

Avaya Solution & Interoperability Test Lab

Application Notes for configuring Logicall IVR from Maximum Network Solutions with Avaya Communication Server 1000E R7.0 and Avaya Network Routing Server R7.0 - Issue 1.0

Abstract

These Application Notes describe the configuration steps necessary for provisioning Maximum Network Solutions Logicall IVR to successfully interoperate with Avaya Communication Server 1000E R7.0 and Avaya Network Routing Server R7.0.

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 a compliance-tested configuration of the interoperability of Logicall IVR from Maximum Network Solutions to successfully interoperate with Avaya Communication Server 1000E R7.0 and Avaya Network Routing Server R7.0 (NRS). Maximum Network Solutions Logicall is a voice processing platform capable of supporting both DTMF (touch tone) and Natural Speech. Maximum Network Solutions Logicall provides a platform on which voice applications and speech self-service applications can be installed enabling the automation of telephone transactions and is available as a pre-configured, out of the box solution.

2. General Test Approach and Test Results

The Maximum Network Solutions Logicall solution (Logicall) resides on Solaris R10 platform installed on a Sun Microsystems server and is administered over a telnet session using a program such as Putty or Reflections. Each solution is pre-configured according to the end-users' specifications. The configuration, regarding connectivity to the Avaya Communication Server 1000E (CS1000E), is contained in a configurable .ini file called Control.ini, an example of which can be seen in the Appendix of these Application Notes. The test approach was to successfully place calls to the Logicall system over SIP trunks between Logicall and the CS1000E. Calls from the CS1000E to Logicall route through the NRS. Outbound calls from Logicall that terminate on the CS1000E are routed directly to the CS1000E SIP Gateway by-passing the NRS.

2.1. Interoperability Compliance Testing

During interoperability compliance testing the following types of calls were made and features of Logicall covered.

- Calls placed into Logicall from CS1000E 1140E IP deskphones and M3820 digital deskphones
- Calls placed into logicall from PSTN users
- Calls transferred to Logicall from CS1000E 1140E IP deskphones and M3820 digital deskphones
- Calls transferred to Logicall from PSTN users
- Calls transferred to Logicall from an Avaya Callpilot menu
- Outbound calls from CS1000E 1140E IP deskphones and M3820 digital deskphones from Logicall
- Blind transfers to CS1000E 1140E IP deskphones and M3820 digital deskphones
- Blind transfers to CS1000E users logged into Automatic Call Distribution (ACD) Queue
- Automatic Speech Recognition (ASR) test on Logicall
- Text to Speech (TTS) test on Logicall

2.2. Test Results

All tests outlined in the Test Plan document passed successfully. Below are listed observations following the compliance test of this solution.

- 1. When sending calls from Logicall to deskphones on the CS1000E over SIP trunks, the IP address to which the calls were sent was the SIP Gateway IP address on the CS1000E and not the Network Routing Server IP address. This is not the Avaya recommended setup for such a scenario but was specifically requested by Maximum Network Solutions for this compliance testing.
- 2. When transferring calls into Logicall from the CS1000E, the B-Party CLID was passed to Logicall and not the A-Party CLID. This was the preferred method and thus the design intent by Maximum Network Solutions as SIP updates are not processed by the Logicall solution, meaning the A-Party CLID cannot be passed to Logicall.

2.3. Support

For more information on Maximum Network Solutions (Maxnet) and product support visit <u>http://www.maxnet.co.uk</u>. The following is the contact information for Maxnet:

Maximum Network Solutions The Old Granary, The Square, Sheffield, South Yorkshire, S26 5QN +44 1909 774477 www.maxnet.co.uk

3. Reference Configuration

The configuration in **Figure 1** was used to compliance test Logicall with the CS1000E using SIP trunks to pass calls from the CS1000E into Logicall. Calls from the CS1000E deskphones to Logicall route via an NRS. Calls from Logicall to the CS1000E by-pass the NRS and are routed directly to the CS1000E SIP Gateway.

