380 1464 785"> add hunt-group 110 Page 1 of 3 HUNT GROUP Group Number: 110 ACD? y Group Name: IR Skill Queue? y Group Extension: 13110 Vector? y Group Type: ead-mia TN: 1 COR: 1 MM Early Answer? n Security Code: Local Agent Preference? n ISDN/SIP Caller Display: Queue Limit: unlimited Calls Warning Threshold: Port: Time Warning Threshold: Port: </pre> <p data-bbox="358 827 1464 898">On Page 2 of the Hunt Group form, set Skill and AAS to y. The AAS option will allow the IVR ports to automatically log into the hunt group via the Agent LoginIDs.</p> <pre data-bbox="358 940 1464 1402"> add hunt-group 110 Page 2 of 3 HUNT GROUP Skill? y Expected Call Handling Time (sec): 180 AAS? y Measured: none Supervisor Extension: Controlling Adjunct: none Timed ACW Interval (sec): Redirect on No Answer (rings): Redirect to VDN: Forced Entry of Stroke Counts or Call Work Codes? n </pre> |

| Step | Description |
|------|--|
| 4. | <p>Use the add agent-loginID n command, where n is a valid extension, to add an agent. Add an Agent LoginID for each IVR port. Set AAS to y and the Port Extension to the corresponding extension of the stations for each IVR port.</p> <pre> add agent-loginID 11101 Page 1 of 2 AGENT LOGINID Login ID: 11101 AAS? y Name: IR #1 AUDIX? n TN: 1 LWC Reception: spe COR: 1 LWC Log External Calls? n Coverage Path: AUDIX Name for Messaging: Security Code: LoginID for ISDN Display? n Port Extension: 10101 Auto Answer: station MIA Across Skills: system ACW Agent Considered Idle: system Aux Work Reason Code Type: system Logout Reason Code Type: system Maximum time agent in ACW before logout (sec): system WARNING: Agent must log in again before skill changes take effect </pre> <p>On Page 2 of the form, set the skill number (SN) to the hunt group configured in Step 3 and the skill level (SL) to I. Repeat this step for each station configured in Step 2. In this configuration, agent login IDs 11101 to 11130 were created.</p> <pre> add agent-loginID 11101 Page 2 of 2 AGENT LOGINID Direct Agent Skill: Call Handling Preference: skill-level Local Call Preference? n SN SL SN SL SN SL SN SL 1: 110 1 16: SL 31: SL 46: SL 2: 17: 32: SL 3: 18: 33: SL 4: 19: 34: SL 5: 20: 35: SL 6: 21: 36: SL 7: 22: 37: SL 8: 23: 38: SL 9: 24: 39: SL 10: 25: 40: SL 11: 26: 41: SL 12: 27: 42: SL 13: 28: 43: SL 14: 29: 44: SL 15: 30: 45: SL </pre> |


| Step | Description |
|------|--|
| 5. | <p>Use the add vdn n command, where n is a valid number, to create the Vector Directory Number (VDN) that will handle all incoming calls to the Avaya IR. Specify an unused Vector for Vector Number.</p> <pre> add vdn 14110 VECTOR DIRECTORY NUMBER Page 1 of 3 Extension: 14110 Name: Queue to IR Vector Number: 110 Meet-me Conferencing? n Allow VDN Override? n COR: 1 TN: 1 Measured: none VDN of Origin Annc. Extension: 1st Skill: 2nd Skill: 3rd Skill: </pre> |
| 6. | <p>Use the change vector n command, where n is the vector number specified in Step 5, to configure the vector. VDN 14110, configured above, will invoke vector 110 which will queue the call to the IVR hunt group via the queue-to skill step. Configure vector 110 as shown below.</p> <pre> change vector 110 CALL VECTOR Page 1 of 3 Number: 110 Name: Q2 IR Meet-me Conf? n Lock? n Basic? y EAS? y G3V4 Enhanced? y ANI/II-Digits? y ASAI Routing? y Prompting? y LAI? y G3V4 Adv Route? y CINFO? y BSR? y Holidays? y Variables? y 3.0 Enhanced? y 01 wait-time 0 secs hearing silence 02 queue-to skill 110 pri m 03 wait-time 30 secs hearing music 04 disconnect after announcement none 05 </pre> |

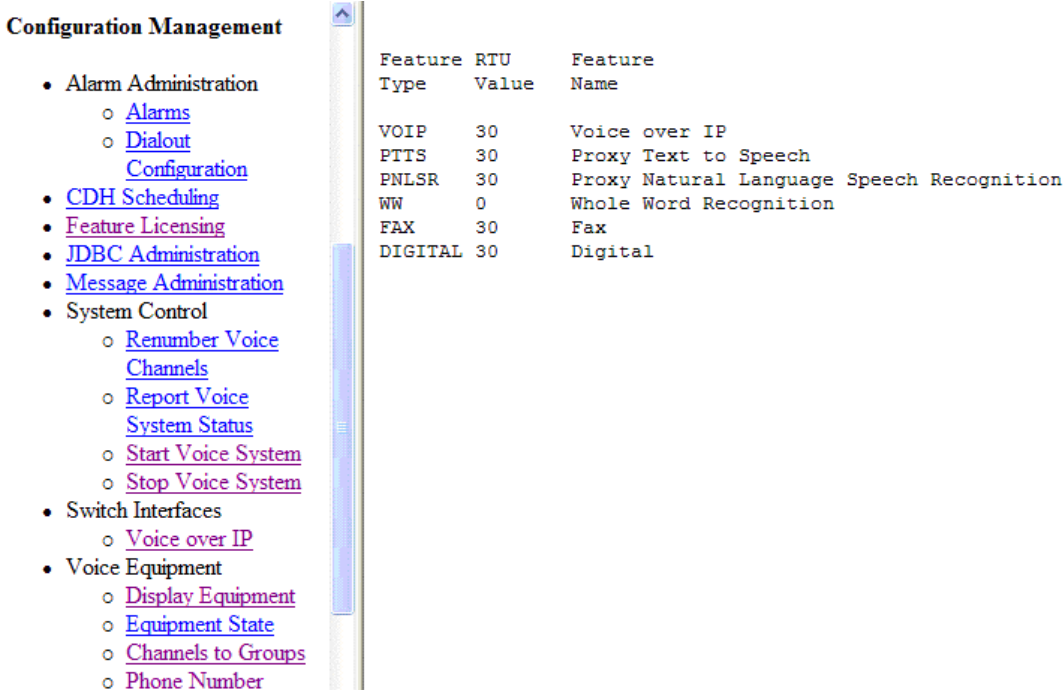
4. Configure Avaya Interactive Response (IR)


This section covers the configuration of Avaya IR. Avaya Communication Manager routes incoming calls to Avaya IR using Voice over IP (VoIP) over the data network. Each VoIP channel is assigned a VXML application and a phone number that matches a corresponding extension configured on Avaya Communication Manager in Section 3 Step 2 above. The VXML applications were developed using Avaya IR Designer and Avaya Dialog Designer on a Microsoft Windows XP Professional PC and then transferred to Avaya IR. The configuration steps required on Avaya IR are summarized below.

- Access Avaya IR via a web browser and log in.
- Stop the Avaya IR Voice System prior to configuring the VoIP interface.
- Configure the VoIP interface to Avaya Communication Manager.
- Start the Voice System.
- Assign phone numbers to channels.
- Assign services (VXML applications) to channels.
- Administer and assign InfoTalk-Speaker Text-to-Speech engine.

