



Avaya Solution & Interoperability Test Lab

Application Notes for Beijing InfoQuick SinoVoice Speech Technology jTTS 6.0 with Avaya Aura® Experience Portal 6.0 – Issue 1.0

Abstract

These Application Notes describe the configuration steps required to integrate Beijing InfoQuick SinoVoice Speech Technology (SinoVoice) jTTS 6.0 with Avaya Aura® Experience Portal 6.0. SinoVoice jTTS uses the Media Resource Control Protocol (MRCP) version 2 for its Text-To-Speech (TTS) features to interface with VoiceXML applications running on the Avaya Aura® Experience Portal.

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

These Application Notes describe the configuration steps required to integrate Beijing InfoQuick SinoVoice Speech Technology (SinoVoice) jTTS 6.0 with Avaya Aura® Experience Portal 6.0. Beijing InfoQuick SinoVoice jTTS uses the Media Resource Control Protocol (MRCP) version 2 for its Text-To-Speech (TTS) features to interface with the VoiceXML (VXML) applications running on Avaya Aura® Experience Portal.

Beijing InfoQuick SinoVoice jTTS is the core text-to-speech technology of SinoVoice which uses large scale recorded voice library and algorithm based on hierarchical prosody structure matching.

2. General Test Approach and Test Results

The general test approach is to place calls manually to Avaya Aura® Experience Portal running VXML applications that uses the TTS resources of SinoVoice jTTS.

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

The interoperability compliance test included feature and serviceability testing. The feature testing focused on placing calls to Avaya Aura® Experience Portal 6.0 that ran VoiceXML applications that use the TTS engines on the SinoVoice jTTS solution. The compliance test focused on placing calls to verify accurate TTS synthesis.

The serviceability testing focused on verifying the ability of the SinoVoice jTTS solution to recover from adverse conditions, such as rebooting of SinoVoice jTTS and Avaya Aura® Experience Portal 6.0 and disconnecting the LAN cables to the SinoVoice jTTS server.

2.2. Test Results

All test cases passed. Avaya Aura® Experience Portal 6.0 was successful in running applications that use the TTS resources of the SinoVoice jTTS solution.

2.3. Support

For technical support on SinoVoice jTTS, contact the SinoVoice support team at:

- Phone: +86-10-82826886
- Email: sinovoicesupport@sinovoice.com.cn

3. Reference Configuration

Figure 1 illustrates the test configuration used to verify the SinoVoice jTTS solution. SinoVoice jTTS was installed on a Microsoft Windows 2003 R2 Server with Service Pack 2. VoiceXML applications were installed on a HTTP server. Avaya Aura® Experience Portal is connected to Avaya Aura® Session Manager and Avaya Aura® Communication Manager using SIP VoIP Connections. Avaya IP telephones were used to place calls to Avaya Aura® Experience Portal, which would run the VoiceXML applications. The applications would use the SinoVoice jTTS Server for speech synthesis.

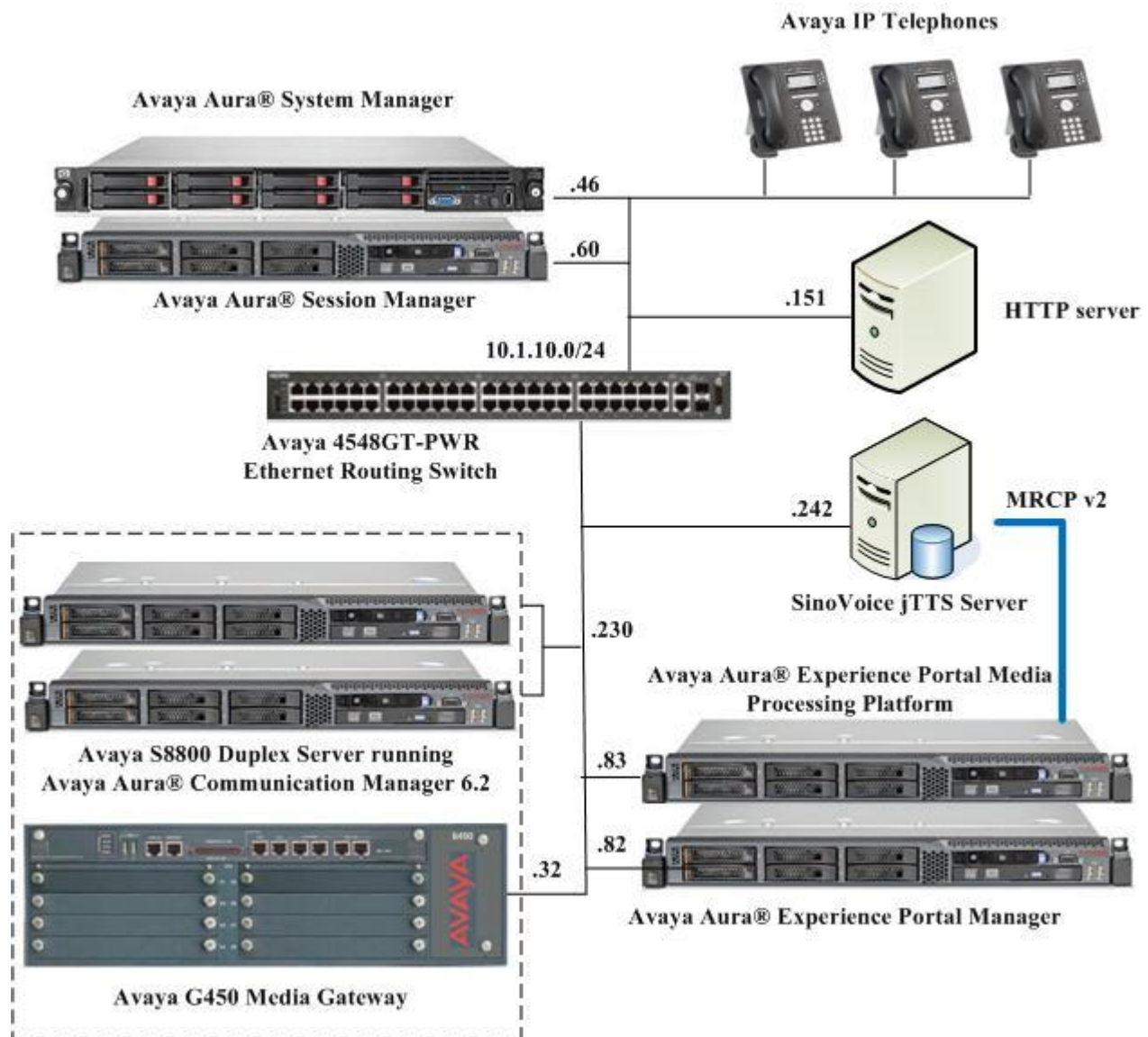


Figure 1: Test Configuration of SinoVoice jTTS with Avaya Aura® Experience Portal 6.0