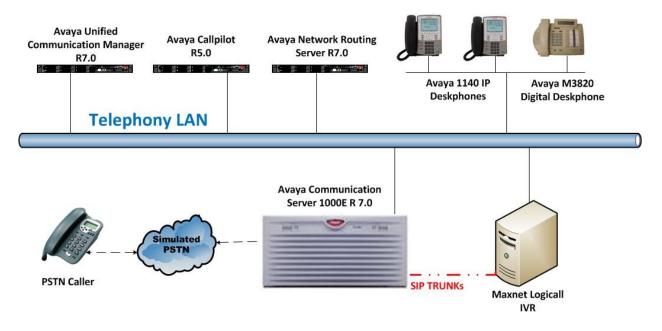


Figure 1: Connection of Maximum Network Solutions Logicall IVR and Avaya Communication Server 1000E R7.0.

4. Equipment and Software Validated

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

Equipment Description	Software Release
Avaya Communication Server 1000E CPPM	Avaya Communication Server 1000E R7.0 **See Appendix for patch list used**
Avaya PVI NRS (IBM x3350)	Avaya Network Routing Server R7.0
Avaya PVI UCM (IBM x3350)	Avaya Unified Communications Management R7.0
Avaya 1140 UNIStim Deskphone	UNIStim V0625C8D
Avaya 3820 Digital Set	N/A
Avaya Call Pilot 600r Server	Avaya Call Pilot Version 5.00.41 Patch Line-up:CP50041SU08S CP500508G09C
Maximum Network Solution Sun Microsystems	Maximum Network Solutions Solaris R10 Maximum Network Solutions Logicall IVR R1.0.0

5. Configuration of Avaya Communication Server 1000E

The configuration operations illustrated in this section were performed using terminal access to the CS1000E over a telnet session. It is assumed a fully working CS1000E is in place with SIP trunks configured and a Route List Index (RLI) associated with the route for the SIP. For all other provisioning information, such Administering Avaya CS1000E, refer to product documentation in **Section 10** of these Application Notes.

Note: Not all prompts need an answer. The prompts highlighted below are mandatory for a basic configuration. Accept the default responses for all other prompts by pressing the Return key.

5.1. Add a Distant Steering Code in a Coordinated Dialling Plan for Logicall

In order to send calls to Logicall, a Directory Number (DN) must be associated with each Logicall application. For compliance testing one application was created in Logicall and therefore one DN must be associated with this application in order to call it. In the example below 6000 is chosen, this DN is created in overlay 87 as type Distant Steering Code (DSC) in a Coordinated Dialling Plan (CDP). Type **LD 87** to enter Overlay 87, then create a new **cdp** and add **6000** as the **dsc** as shown below.

Prompt	Response	Description
>	LD 87	Enter Overlay 87
REQ	NEW	New
CUST	0	Customer 0
FEAT	cdp	Coordinated Dialling Plan
TYPE	dsc	Distant Steering Code
DSC	6000	6000 is the DN associated with Logicall
FLEN	4	Number Length (4 digits)
RLI	14	Route List index associated with this DSC
.		

6. Configure Avaya Network Routing Server

The Avaya Network Routing Server is configured through a http session to the Avaya Unified Communications Management. Open internet explorer and log into **Unified Communications Management**, then open the NRS named **NRSM on sps** as highlighted below.

NØRTEL	UNIFIED COMMUNICATI				<u>Help</u> <u>L</u>
Network Elements		rsion: 02.10.0029.01(3780)	User Name paulg	4/.165.92.19/	Base OS
- Security	5 <u>coresz.garcuab.com (member)</u>	Emax Booo			element.
Active Sessions — Tools	6 cores1.galctlab.com (member)	Linux Base	7.0	47.166.92.206	Base OS element.
Data	7 sps.galctlab.com (backup)	Linux Base	7.0	47.166.92.198	Base OS element.
	8 172.18.20.16	Media Gateway Controller	6.0	172.18.20.16	New element.
	9 172.18.20.17	Media Gateway Controller	6.0	172.18.20.17	New element.
	10 172.18.20.3	Media Gateway Controller	6.0	172.18.20.3	New element.
	11 172.18.20.15	Media Gateway Controller	7.0	172.18.20.15	New element.
	12 NRSM on sps	Network Routing Service	7.0	172.18.20.13	New element.
	13 NRSM on cores2	Network Routing Service	6.0	172.18.20.12	New element

Upon entering the NRS, click on **Endpoints** in the left hand pane and select **Standby database** at the top of the screen highlighted below.