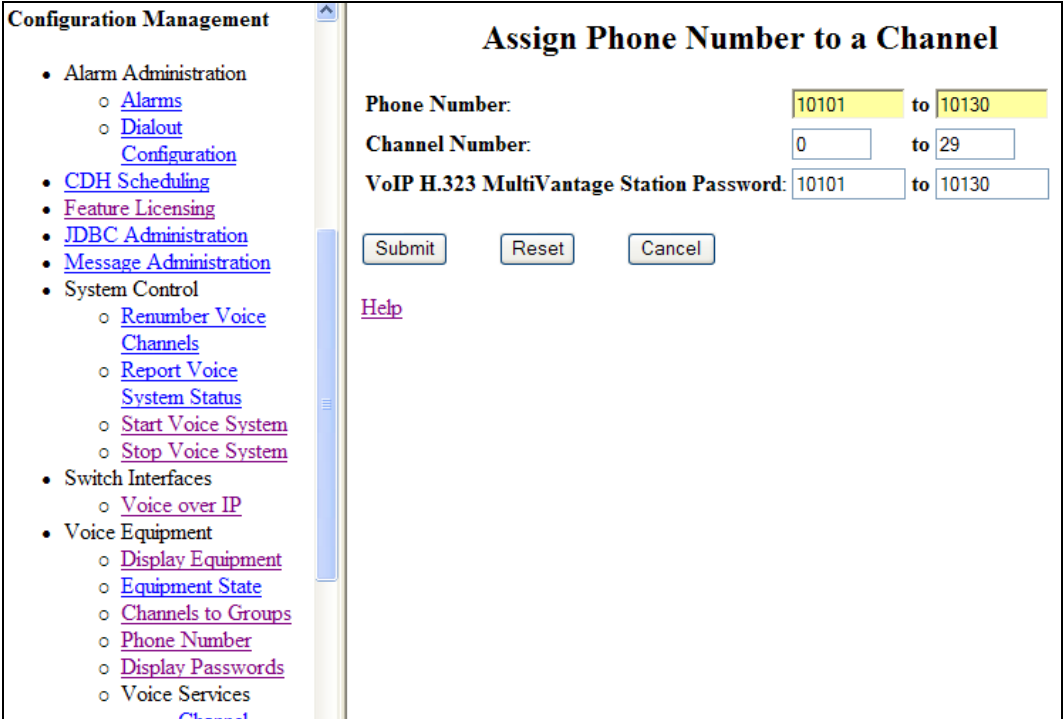
| Step | Description |
|------|---|
| 1. | <p>The following packages were installed on Avaya IR to support MRCP Text to Speech.</p> <ul style="list-style-type: none"> • Speech Proxy package (AVsproxy) • Proxy Text-to-Speech Package (AVttsprxy) • MRCP Text-to-Speech package (AVmrcptts) <p>To verify the installed packages, run “pkginfo grep AV” command from Avaya IR’s command prompt.</p> <pre> irl(root)# pkginfo grep AV IVR AVbackrst Backup/Restore Utilities IVR AVftst Feature Test Script Package IVR AVir Interactive Response Base System IVR AVjdbcint JDBC Integration IVR AVmigr Migration Tools IVR AVmrcptts MRCP TTS Proxy IVR AVsc Service Creation Integration Package Release 5.2 IVR AVsproxy Speech Proxy Base Software IVR AVtscrtu License Modification Package IVR AVtsm Transaction State Machine IVR AVttsprxy Proxy Text-to-Speech Package IVR AVucid Universal Call ID IVR AVval Avaya IR System Validation Package IVR AVvoicxml Voice XML Interpreter IVR AVvoip Voice Over IP IVR AVwebadm Web Administration IVR AVxfer Call Transfer and Bridge Package </pre> |

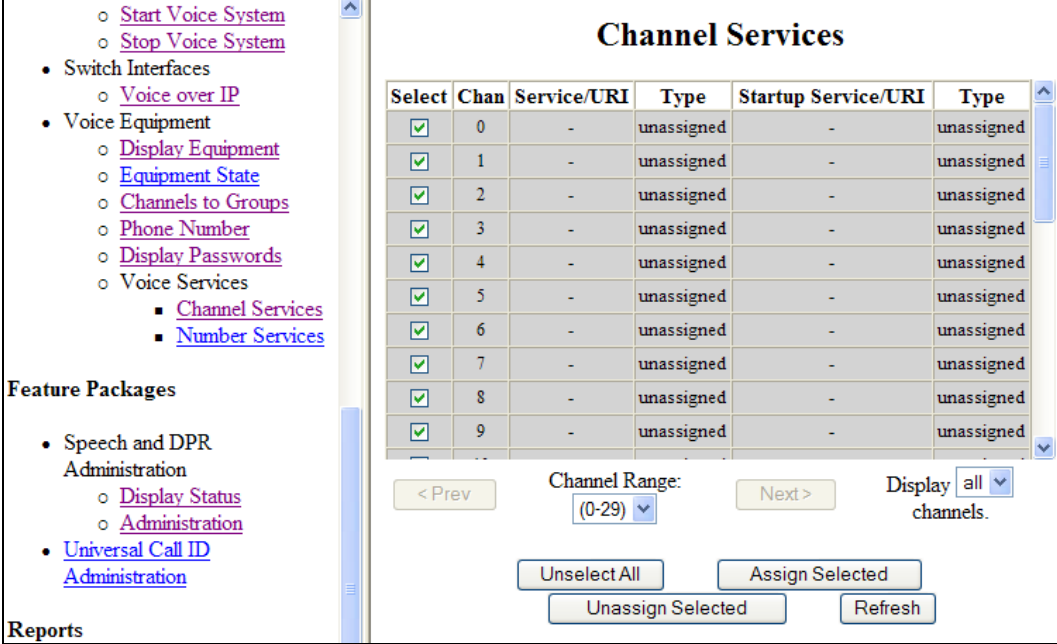
| Step | Description |
|------|---|
| 2. | <p>Avaya IR configuration was performed via a web browser. Enter the IP address of Avaya IR in the URL field of the web browser. The initial Avaya IR web page is displayed. Click Web Administration to display the login screen shown below, and log into Avaya IR as “root”.</p> <div data-bbox="688 436 1154 575" style="text-align: center;"></div> <p data-bbox="492 659 1317 806" style="text-align: center;"><i>WARNING: This system is restricted to authorized users for business purposes. Unauthorized access is a violation of the law. This system may be monitored for administrative and security reasons. By proceeding, you consent to this monitoring.</i></p> <div data-bbox="678 890 1166 1083" style="text-align: center;"><p>Username: <input data-bbox="813 890 1166 932" type="text" value="root"/></p><p>Password: <input data-bbox="813 940 1166 989" type="password" value="●●●●●●●●"/></p><p><input data-bbox="875 1037 971 1083" type="button" value="Login"/></p></div> |