4. Equipment and Software Validated

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

Equipment/Software	Release/Version
Avaya Aura® Experience Portal 6.0 on Avaya S8800 Server	R6.0 SP1
Avaya Aura® Communication Manager on Avaya S8800 Server (Duplex)	R6.2 SP2.01
Avaya G450 Media Gateway	31.22.0
Avaya Aura® System Manager on HP DL360 G7	6.2 SP 3
Avaya Aura® Session Manager on Avaya S8800 Server	6.1 SP 3
Avaya 9621 IP Telephones	6.2 SP2 (H.323)
Avaya 4548GT-PWR Ethernet Routing Switch	V6.2.4.010
SinoVoice jTTS on Microsoft Windows Server 2003 R2 Standard Edition SP2	6.0
HTTP server on Windows Vista Business Edition	Service Pack 2

5. Configure Avaya Aura® Communication Manager

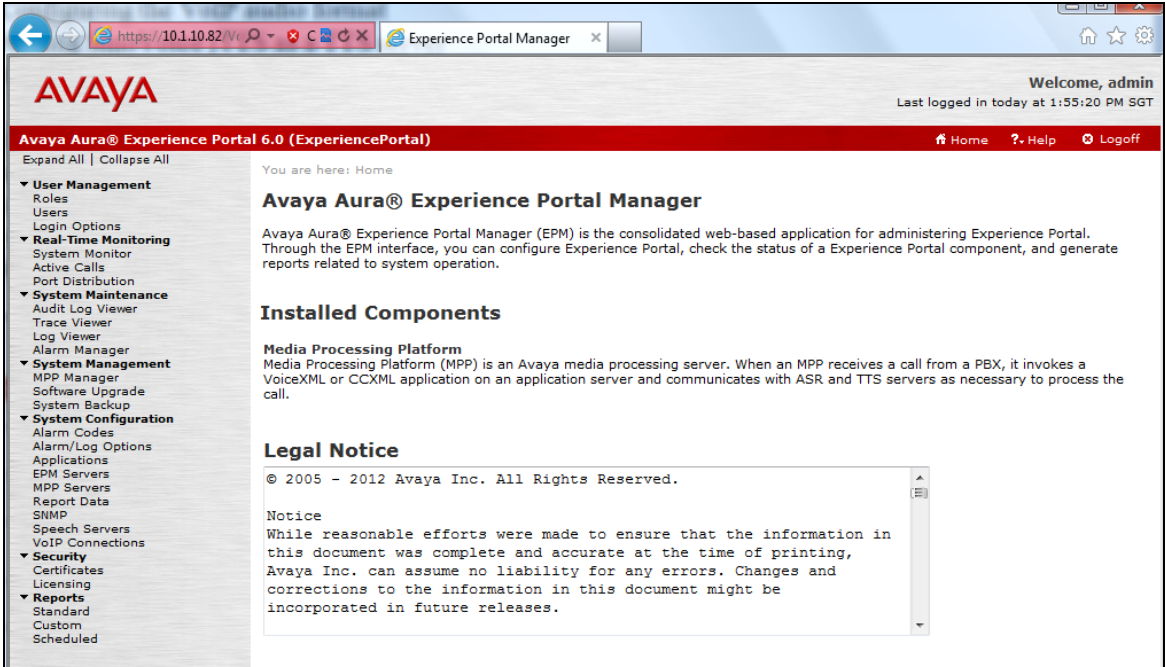
The configuration of the SIP Trunks between Communication Manager and Session Manager, and the routing of calls to Experience Portal are assumed to be in place and will not be discussed here. This section provides the additional procedures to configure Communication Manager for the purpose of administering SinoVoice jTTS. The configuration is performed via the System Access Terminal (SAT).