N@RTEL	NETWORK ROUTING SERVICE MANAGER	<u>Help</u> L	
«UCM Network Services - System NRS Server Database System Wide Settings	Managing: Active database 172.18.20.13 Standby database Numbering Plans » Endpoints		^
- Numbering Plans Domains Endpoints Routes	Search for Endpoints Enter an endpoint ID (use * for all) and click Search. You may narrow the search by specifying a particular domain.		Hide
Network Post-Translation Collaborative Servers - Tools SIP Phone Context	Endpoint ID: * Limit results to Domain: All service domains V / All L1 domains V / All L0 domains V		1000
 Routing Tests H.323 SIP Backup 	Gateway Endpoints (7) User Endpoints (0)	Results per page: 50 👻 Search	
Restore GK/NRS Data upgrade	Add Delete SIP phone context	Refre	ish

Ensure the correct **Domain** is selected in the highlighted box below and click **Add** to add Logicall as a **Gateway Endpoint**.

«UCM Network Services								_
System NRS Server	Enter an endpoint ID (use * f	or all) and click Search.You	I may narrow the searc	h by specifying a particular	domain.			
Database System Wide Settings Endpoint ID: *								
Numbering Plans								
Domains	Limit results to Domain: dpp	, v 🗹 nortel	udp 🕑 /	cdp 😽				
Endpoints							. 1	
Routes						Results per pa	age: 50 🔽 Searc	h
Network Post-Translation								
Collaborative Servers	Gateway Endpoints (7) User Endpoin	ts (0)					
ools SIP Phone Context		,	.,					
- Routing Tests	Add Delete S	IP phone context					Refr	resh
H.323 SIP		Supported Protocols	SIP mode:	Call Signaling IP	Description	# of Routing Entries	Context	
Backup	1 ACodes	Static SIP endpoint	Proxy Mode	47.166.92.195	AudioCodes	1	dpp.nortel / udp / cdp	П
Restore	2 BCM450	RAS H.323 endpoint / Static SIP endpoint	Proxy Mode	Not available / 47.166.92.203		0	dpp.nortel / udp / cdp	
GK/NRS Data upgrade	3 BCM50	RAS H.323 endpoint /	Proxy Mode	Not available / 47,166,92,193		0	dpp.nortel / udp / cdp	

Enter a suitable **Endpoint name** and enter the IP address of the Logicall SIP stack in the **Static** endpoint address field. Ensure **SIP support** is set to **Static SIP endpoint** and that the **SIP UDP** transport enabled box is ticked as Logicall used UDP for SIP transport. Click **Save**.

End point name:	LogiCall	*	
Description:	LogiCall IVR	< >	
Trust Node:			
Tandem gateway endpoint name:	Not Applicable 💌		
Static endpoint address type:	IP version 4 💌	-	
Static endpoint address:	47.166.92.15		
H.323 support:	H.323 not supported	~	
SIP support:	Static SIP endpoint	~	
SIP mode:	 Proxy Mode Redirect Mode 		
SIP TCP transport enabled:			
SIP TCP port:	5060		
SIP UDP transport enabled:			
SIP UDP port:	5060		
SIP TLS transport enabled:			
SIP TLS port:	5061		
Persistent TCP support enabled:			
End to end security support:			
Network Connection Server enabled:			
quired value			

Gateway Endpoint dpp.nortel / udp / cdp)

Click on **Routes** in the left hand pane to add a routing entry for the new SIP endpoint and ensure this endpoint is selected under **Endpoint Name** as highlighted below. Click **Add** to add a new route.

NØRTEL	NETWORK ROUTING SERVICE MANAGER					
«UCM Network Services - System NRS Server Database	Managing: O Active database 172.18.20.13 Standby database Numbering Plans » Routes					
System Wide Settings Search for Routing Entries Domains						
Endpoints Routes Network Post-Translation Collaborative Servers - Tools SIP Phone Context - Routing Tests H.323 SIP Backup Restore	Enter a DnPrefix and Dn Type (use * for all) and click Search You may narrow the search by specifying a particular domain. DN Prefix: * DN Type: All DN Types Limit results to Domain: dpp.nortel / Judp / Cdp Endpoint Name: LogiCall					
GK/NRS Data upgrade	Routing Entries (2) Default Routes (0) Emergency Fallback Routes (0)					
	Add Copy Move Import Export Routing test Delete					
	DN Prefix DN Type Route Cost SIP URI Phone Context					

In the example below the Directory Number (DN) 6000 is associated with Logicall so when 6000 is called from a CS1000E deskphone the call is sent to the Logicall endpoint configured above. DN 6000 is configured as a Distant Steering Code (DSC) in a CDP (Coordinated Dialling Plan) on the CS1000E in Section 5.1, so ensure the DN type is set to Private level 0 regional (CDP steering code). The DN prefix is the DN associated with Logicall, in this case 6000, and the Route cost is set to 1. Click Save when finished.