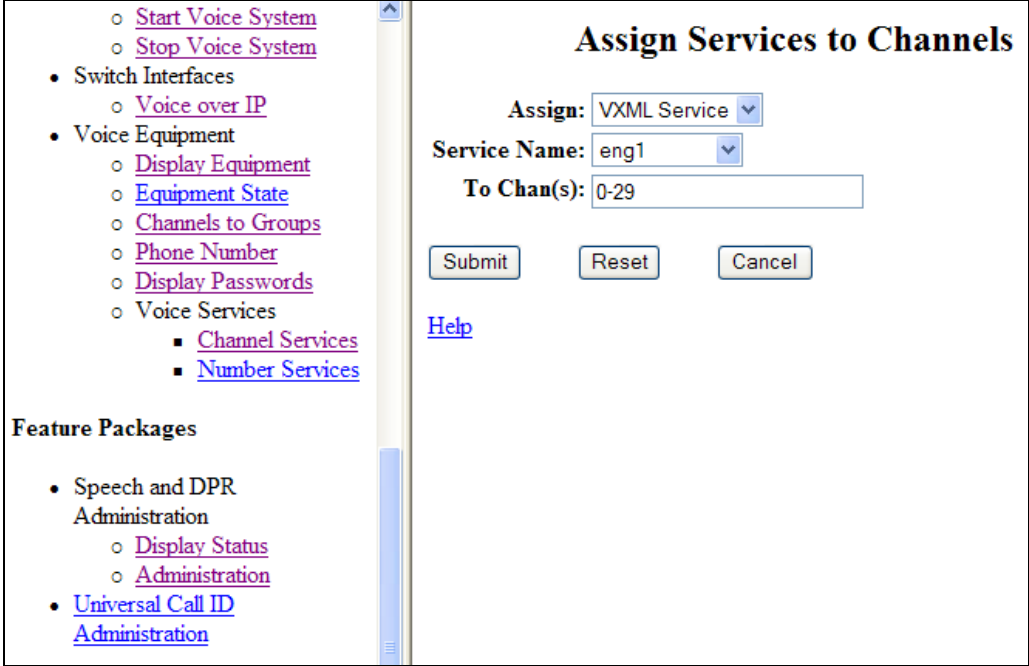
| Step | Description | | | | | | | | | | | | | | | | | | | | | |
|--------------|--|---|-----------|--------------|------|----|---------------|------|----|----------------------|-------|----|---|----|---|------------------------|-----|----|-----|---------|----|---------|
| 3. | <p>After successfully logging into Avaya IR, the main Avaya IR configuration web page is displayed. Click Feature Licensing to display the Feature License page. Verify that the Right-To-Use (RTU) Value for the Feature Type VOIP is set to the number of VOIP channels used and PTTS is set to an appropriate value to support the number of channels.</p>  <p>The screenshot shows the 'Configuration Management' page with a left-hand navigation menu and a table of feature licenses. The navigation menu includes: Alarm Administration (Alarms, Dialout, Configuration), CDH Scheduling, Feature Licensing, JDBC Administration, Message Administration, System Control (Renumber Voice Channels, Report Voice System Status, Start Voice System, Stop Voice System), Switch Interfaces (Voice over IP), and Voice Equipment (Display Equipment, Equipment State, Channels to Groups, Phone Number). The table on the right lists features with their RTU values: VOIP (30), PTTS (30), PNLSR (30), WW (0), FAX (30), and DIGITAL (30).</p> <table border="1" data-bbox="738 472 1421 693"> <thead> <tr> <th>Feature Type</th> <th>RTU Value</th> <th>Feature Name</th> </tr> </thead> <tbody> <tr> <td>VOIP</td> <td>30</td> <td>Voice over IP</td> </tr> <tr> <td>PTTS</td> <td>30</td> <td>Proxy Text to Speech</td> </tr> <tr> <td>PNLSR</td> <td>30</td> <td>Proxy Natural Language Speech Recognition</td> </tr> <tr> <td>WW</td> <td>0</td> <td>Whole Word Recognition</td> </tr> <tr> <td>FAX</td> <td>30</td> <td>Fax</td> </tr> <tr> <td>DIGITAL</td> <td>30</td> <td>Digital</td> </tr> </tbody> </table> | Feature Type | RTU Value | Feature Name | VOIP | 30 | Voice over IP | PTTS | 30 | Proxy Text to Speech | PNLSR | 30 | Proxy Natural Language Speech Recognition | WW | 0 | Whole Word Recognition | FAX | 30 | Fax | DIGITAL | 30 | Digital |
| Feature Type | RTU Value | Feature Name | | | | | | | | | | | | | | | | | | | | |
| VOIP | 30 | Voice over IP | | | | | | | | | | | | | | | | | | | | |
| PTTS | 30 | Proxy Text to Speech | | | | | | | | | | | | | | | | | | | | |
| PNLSR | 30 | Proxy Natural Language Speech Recognition | | | | | | | | | | | | | | | | | | | | |
| WW | 0 | Whole Word Recognition | | | | | | | | | | | | | | | | | | | | |
| FAX | 30 | Fax | | | | | | | | | | | | | | | | | | | | |
| DIGITAL | 30 | Digital | | | | | | | | | | | | | | | | | | | | |

| Step | Description |
|------|---|
| 4. | <p>Click Stop Voice System to stop the Voice System so that the VoIP interface can be configured. When the Stop Voice System page is displayed, click Submit and wait until the system displays a message at the bottom of the page indicating that the Voice System has completely stopped.</p>  |

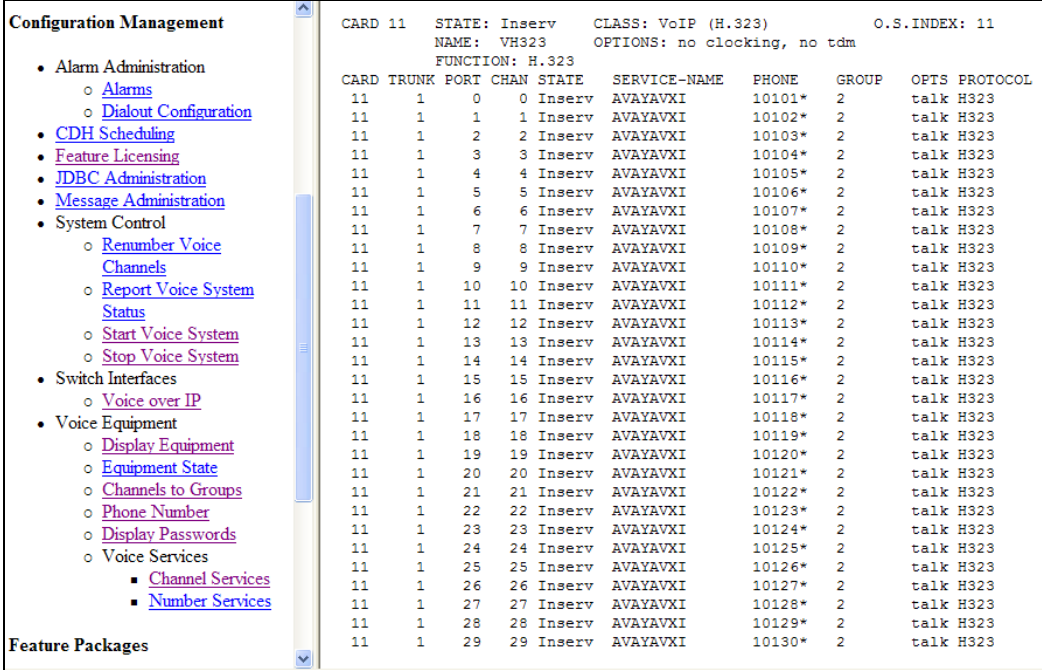
| Step | Description |
|------|---|
| 5. | <p>To configure the VoIP interface to Avaya Communication Manager, follow these steps:</p> <ol style="list-style-type: none"> Under Switch Interfaces in the left pane, click Voice over IP to display the Voice over IP page. Click Assign Card to display the Assign VoIP Card page. Set Card IP Address to the IP address of the NIC card on IR used for VoIP, Gatekeeper IP Address to the IP address of the CLAN board on Avaya Communication Manager and Station Authentication Enabled to <i>yes</i>. Click Submit. <div data-bbox="391 630 1404 1449" style="border: 1px solid black; padding: 10px;"> <p>Configuration Management</p> <ul style="list-style-type: none"> • Alarm Administration <ul style="list-style-type: none"> ○ Alarms ○ Dialout Configuration • CDH Scheduling • Feature Licensing • JDBC Administration • Message Administration • System Control <ul style="list-style-type: none"> ○ ReNUMBER Voice Channels ○ Report Voice System Status ○ Start Voice System ○ Stop Voice System • Switch Interfaces <ul style="list-style-type: none"> ○ Voice over IP • Voice Equipment <ul style="list-style-type: none"> ○ Display Equipment ○ Equipment State ○ Channels to Groups ○ Phone Number ○ Display Passwords ○ Voice Services <p>Assign VoIP Card</p> <p>Card: <input type="text" value="11"/></p> <p>Card Name: <input type="text" value="VH323"/></p> <p>Card Enabled?: <input type="text" value="yes"/></p> <p>Card IP Address: <input type="text" value="10.1.10.60"/></p> <p>Gatekeeper IP Address: <input type="text" value="10.1.10.21"/></p> <p>H.323 Gatekeeper Port: <input type="text" value="1719"/></p> <p>Low RTP Port: <input type="text" value="8000"/></p> <p>High RTP Port: <input type="text" value="10000"/></p> <p>RTP Packet Size: <input type="text" value="50"/></p> <p>RTCP Monitor Enabled?: <input type="text" value="no"/></p> <p>RTCP Monitor IP Address: <input type="text" value="127.0.0.0"/></p> <p>RTCP Monitor Port: <input type="text" value="5005"/></p> <p>Station Authentication Enabled?: <input type="text" value="yes"/></p> <p><input type="button" value="Submit"/> <input type="button" value="Reset"/> <input type="button" value="Cancel"/></p> <p>Help</p> </div> <p>After the VoIP card is successfully configured, start the Voice System by clicking on Start Voice System. When the Start Voice System page is displayed, click Submit and wait until the system displays a message at the bottom of the page indicating that the startup of the Voice System is complete.</p> |