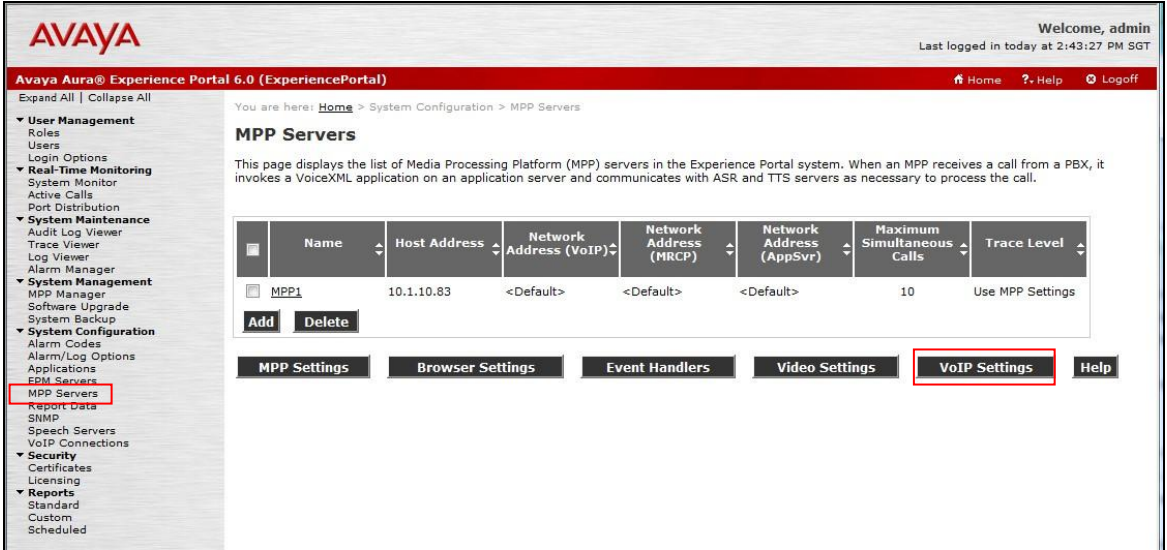
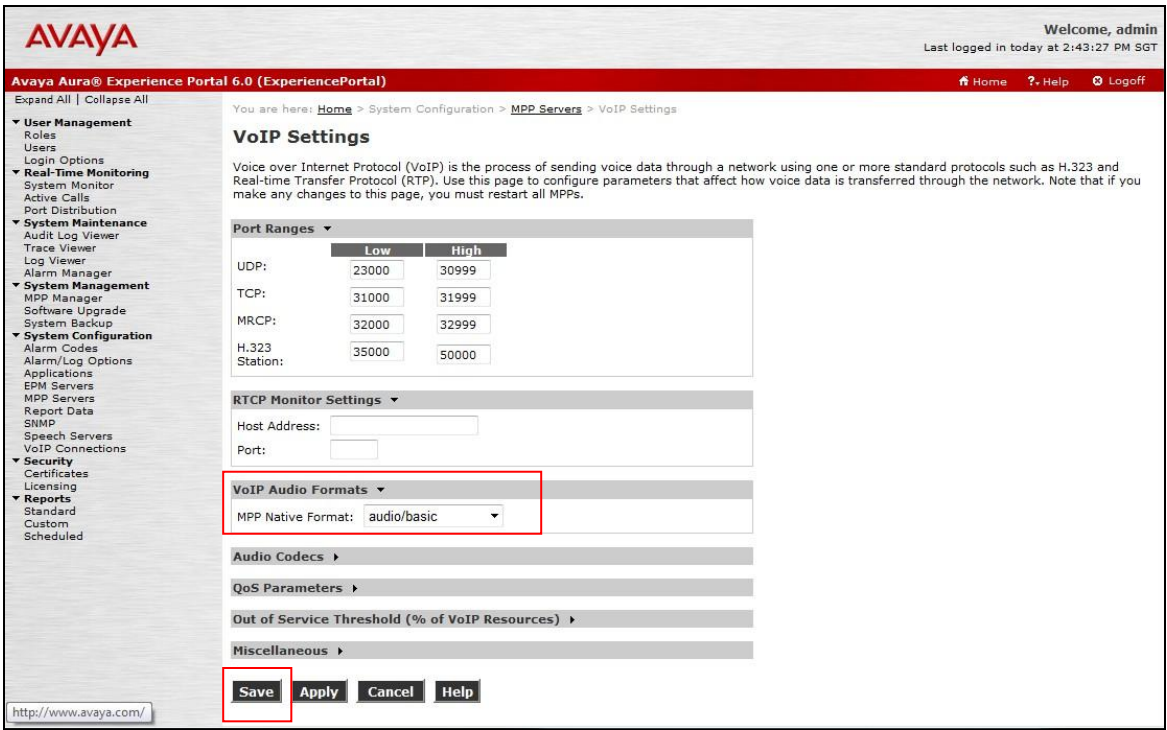
Step	Description																																
1.	Enter the change ip-codec-set n command where n is a valid IP codec-set associated with the IP network region that is used by Experience Portal, typically the IP network region assigned to the Session Manager SIP Trunk signaling group. Set Audio Codec to an appropriate value supported by Avaya Experience Portal and SinoVoice jTTS. In this configuration, the G.711Mu codec was used.																																
<div>change ip-codec-set 6<div>Page1 of 2</div></div> <div>IP Codec Set</div> <div>Codec Set: 6</div> <table><thead><tr><th>Audio Codec</th><th>Silence Suppression</th><th>Frames Per Pkt</th><th>Packet Size (ms)</th></tr></thead><tbody><tr><td>1: G.711MU</td><td>n</td><td>2</td><td>20</td></tr><tr><td>2:</td><td></td><td></td><td></td></tr><tr><td>3:</td><td></td><td></td><td></td></tr><tr><td>4:</td><td></td><td></td><td></td></tr><tr><td>5:</td><td></td><td></td><td></td></tr><tr><td>6:</td><td></td><td></td><td></td></tr><tr><td>7:</td><td></td><td></td><td></td></tr></tbody></table>		Audio Codec	Silence Suppression	Frames Per Pkt	Packet Size (ms)	1: G.711MU	n	2	20	2:				3:				4:				5:				6:				7:			
Audio Codec	Silence Suppression	Frames Per Pkt	Packet Size (ms)																														
1: G.711MU	n	2	20																														
2:																																	
3:																																	
4:																																	
5:																																	
6:																																	
7:																																	