DN type: Private level 0 regional (CDP steering code) 🛩	
DN prefix: 6000 *	
Route cost: 1 * (1-255)	

Once all configuration changes are made the database must be cutover to be made active. Click on database in the left hand plane as shown. Click on **Cut over** and then **Commit** both highlighted.

NORTEL NET	TWORK ROUTING SERVICE MANAGER	
«UCM Network Services - System NRS Server Database System Wide Settings Ormains Domains Domains Endpoints Routies Network Post-Translation Collaborative Servers - Tools SIP Phone Context Routing Tests H 323 SIP Bactup Restore GkINRS Data upgrade	Managing: 172.18.20.13 System > Database Database NRS uses a redundant database with Active and Standby copies. Normally changes are made to the state of th	andby database, tested, then cut over into active status. Cut over Revent Commun

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7. Configuration of Maximum Network Solutions Logical IVR

The Logicall IVR is preconfigured to specifications depending on the requirements of each solution. The configuration used to connect to the CS1000E is located in both the Control.ini and MYSIP.cfg files for each application on Logicall server. Logicall does not register with the NRS as a SIP endpoint so any calls made from the Logicall system are manually configured on the application to route directly to the SIP Gateway on the CS1000E. See the example below from the Control.ini file where a blind transfer and supervised transfer are setup in a menu service on a test application to ring extension number 2600 on the CS1000E.

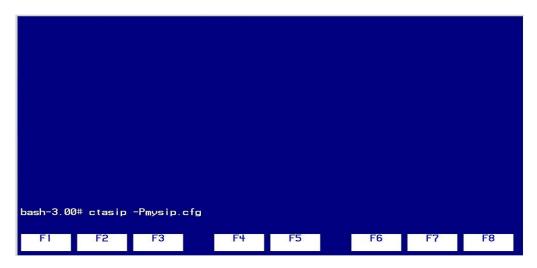
NOTE: 2600 is a directory number (DN) on the CS1000E and **47.166.92.207** is the node IP address of the CS1000E.

```
[BlindTransfer]
BlindTransferDefault=sip:2600@47.166.92.207:5060
[SupervisedTransfer]
SupervisedTransferOriginator1=5060
SupervisedTransfer1=sip:2600@47.166.92.207:5060
```

The Control.ini and MYSIP.cfg files used in the compliance testing are included in the **Appendix** of these Application Notes.

8. Verification Steps

To verify a successful configuration of Logicall and CS1000E, a call is placed from the CS1000E to the Logicall IVR application with the caller getting answered and successfully hearing clear and audible speech. A second call is made from the Logicall IVR to an extension on the CS1000E by opening a demo application on Logicall. Open a telnet session to the Logicall IVR software. This brings up the screen below.



Type **ctasip** -**Pmysip.cfg** as shown above which brings up a test application (as shown below) that allows a call to be generated. Type the following:

- 1. **op** (to open the port)
- 2. **sp** (to start the SIP)
- 3. **pc** (to place the call)
- 4. Type in the number to call out to (2600 in the example below)

bash-3.00# ctasip -Pmysip.cfg					
NMS-SIP Demo Program					
(c) NMS Communications, 2005					
Type 'h' for help. SWI: Connected DS0s: IUR(1) <-> MSPP(30) op ok					
Event: CTAEUN_OPEN_SERVICES_DONE, Finished sp Enter protocol name ['sip0']: ok Event: NCCEUN_START_PROTOCOL_DONE, CTA_REASON_FINISHED					
Event: NCCEUN_LINE_IN_SERVICE pc To ['sip:26000+7.166.92.198']: sip:26000+7.166.92.207 ok Event: NCCEUN_PLACING_CALL Event: NCCEUN_CALL_PROCEEDING Event: NCCEUN_PROTOCOL EVENT, NCC PROTEUN RING FALSE					
F1 F2 F3 F4 F5 F6 F7 F8					

If Logicall is receiving SIP messages from the CS1000E, these can be viewed as shown below. SIP messages can be seen only when a successful call is being made.