| Step | Description |
|------|---|
| 6. | <p>To assign phone numbers to channels, click Phone Number to display the Phone Number - Channel Assignment page and click Assign. On the Assign Phone Number to a Channel page, set Phone Number to <i>10101 to 10130</i>, Channel Number to <i>0 to 29</i>, VoIP H.323 MultiVantage Station Password to <i>10101 to 10130</i> and click Submit. The Phone Number and VoIP H.323 MultiVantage Station Password fields must match the stations created in Section 3 Step 2.</p>  |

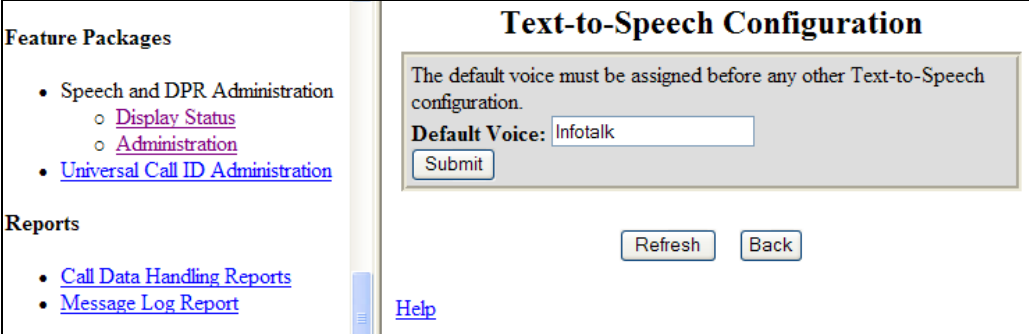
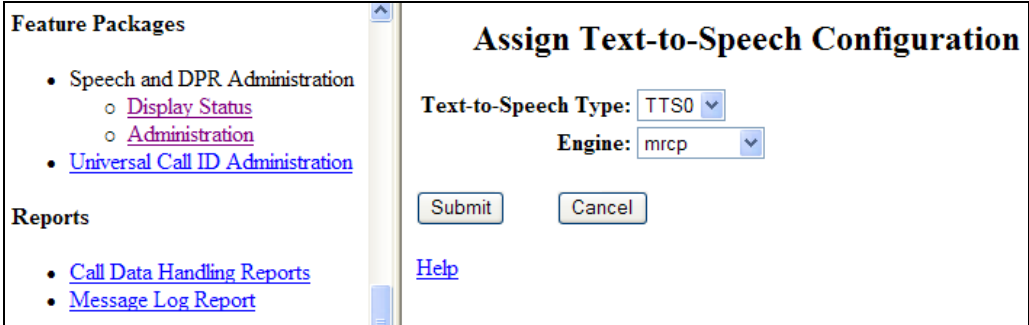
| Step | Description | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
|-------------------------------------|---|-------------|------------|---------------------|------------|---------------------|------|-------------------------------------|---|---|------------|---|------------|-------------------------------------|---|---|------------|---|------------|-------------------------------------|---|---|------------|---|------------|-------------------------------------|---|---|------------|---|------------|-------------------------------------|---|---|------------|---|------------|-------------------------------------|---|---|------------|---|------------|-------------------------------------|---|---|------------|---|------------|-------------------------------------|---|---|------------|---|------------|-------------------------------------|---|---|------------|---|------------|-------------------------------------|---|---|------------|---|------------|
| 7. | <p>Assign a VXML application to each channel. This specifies which VXML application a particular IVR channel should run when it receives a call. Click Channel Services to display the Channel Services page. Click the drop-down beside Display and select all so as to display all channels. Click Select All and click Assign Selected. Thirty channels are used in this configuration because InfoTalk-Speaker 2.0 is configured with a 30-ports license.</p>  <p>The screenshot shows the 'Channel Services' configuration page. On the left is a navigation tree with categories like 'Start Voice System', 'Switch Interfaces', 'Voice Equipment', 'Voice Services', 'Feature Packages', and 'Reports'. The 'Channel Services' link is selected. The main area displays a table with the following data:</p> <table border="1" data-bbox="743 569 1386 940"> <thead> <tr> <th>Select</th> <th>Chan</th> <th>Service/URI</th> <th>Type</th> <th>Startup Service/URI</th> <th>Type</th> </tr> </thead> <tbody> <tr><td><input checked="" type="checkbox"/></td><td>0</td><td>-</td><td>unassigned</td><td>-</td><td>unassigned</td></tr> <tr><td><input checked="" type="checkbox"/></td><td>1</td><td>-</td><td>unassigned</td><td>-</td><td>unassigned</td></tr> <tr><td><input checked="" type="checkbox"/></td><td>2</td><td>-</td><td>unassigned</td><td>-</td><td>unassigned</td></tr> <tr><td><input checked="" type="checkbox"/></td><td>3</td><td>-</td><td>unassigned</td><td>-</td><td>unassigned</td></tr> <tr><td><input checked="" type="checkbox"/></td><td>4</td><td>-</td><td>unassigned</td><td>-</td><td>unassigned</td></tr> <tr><td><input checked="" type="checkbox"/></td><td>5</td><td>-</td><td>unassigned</td><td>-</td><td>unassigned</td></tr> <tr><td><input checked="" type="checkbox"/></td><td>6</td><td>-</td><td>unassigned</td><td>-</td><td>unassigned</td></tr> <tr><td><input checked="" type="checkbox"/></td><td>7</td><td>-</td><td>unassigned</td><td>-</td><td>unassigned</td></tr> <tr><td><input checked="" type="checkbox"/></td><td>8</td><td>-</td><td>unassigned</td><td>-</td><td>unassigned</td></tr> <tr><td><input checked="" type="checkbox"/></td><td>9</td><td>-</td><td>unassigned</td><td>-</td><td>unassigned</td></tr> </tbody> </table> <p>Below the table, the 'Channel Range' is set to '(0-29)', 'Display' is set to 'all channels', and buttons for 'Unselect All', 'Assign Selected', 'Unassign Selected', and 'Refresh' are visible.</p> | Select | Chan | Service/URI | Type | Startup Service/URI | Type | <input checked="" type="checkbox"/> | 0 | - | unassigned | - | unassigned | <input checked="" type="checkbox"/> | 1 | - | unassigned | - | unassigned | <input checked="" type="checkbox"/> | 2 | - | unassigned | - | unassigned | <input checked="" type="checkbox"/> | 3 | - | unassigned | - | unassigned | <input checked="" type="checkbox"/> | 4 | - | unassigned | - | unassigned | <input checked="" type="checkbox"/> | 5 | - | unassigned | - | unassigned | <input checked="" type="checkbox"/> | 6 | - | unassigned | - | unassigned | <input checked="" type="checkbox"/> | 7 | - | unassigned | - | unassigned | <input checked="" type="checkbox"/> | 8 | - | unassigned | - | unassigned | <input checked="" type="checkbox"/> | 9 | - | unassigned | - | unassigned |
| Select | Chan | Service/URI | Type | Startup Service/URI | Type | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| <input checked="" type="checkbox"/> | 0 | - | unassigned | - | unassigned | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| <input checked="" type="checkbox"/> | 1 | - | unassigned | - | unassigned | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| <input checked="" type="checkbox"/> | 2 | - | unassigned | - | unassigned | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| <input checked="" type="checkbox"/> | 3 | - | unassigned | - | unassigned | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| <input checked="" type="checkbox"/> | 4 | - | unassigned | - | unassigned | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| <input checked="" type="checkbox"/> | 5 | - | unassigned | - | unassigned | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| <input checked="" type="checkbox"/> | 6 | - | unassigned | - | unassigned | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| <input checked="" type="checkbox"/> | 7 | - | unassigned | - | unassigned | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| <input checked="" type="checkbox"/> | 8 | - | unassigned | - | unassigned | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| <input checked="" type="checkbox"/> | 9 | - | unassigned | - | unassigned | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |

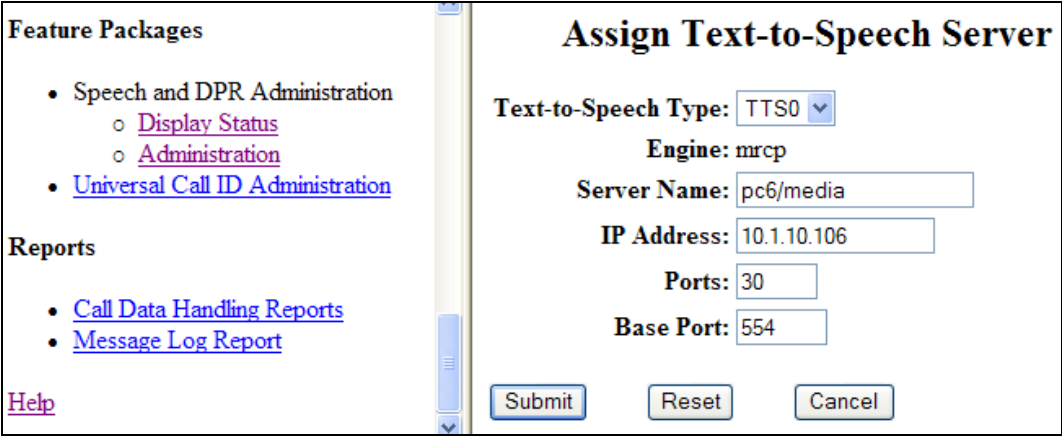
| Step | Description |
|------|---|
| | <p>On the Assign Services to Channels page that appears, select VXML Service for Assign and select the VXML application to be assigned for Service Name. Select Submit.</p>  |

8. To view the status of the channels and the configuration details, select **Display Equipment** from the left pane. Verify that the **STATE** for each channel is **Inserv**.



| Configuration Management | | CARD 11 STATE: Inserv CLASS: VoIP (H.323) O.S.INDEX: 11 | | | | | | | | | |
|--------------------------|-------|---|------|--------|--------------|--------|-------|------|----------|--|--|
| | | NAME: VH323 OPTIONS: no clocking, no tdm | | | | | | | | | |
| | | FUNCTION: H.323 | | | | | | | | | |
| CARD | TRUNK | PORT | CHAN | STATE | SERVICE-NAME | PHONE | GROUP | OPTS | PROTOCOL | | |
| 11 | 1 | 0 | 0 | Inserv | AVAYAVXI | 10101* | 2 | talk | H323 | | |
| 11 | 1 | 1 | 1 | Inserv | AVAYAVXI | 10102* | 2 | talk | H323 | | |
| 11 | 1 | 2 | 2 | Inserv | AVAYAVXI | 10103* | 2 | talk | H323 | | |
| 11 | 1 | 3 | 3 | Inserv | AVAYAVXI | 10104* | 2 | talk | H323 | | |
| 11 | 1 | 4 | 4 | Inserv | AVAYAVXI | 10105* | 2 | talk | H323 | | |
| 11 | 1 | 5 | 5 | Inserv | AVAYAVXI | 10106* | 2 | talk | H323 | | |
| 11 | 1 | 6 | 6 | Inserv | AVAYAVXI | 10107* | 2 | talk | H323 | | |
| 11 | 1 | 7 | 7 | Inserv | AVAYAVXI | 10108* | 2 | talk | H323 | | |
| 11 | 1 | 8 | 8 | Inserv | AVAYAVXI | 10109* | 2 | talk | H323 | | |
| 11 | 1 | 9 | 9 | Inserv | AVAYAVXI | 10110* | 2 | talk | H323 | | |
| 11 | 1 | 10 | 10 | Inserv | AVAYAVXI | 10111* | 2 | talk | H323 | | |
| 11 | 1 | 11 | 11 | Inserv | AVAYAVXI | 10112* | 2 | talk | H323 | | |
| 11 | 1 | 12 | 12 | Inserv | AVAYAVXI | 10113* | 2 | talk | H323 | | |
| 11 | 1 | 13 | 13 | Inserv | AVAYAVXI | 10114* | 2 | talk | H323 | | |
| 11 | 1 | 14 | 14 | Inserv | AVAYAVXI | 10115* | 2 | talk | H323 | | |
| 11 | 1 | 15 | 15 | Inserv | AVAYAVXI | 10116* | 2 | talk | H323 | | |
| 11 | 1 | 16 | 16 | Inserv | AVAYAVXI | 10117* | 2 | talk | H323 | | |
| 11 | 1 | 17 | 17 | Inserv | AVAYAVXI | 10118* | 2 | talk | H323 | | |
| 11 | 1 | 18 | 18 | Inserv | AVAYAVXI | 10119* | 2 | talk | H323 | | |
| 11 | 1 | 19 | 19 | Inserv | AVAYAVXI | 10120* | 2 | talk | H323 | | |
| 11 | 1 | 20 | 20 | Inserv | AVAYAVXI | 10121* | 2 | talk | H323 | | |
| 11 | 1 | 21 | 21 | Inserv | AVAYAVXI | 10122* | 2 | talk | H323 | | |
| 11 | 1 | 22 | 22 | Inserv | AVAYAVXI | 10123* | 2 | talk | H323 | | |
| 11 | 1 | 23 | 23 | Inserv | AVAYAVXI | 10124* | 2 | talk | H323 | | |
| 11 | 1 | 24 | 24 | Inserv | AVAYAVXI | 10125* | 2 | talk | H323 | | |
| 11 | 1 | 25 | 25 | Inserv | AVAYAVXI | 10126* | 2 | talk | H323 | | |
| 11 | 1 | 26 | 26 | Inserv | AVAYAVXI | 10127* | 2 | talk | H323 | | |
| 11 | 1 | 27 | 27 | Inserv | AVAYAVXI | 10128* | 2 | talk | H323 | | |
| 11 | 1 | 28 | 28 | Inserv | AVAYAVXI | 10129* | 2 | talk | H323 | | |
| 11 | 1 | 29 | 29 | Inserv | AVAYAVXI | 10130* | 2 | talk | H323 | | |