6. Configure Avaya Aura® Experience Portal

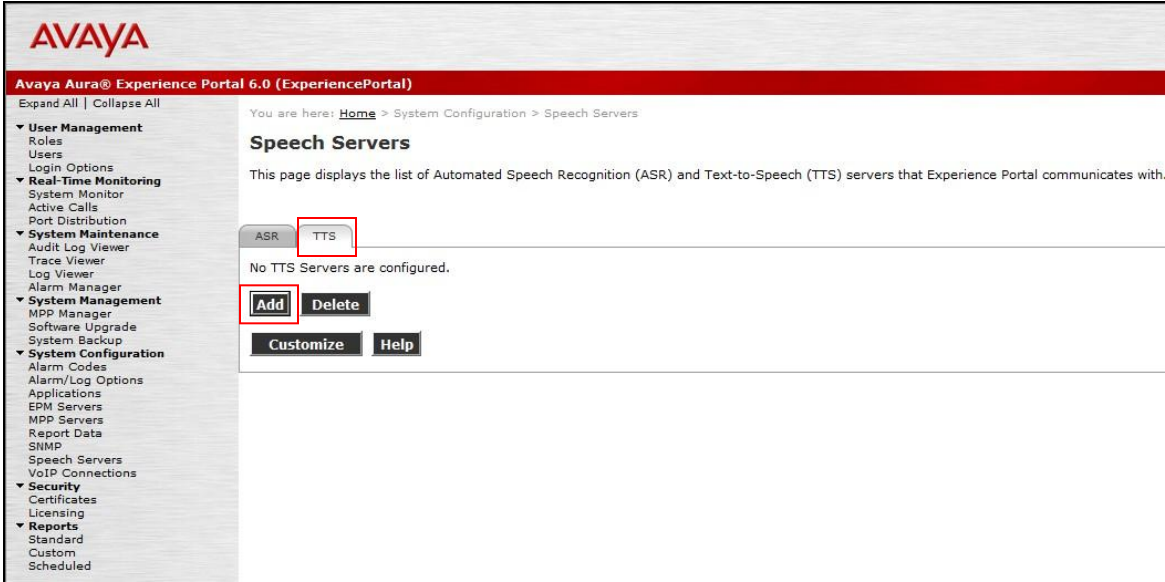
The initial administration of Avaya Aura® Experience Portal and the configuration of the SIP VoIP Connection to Session Manager are assumed to be in place and will not be discussed here. This section covers the additional procedures of Avaya Aura® Experience Portal that is required for the purpose of administering SinoVoice jTTS. The following steps will be covered:

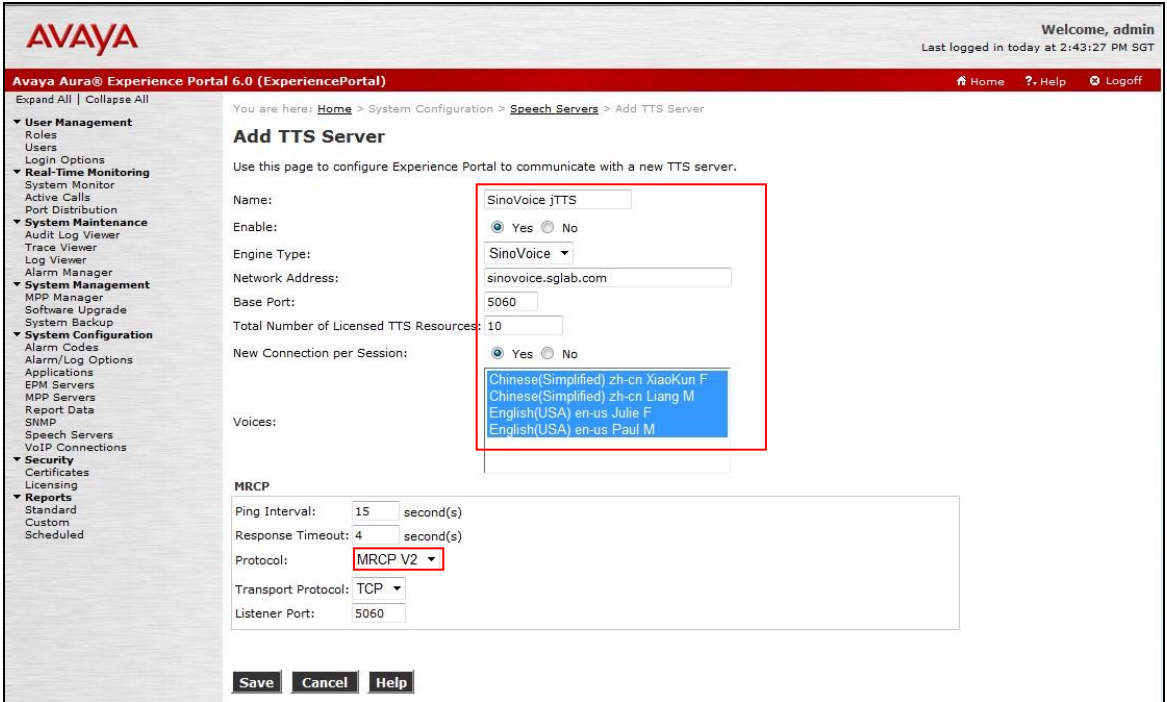
- Configuring the VoIP audio format
- Adding SinoVoice jTTS as a TTS server
- Adding applications

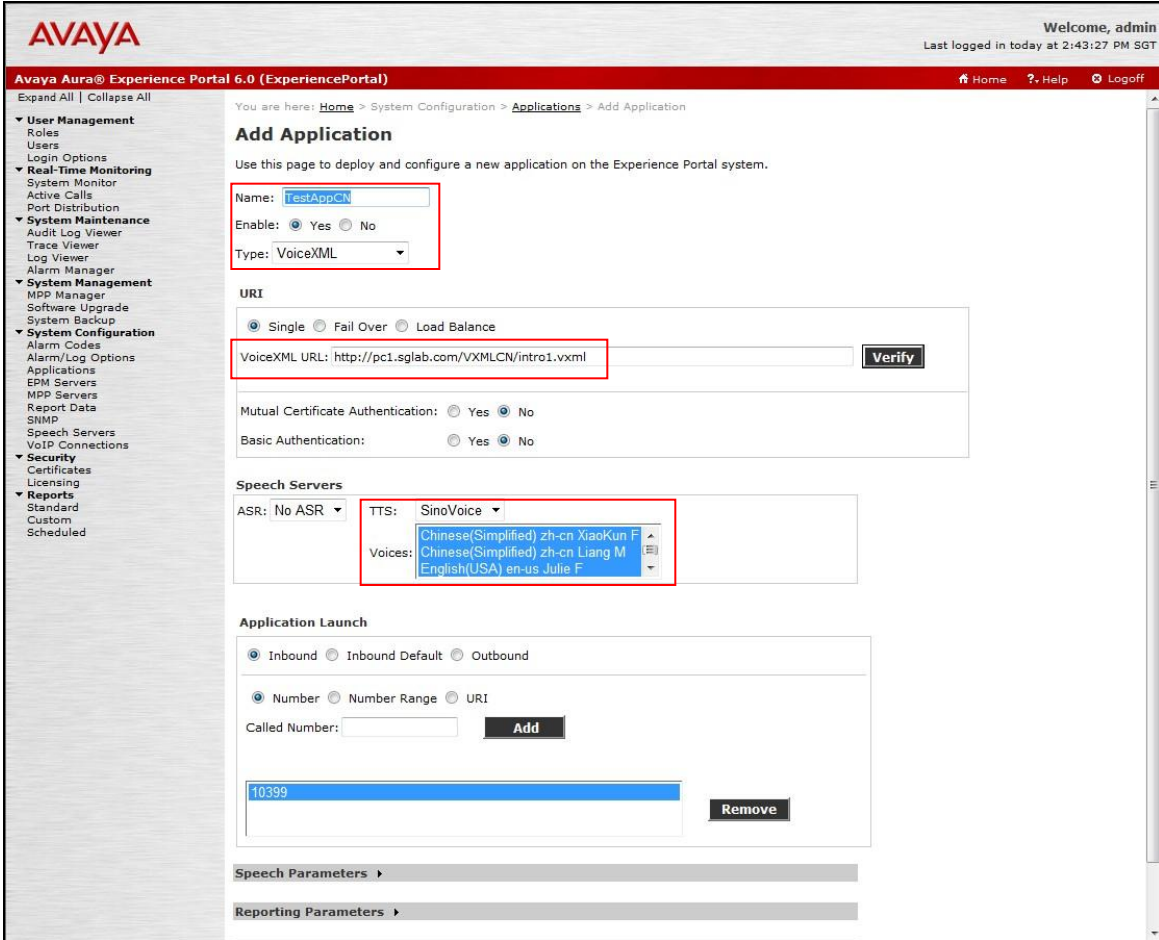
Step	Description
1.	<p>Avaya Aura® Experience Portal is configured via the Experience Portal Manager (EPM) web interface. To access the web interface, enter https://<ip-addr> as the URL in an internet browser, where <ip-addr> is the IP address of the EPM. Log in using an account with the Administration role to display the main page.</p> 

Step	Description
2.	<p>To configure the codec used by the Media Processing Platform (MPP) server, click System Configuration → MPP Servers in the left pane and click VoIP Settings.</p> 
3.	<p>Set MPP Native Format to audio/basic to configure the MPP server for G.711 mu-law to match the configuration on Communication Manager in Section 5. Scroll down the page and click Save.</p> 

Step	Description
5.	<p>Locate the <code>languages.properties</code> file found in <code>/opt/Tomcat/apache-tomcat-6.0.32/webapps/VoicePortal/WEB-INF/classes/config</code>. Edit the file and add the fields and lines shown below to the appropriate section.</p> <pre> # # Engine Type options displayed on the page # asrEngines=IBM WVS,Loquendo,Nuance ttsEngines=IBM WVS,Loquendo,Nuance,SinoVoice asrEnginesAmsOnly=Nuance ttsEnginesAmsOnly=Nuance # Engine Type conversion from display to internal data in the databas < Some lines removed for brevity > SinoVoiceTTS=sinovoice # Engine Type conversion from internal data in the database to display < Some lines removed for brevity > sinovoice=SinoVoice # TTS LANGUAGE < Some lines removed for brevity > SinoVoiceTTSLanguages=zh-cn XiaoKun F,zh-cn Liang M,en-us Julie F,en-us Paul M # # Language Default # < Some lines removed for brevity > # SinoVoiceTTSLanguagesDefault=zh-cn XiaoKun F # # default base port # < Some lines removed for brevity > SinoVoiceBasePort=5060 # # default New Connection per Session # < Some lines removed for brevity > SinoVoicePerPort=Yes # # default URL # < Some lines removed for brevity > SinoVoiceRtspUrlTts=/media/sinovoicesynthesizer </pre>

Step	Description
	<pre> < Some lines removed for brevity > # # MRCP Protocol # < Some lines removed for brevity > SinoVoiceMRCPValues=mrcpv1,mrcpv2 # # Transport # < Some lines removed for brevity > SinoVoiceTransportValues=tcp < remaining lines removed for brevity > </pre>
6.	Reboot the EPM server for the above changes to take effect.
7.	<p>To configure the SinoVoice jTTS server, click System Configuration → Speech Servers. Click the TTS tab and click Add.</p> 