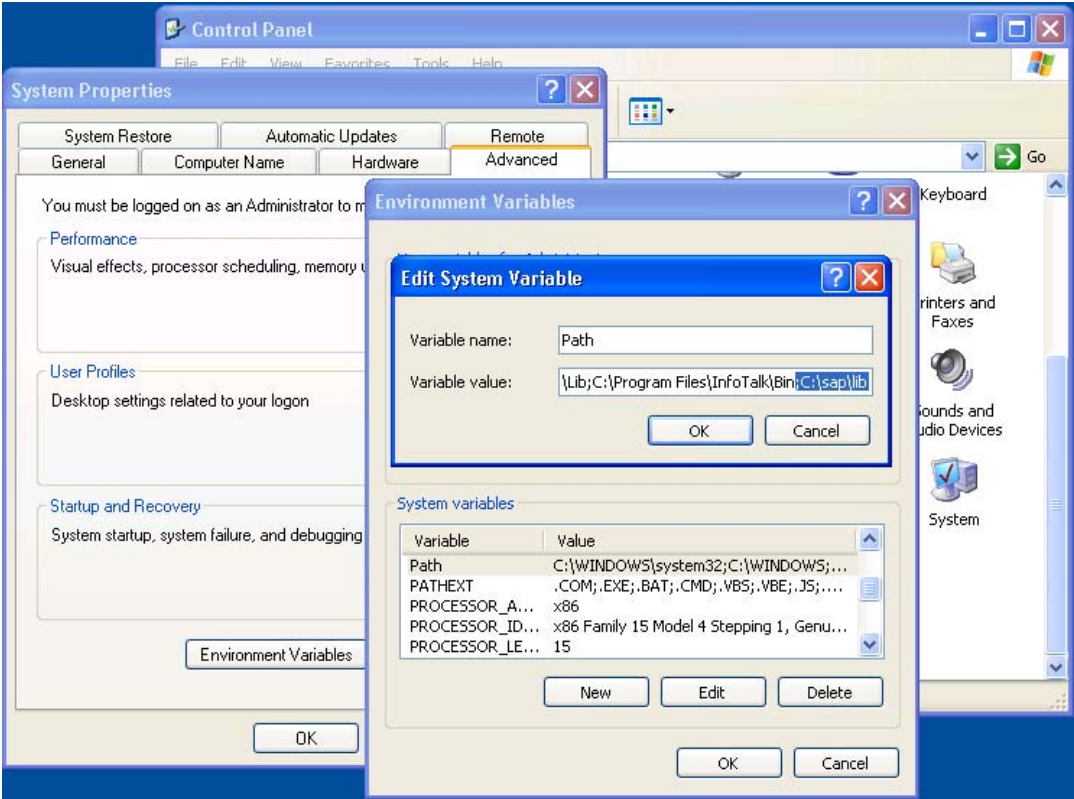
| Step | Description |
|------|---|
| 9. | <p>Configure Avaya IR to use the Server installed with InfoTalk-Speaker 2.0 as a Text-to-Speech (TTS) Server.</p> <p>To assign a default voice for TTS:</p> <ol style="list-style-type: none"> Click on Speech and DPR Administration → Administration. Select Text-to-Speech Configuration. The system displays the Text-to-Speech Configuration screen. In the Default Voice field, enter <i>Infotalk</i>. Select Submit. The system displays information about the success or failure of the administration attempt.  |
| | <p>To assign a TTS type:</p> <ol style="list-style-type: none"> Click on Speech and DPR Administration → Administration. Select Text-to-Speech Configuration. The system displays the Text-to-Speech Configuration screen. If a Text-to-Speech type has not been previously assigned, select Assign New Text-to-Speech Type, or if a Text-to-Speech type has been previously assigned, in the Text-to-Speech Type field, select Assign New. In the Text-to-Speech Type field, select the first available TTS type (TTS0 to TTS9) to assign. In the Engine field, select <i>mrcp</i>. Select Submit. The system displays information about the success or failure of the administration attempt.  |

| Step | Description |
|------|--|
| | <p>To assign a TTS server with an MRCP engine:</p> <ol style="list-style-type: none"> Click on Speech and DPR Administration → Administration. Select Text-to-Speech Configuration. The system displays the Text-to-Speech Configuration screen. In the Text-to-Speech Type field, select the TTS type (TTS0 to TTS9) assigned above for MRCP. Select Assign New Server. The system displays the Assign Text-to-Speech Server screen. In the Server Name field, type <i>name/media</i>, where <i>name</i> is the Computer name of the PC running InfoTalk-Speaker. In the IP Address field, type the IP address of the PC running InfoTalk-Speaker. In the Ports field, type the number of ports to be used. The number must be less than or equal to the number of licensed ports on the Avaya IR. In the Base Port field, enter 554. The Base Port setting must match the InfoTalk MRCP Server setting in Section 5 Step 6. Select Submit. The system assigns the TTS server and displays information about the success or failure of the administration attempt. <p>To complete the MRCP TTS configuration, stop and start the voice system.</p>  |

5. Configure InfoTalk-Speaker and InfoTalk MRCP Server

InfoTalk-Speaker and InfoTalk MRCP Server were installed on a Generic Pentium 4, 2.8 GHz server with 1 GB of memory running Microsoft Windows XP Professional with Service Pack 2. As all communication between the InfoTalk server and Avaya IR is via TCP/IP, it is strongly suggested that both systems be placed on the same IP subnet with minimal network traffic in order to minimize network latency.