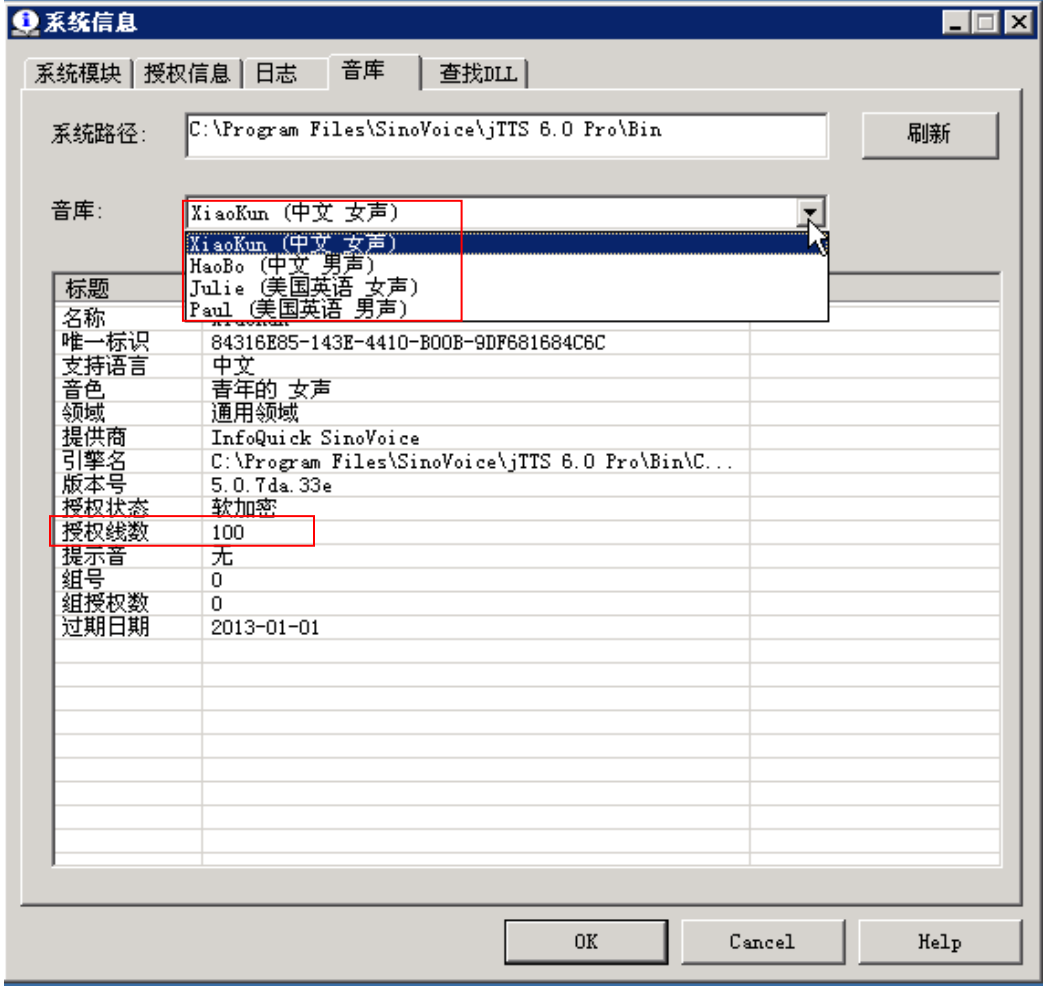
Step	Description
8.	<p>In the Add TTS Server page, select SinoVoice as the Engine Type. This engine type option was added by modifying the <code>languages.properties</code> files in Steps 4 and 5. In the MRCP section, set Protocol to MRCP V2. Specify the Name, select Yes to Enable, set Network Address to the IP address or Full FQDN of the SinoVoice jTTS Server and select the desired Voices used by the applications. The Total Number of Licensed TTS Resources should also be set to the number of licenses available on the SinoVoice jTTS Server. All other fields were left at their default values. Click Save.</p> 

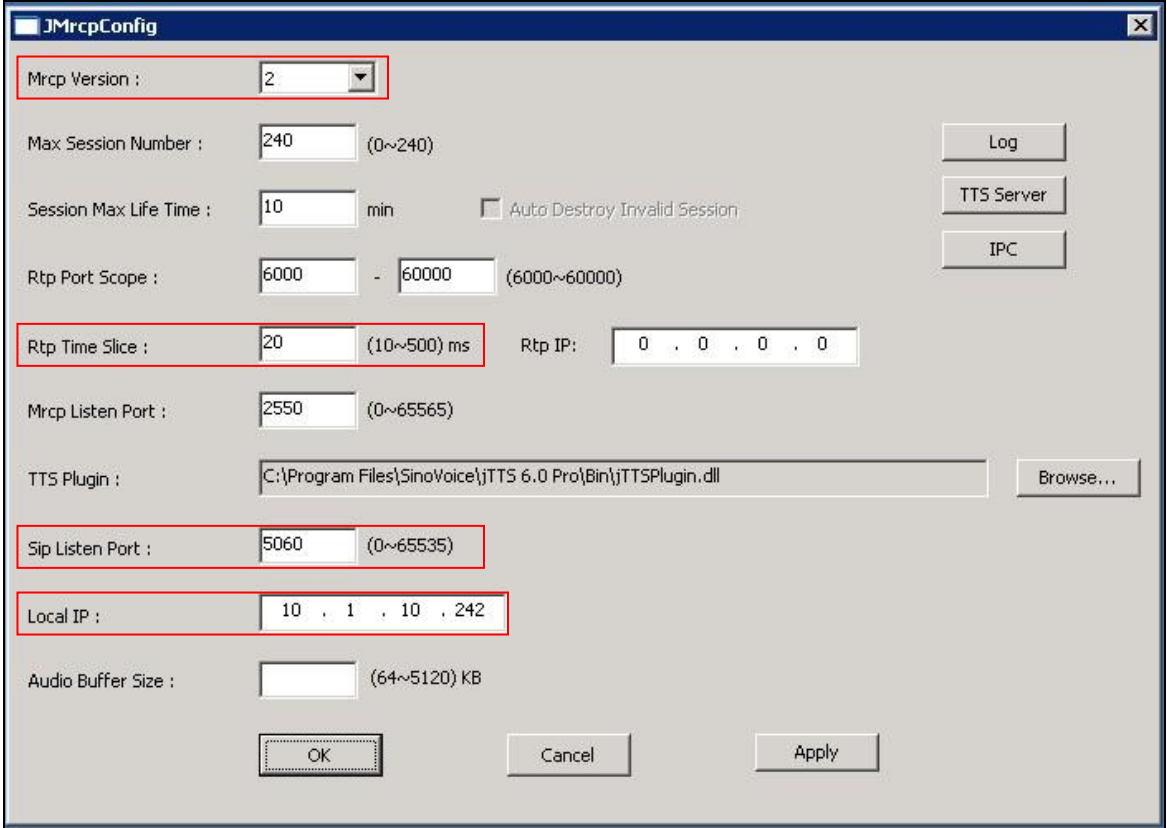
Step	Description
9.	<p>To assign SinoVoice jTTS to an Avaya Experience Portal application, click System Configuration → Applications and then click Add on the Applications page (not shown). Configure the Add Application page as shown below. This configuration assigns the default Avaya Aura® Experience Portal test application deployed on the http server to the called number 10399. Specify the Name, select Yes to Enable, set MIME Type to VoiceXML and set VoiceXML URL to HTTP server address location of the VoiceXML script. Select SinoVoice for TTS and then select the appropriate Voices to use. Click Save (not shown).</p> <p>Repeat this procedure to assign SinoVoice jTTS to other Experience Portal applications.</p> 

7. Configure SinoVoice jTTS

The following components are required to run the SinoVoice jTTS MRCP Server. In this test configuration, both the jTTS Platform Engine and jMRCP Server are installed on the same server.

- jTTS Platform Engine
- jTTS Voice Library
- jMRCP Server

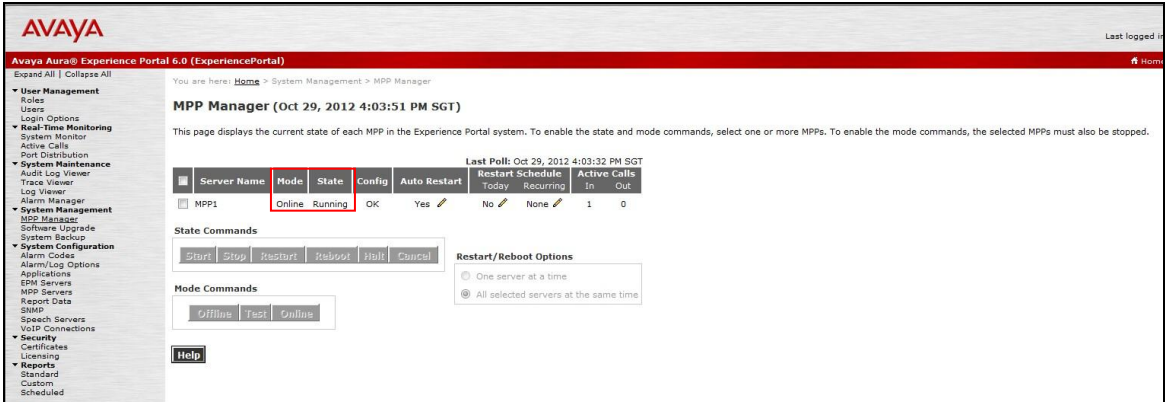
Step	Description
1.	<p>On the SinoVoice jTTS server, click Start → All Programs → jTTS 6.0 Professional → jTTS System Information. On the jTTS SysInfo (系统信息) window, click the Voice (音库) tab. Verify that the desired voices are installed. Verify also that the value for Lines (授权线数) shows sufficient number of license required. Click OK to close the window.</p> 

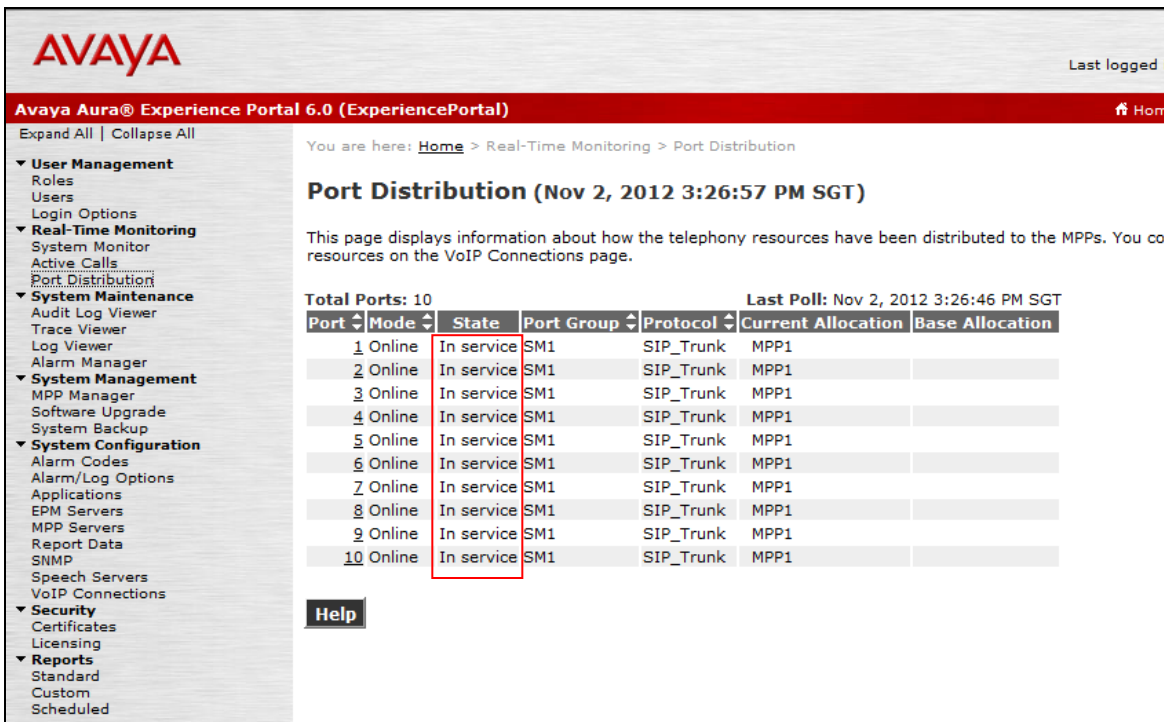
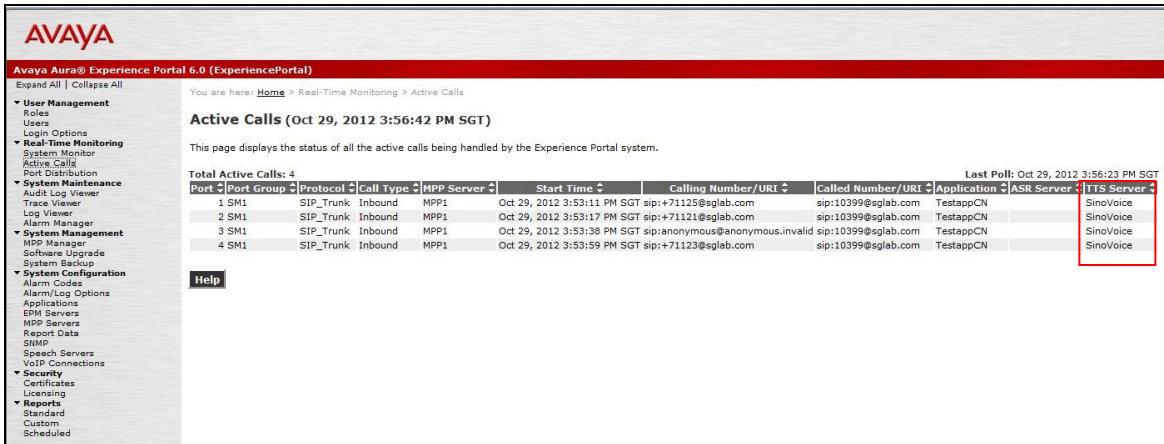
Step	Description
2.	<p>Click Start → All Programs → jTTS 6.0 Professional → Mrcp Config Tool. On the jMrcpConfig window, configure as shown below.</p> <ul style="list-style-type: none"> • Mrcp Version: Select 2. • Rtp Time Slice: Enter 20. • Sip Listen Port: Enter 5060, which is the default port value for SIP. • Local IP: Enter the SinoVoice jTTS Server IP address as shown in Figure 1. <p>Click Apply, and then OK to complete the configuration.</p> 