| Step | Description |
|---|--|
| Installing InfoTalk-Speaker 2.0 software | |
| 1. | <p>The InfoTalk-Speaker 2.0 software is distributed on two CD-ROMs. To install, place the first CD-ROM into the drive and run the file Setup.exe. The installation runs through the following steps:</p> <ol style="list-style-type: none"> A welcome window will be displayed. Click Next to continue. Read and accept the license agreement and click Next. Select Complete for Setup Type and click Next. Select the destination folder and click Next. The default installation path is C:\Program Files\InfoTalk. Check the option Install JRE after Installation? and click Install. The installation wizard will install the product. At the end of installation process click on the “Finish” button. Restart the server after the installation. After the server has restarted, register InfoTalk-Speaker 2.0 License using the instructions found in <i>InfoTalk-Speaker Installation And User Guide</i> ⁴. |
| Installing InfoTalk MRCP Server 1.0 software | |
| 2. | <ol style="list-style-type: none"> Download JRE 1.4.2_09 or above from Sun http://java.sun.com/ and install it if JRE 1.4 is not installed in Step 1. Insert the CD-ROM containing InfoTalk MRCP Server 1.0 software into the drive and run InstallSAP.exe. The default installation path is C:\sap. To add the path C:\sap\lib to the System environment variable PATH, double-click System from Control Panel. Click Advanced tab and Environment Variables. Under System variables, select Path and click Edit. Add ;C:\sap\lib to the end of Variable value. Click Ok three times to exit the windows. |

| Step | Description | | | | | | | | | | | | |
|-----------------|---|----------|-------|------|------------------------------------|---------|---------------------------------------|----------------|-----|-----------------|---|-----------------|----|
| |  <p>The screenshot shows a Windows Control Panel window with the System Properties dialog box open. The Advanced tab is selected, and the Environment Variables button is clicked. This opens the Environment Variables dialog box, which in turn opens the Edit System Variable dialog box. The Edit System Variable dialog box shows the variable name 'Path' and its value '\Lib;C:\Program Files\InfoTalk\Bin;C:\sap\lib'. Below this, a table lists system variables:</p> <table border="1" data-bbox="755 751 1247 919"> <thead> <tr> <th>Variable</th> <th>Value</th> </tr> </thead> <tbody> <tr> <td>Path</td> <td>C:\WINDOWS\system32;C:\WINDOWS;...</td> </tr> <tr> <td>PATHEXT</td> <td>.COM;.EXE;.BAT;.CMD;.VBS;.VBE;.JS;...</td> </tr> <tr> <td>PROCESSOR_A...</td> <td>x86</td> </tr> <tr> <td>PROCESSOR_ID...</td> <td>x86 Family 15 Model 4 Stepping 1, Genu...</td> </tr> <tr> <td>PROCESSOR_LE...</td> <td>15</td> </tr> </tbody> </table> | Variable | Value | Path | C:\WINDOWS\system32;C:\WINDOWS;... | PATHEXT | .COM;.EXE;.BAT;.CMD;.VBS;.VBE;.JS;... | PROCESSOR_A... | x86 | PROCESSOR_ID... | x86 Family 15 Model 4 Stepping 1, Genu... | PROCESSOR_LE... | 15 |
| Variable | Value | | | | | | | | | | | | |
| Path | C:\WINDOWS\system32;C:\WINDOWS;... | | | | | | | | | | | | |
| PATHEXT | .COM;.EXE;.BAT;.CMD;.VBS;.VBE;.JS;... | | | | | | | | | | | | |
| PROCESSOR_A... | x86 | | | | | | | | | | | | |
| PROCESSOR_ID... | x86 Family 15 Model 4 Stepping 1, Genu... | | | | | | | | | | | | |
| PROCESSOR_LE... | 15 | | | | | | | | | | | | |

Configuring InfoTalk MRCP Server 1.0 software

3. Modify the file `sap_config.xml` found at `C:\SAP\Config`.
 - a. Locate the line as shown below and change `mrcp_server_name` to the Computer Name of the system:

```
<sap name="mrcp_server_name">
```

- b. Locate the lines below and comment the lines out as shown.

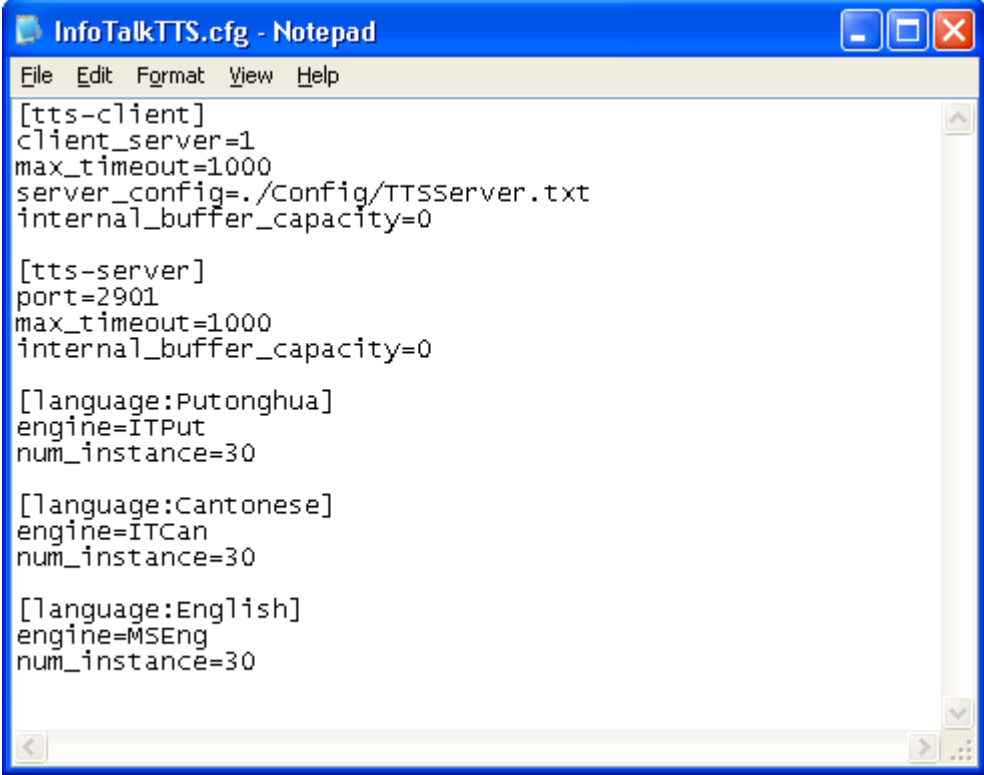
```
<!-- <audio-service name="ASR/recognizer" uri="ASR.xml" /> -->
<!-- <audio-service name="TTS/synthesizer" uri="TTS.xml" /> -->
```

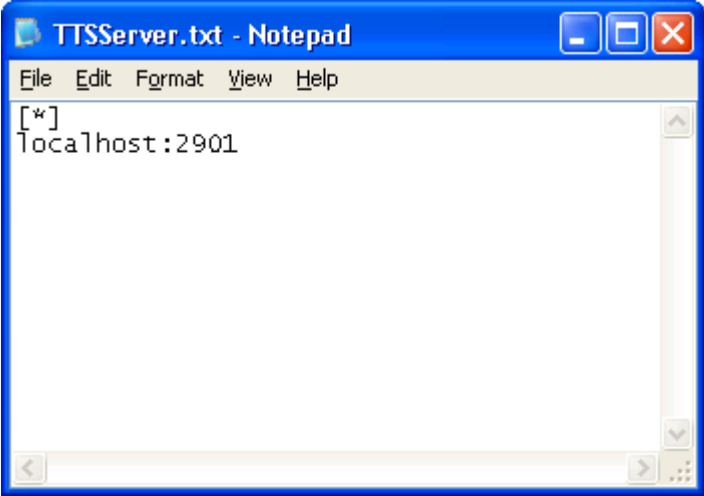
The following line must not be commented.

```
<audio-service name="TTS/media" uri="TTS.xml" />
```

- c. Locate the line as shown below. This property sets the encoding of the requests. The `data` field should be set to the default value “**UTF-8**” in order to support double-byte languages. If Avaya IR is configured to send requests in another encoding such as ISO-8859-1, change the `data` field value accordingly. To determine the encoding used by Avaya IR, locate the **encoding** attribute in

| Step | Description |
|------|---|
| | <p>the <code><xml></code> tag in the VoiceXML application file.</p> <pre data-bbox="412 306 1081 338"><property name="request-decoder" data="UTF-8"/></pre> <p>d. Locate the line as shown below.</p> <pre data-bbox="412 457 1073 489"><property name="tts-default-lang" data="en-US"/></pre> <p>This property sets the default language for the TTS server when the request is received without the language parameter. Below are the available values:</p> <pre data-bbox="412 638 691 779">"zh-CN": Putonghua "zh-TW": Mandarin "zh-HK": Cantonese "en-US": English</pre> |
| 4. | <p>Modify the file sap.properties found at C:\SAP\Config. The line shown below determines the location of the configuration file for the InfoTalk-Speaker TTS engine.</p> <pre data-bbox="412 932 1105 963">sap.InfoTalk.tts.configFile = ./config/InfoTalkTTS.cfg</pre> |
| 5. | <p>Modify the file InfoTalkTTS.cfg found at C:\SAP\Config.</p> <p>a. Setup the available TTS language and maximum simultaneous instances of each language. If a language is not available, comment it out by inserting “#” at the beginning of the line. The following setting configures the InfoTalk MRCP server to handle a maximum of 30 simultaneous instances for each TTS language.</p> |

| Step | Description |
|------|---|
| |  <p>The relevant TTS Server default configuration file in the folder <InfoTalk-Speaker installed path>\config must be changed as well, depending on what TTS language is configured for TTS.</p> <p>TTS_put.cfg - Putonghua and English TTS_can_put.cfg - Putonghua, Cantonese and English</p> |
| 6. | <p>Modify TTS.xml found at C:\SAP\Config.</p> <p>a. Locate the line as shown below. This property sets the port number of InfoTalk MRCP Server for the connection of MRCP client. It has to be matched with the settings in Section 4 Step 9 “Configure Avaya Interactive Response (IR)”.</p> <pre><property name="tcpport"> 554 </property></pre> |

| Step | Description |
|------|---|
| 7. | <p>Modify the file TTSServer.txt found at C:\SAP\Config. Specify the location of the InfoTalk-Speaker TTS Server that InfoTalk MRCP Server should connect to. The default values of location name and port number are as follows where both the software are installed on the same server. If InfoTalk-Speaker TTS Server is installed in another machine, its hostname or IP address has to be specified to replace the default “localhost”.</p>  |

5. Interoperability Compliance Testing

This Interoperability Compliance Test included feature functionality and serviceability testing. Feature functionality testing focused on verifying that InfoTalk-Speaker and InfoTalk MCRP Server could successfully work with the Avaya IR for the use of synthesized voice in system responses (via Text-to-Speech). Voice XML scripts in English, Cantonese (Traditional Chinese) and Putonghua (Simplified Chinese) were used on Avaya IR to test TTS. Serviceability tests were used to verify that the InfoTalk Server recovered from adverse conditions, such as rebooting of the InfoTalk server, Avaya IR, and Avaya Communication Manager and disconnecting the ethernet cable to the InfoTalk server.

5.1. General Test Approach

All feature functionality test cases were performed manually to verify proper operation. The general test approach included:

- Verifying Text-To-Speech Feature Functionality between Avaya IR, InfoTalk-Speaker and InfoTalk MCRP Server.
 - Complete synthesized prompts could be heard by the caller.
 - The Barge-in feature worked when DTMF was pressed.
 - Six simultaneous users could access the synthesized prompts.
 - The synthesized prompts could be heard in English, Cantonese and Putonghua.

5.2. Test Results

All feature functionality and serviceability test cases passed. InfoTalk-Speaker successfully communicated with Avaya IR using VoiceXML applications through the MRCPP V1 protocol. For serviceability testing, InfoTalk Server was able to recover after resets of the InfoTalk Server, Avaya IR and Avaya Communication Manager. The InfoTalk Server was also able to recover from network disconnects and reconnects.

6. Verification Steps

The status of the connectivity between the InfoTalk Server and Avaya IR can be obtained by viewing the Speech Resource Status Window in Avaya IR (**Web Administration** → **Speech and DPR Administration** → **Display Status** → **Speech Resource Status**).

Select the **Resource Type** that was configured for TTS, then click **Submit**.

Verify that the **STATE** of the ports show **INSERV**. When TTS resources are being requested by the Avaya IR, the **CHAN** field will be populated with the corresponding IR channel.

- [Renumber Voice Channels](#)
- [Report Voice System Status](#)
- [Start Voice System](#)
- [Stop Voice System](#)
- Switch Interfaces
 - [Voice over IP](#)
- Voice Equipment
 - [Display Equipment](#)
 - [Equipment State](#)
 - [Channels to Groups](#)
 - [Phone Number](#)
 - [Display Passwords](#)
 - Voice Services
 - [Channel Services](#)
 - [Number Services](#)

Feature Packages

- Speech and DPR Administration
 - [Display Status](#)
 - [Administration](#)
- [Universal Call ID Administration](#)

Reports

- [Call Data Handling Reports](#)
- [Message Log Report](#)

[Help](#)

RESOURCES: TTS
Default Voice: Infotalk

ALL_SERVERS SUMMARY TOTAL PORTS AVAILABLE: 29
SERVER: pc6/media IP: 10.1.10.106
PORT CAPACITY: 30 PORTS AVAILABLE: 29

| PORT | STATE | CHAN |
|------|--------|------|
| 0 | INSERV | N/A |
| 1 | INSERV | N/A |
| 2 | INSERV | N/A |
| 3 | INSERV | N/A |
| 4 | INSERV | N/A |
| 5 | INSERV | N/A |
| 6 | INSERV | 5 |
| 7 | INSERV | N/A |
| 8 | INSERV | N/A |
| 9 | INSERV | N/A |
| 10 | INSERV | N/A |
| 11 | INSERV | N/A |
| 12 | INSERV | N/A |
| 13 | INSERV | N/A |
| 14 | INSERV | N/A |
| 15 | INSERV | N/A |
| 16 | INSERV | N/A |
| 17 | INSERV | N/A |
| 18 | INSERV | N/A |
| 19 | INSERV | N/A |
| 20 | INSERV | N/A |
| 21 | INSERV | N/A |
| 22 | INSERV | N/A |
| 23 | INSERV | N/A |
| 24 | INSERV | N/A |
| 25 | INSERV | N/A |

7. Support

For technical support on InfoTalk-Speaker and MRCP Server contact:

Telephone : +65 6890-5967
Fax : +65 6895-4110
Email : Chunlei.pan@infotalkcorp.com

8. Conclusion

These Application Notes describe the compliance-tested configuration used to validate Avaya Communication Manager 3.0.1 and Avaya Interactive Response 1.3 with InfoTalk-Speaker 2.0 and InfoTalk MRCP Server 1.0. All test cases were completed successfully.

9. Additional References

This section references the product documentations that are relevant to these Application Notes.

The following documents are available at <http://support.avaya.com>.

- [1] *Avaya Interactive Response Online Documentation*.
- [2] *Administrator's Guide for Avaya Communication Manager*, Issue 1, June 2005, Document ID 03-300509.
- [3] *Avaya Communication Manager Release 3.0 Call Center Software Automatic Call Distribution (ACD) Guide*, Issue 1, June 2005, Document ID 07-300301.
- [4] *Avaya Interactive Response Release 1.2.1 MRCP Installation and Reference Guide*, Issue 1.0, November 2004, Document ID 107222.

The following documents are available from InfoTalk:

- [4] *InfoTalk-Speaker Installation And User Guide*, Version 2.0.X.303
- [5] *InfoTalk-Speaker Developer's Guide*, Version 2.0.X.303

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