8. Verification Steps

This section provides the verification steps that may be performed to verify that Avaya Aura® Experience Portal can run VoiceXML applications that use the SinoVoice jTTS for TTS speech synthesis.

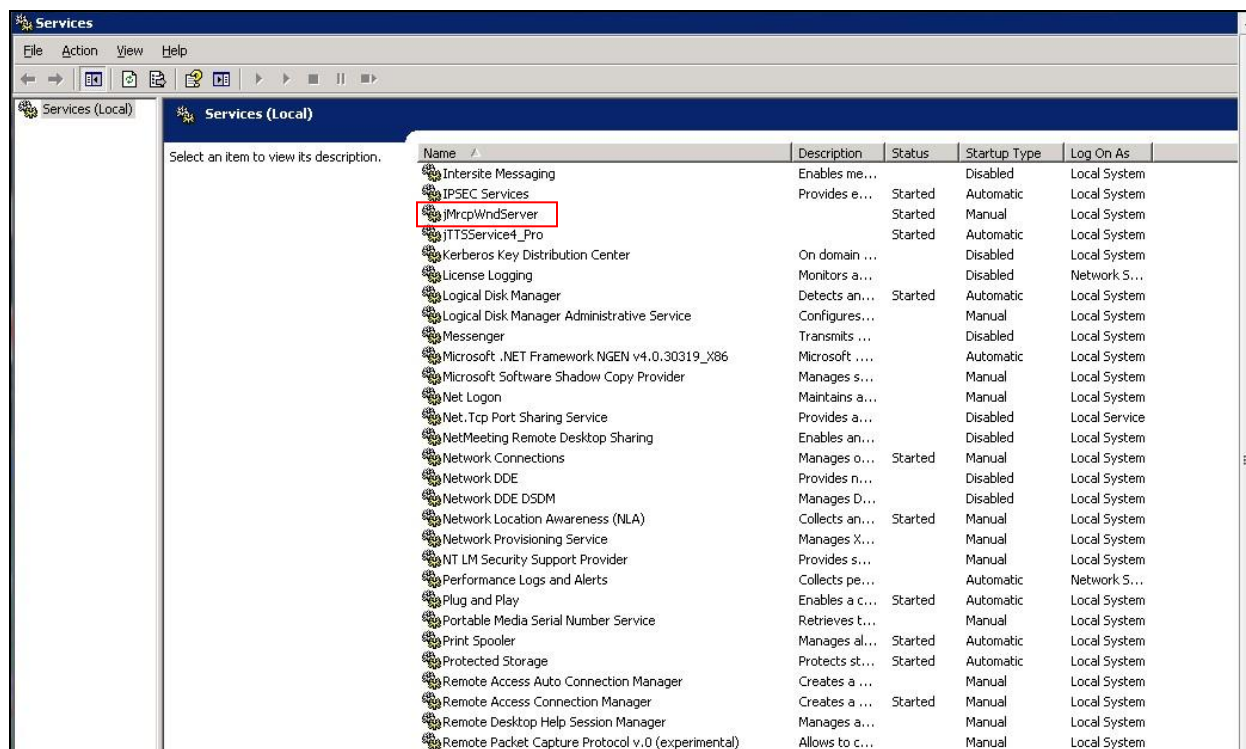
8.1. Verify Avaya Aura® Experience Portal

Step	Description
1.	<p>From the VPMS web interface, click MPP Manager on the left pane. On the MPP Manager page, verify that the MPP server is Online and Running.</p> 

Step	Description
2.	<p>From the VPMS web interface, click Port Distribution on the left pane. On the Port Distribution page, verify that the State of the ports on the MPP server is In service.</p> <div></div>
3.	<p>Place some calls to Avaya Aura® Experience Portal that runs a VoiceXML script which uses the SinoVoice jTTS for speech synthesis. Verify that the application answers the calls and that the application is able to announce the TTS synthesized prompts to the caller. From the VPMS web interface, click Active Calls on the left pane and verify that the TTS Servers in use is SinoVoice jTTS.</p> <div></div>

8.2. Verify SinoVoice jTTS

On the SinoVoice jTTS server, click **Start** → **Administrative Tools** → **Services**. In the Services window, verify that the **jMrpWndServer** is started.



9. Conclusion

These Application Notes describe the compliance-tested configuration used to validate Avaya Aura® Experience Portal 6.0 with Beijing InfoQuick SinoVoice jTTS 6.0. All test cases were completed successfully.

10. Additional References

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

[1] *Administering Avaya Aura® Experience Portal*, Aug 2011.

Product information on Beijing InfoQuick SinoVoice jTTS 6.0 can be found at <http://www.sinovoice.com/english/jtts.html>.

©2012 Avaya Inc. All Rights Reserved.

Avaya and the Avaya Logo are trademarks of Avaya Inc. All trademarks identified by ® and ™ are registered trademarks or trademarks, respectively, of Avaya Inc. All other trademarks are the property of their respective owners. The information provided in these Application Notes is subject to change without notice. The configurations, technical data, and recommendations provided in these Application Notes are believed to be accurate and dependable, but are presented without express or implied warranty. Users are responsible for their application of any products specified in these Application Notes.

Please e-mail any questions or comments pertaining to these Application Notes along with the full title name and filename, located in the lower right corner, directly to the Avaya DevConnect Program at devconnect@avaya.com.