9. Conclusion

The interoperation of Logicall IVR from Maximum Network Solutions with Avaya Communication Server 1000E was successful for this specific setup requested by Maximum Network Solutions with some workarounds put in place in order to place calls from the Logicall IVR to the CS1000E. These workarounds are outlined in **Section 2.2**.

10. Additional References

Additional Avaya product documentation is available at <u>http://support.avaya.com</u>.

- [1] Software Input Output Reference Administration Avaya Communication Sever 1000, R7.0 NN43001-611, 05.09 Sept 2011
- [2] Network Routing Server Fundementals Avaya Communication Sever 1000, R7.0 NN43001-130

For information on Maximum Network Solutions (Maxnet) and product support visit <u>http://www.maxnet.co.uk</u>

Appendix

Patch version for Avaya Communication Server R7.0

VERSION 4121 RELEASE 7 ISSUE 00 Q + DepList 1: core Issue: 01 (created: 2011-05-17 10:40:44 (est)) ALTERED IN-SERVICE PEPS PAT# CR # PATCH REF # NAME DATE FILENAME SPECINS ISS1:10F1 p30683 1 01/12/2011 p30683 1.cpl 001 wi00849951 NO p30210_1.cpl p30210 1 019 wi00853778 ISS1:10F1 01/12/2011 NO 021 wi00728462 ISS1:10F1 p30346_1 01/12/2011 p30346_1.cpl NO p28647_1 022 wi00839793 ISS1:10F1 01/12/2011 p28647 1.cpl NO 023 wi00860278 ISS1:10F1 p30789 1 01/12/2011 p30789 1.cpl NO 024 wi00853798 ISS1:10F1 p30131 1 01/12/2011 p30131 1.cpl NO 025 wi00856984 ISS1:10F1 p17588 1 01/12/2011 p17588 1.cpl NO 01/12/2011 p30656_1.cpl 026 wi00847002 ISS1:10F1 p30656 1 NO 027 wi00647104 ISS2:1of1 p29747_2 01/12/2011 p29747_2.cpl NO 028 wi00865953 ISS1:10F1 p30832 1 01/12/2011 p30832_1.cpl NO 01/12/2011 029 wi00876852 ISS1:10F1 p30952_1 p30952_1.cpl NO wi00843647 p30186_1 p30186_1.cpl 030 ISS1:10F1 01/12/2011 NO p29938_1 p29732_1 p30223_1 031 wi00853750 ISS1:10F1 01/12/2011 p29938_1.cpl NO p29732_1.cpl p30223_1.cpl 032 wi00688477 ISS1:10F1 01/12/2011 NO 033 wi00686977 ISS1:10F1 01/12/2011 NO p30468_1 p30468 1.cpl 034 wi00825672 ISS1:10F1 01/12/2011 NO p30418 1 01/12/2011 035 ISS1:10F1 p30418 1.cpl wi00856244 NO p30826_1.cpl ISS1:10F1 p30826 1 01/12/2011 036 wi00865152 NO p30167 1 01/12/2011 p30167 1.cpl 037 wi00852325 ISS1:10F1 NO p30215 1 01/12/2011 p30215 1.cpl 038 wi00852304 ISS1:10F1 NO 039 wi00855276 ISS1:10F1 p29903 1 01/12/2011 p29903 1.cpl NO 040 wi00816794 ISS1:10F1 p30443 1 01/12/2011 p30443 1.cpl NO ISS1:10F1 041 wi00641909 p30004_1 01/12/2011 p30004_1.cpl NO 042 wi00824249 ISS1:10F1 p30447_1 01/12/2011 p30447_1.cpl NO ISS1:10F1 043 Q02150073-01 p30160_1 01/12/2011 p30160 1.cpl NO 044 wi00688048 ISS1:10F1 p25747_1 01/12/2011 p25747_1.cpl NO p30282_1 p30541_1 p29744_1 045 wi00853453 ISS1:10F1 01/12/2011 p30282_1.cpl NO p30541_1.cpl p29744_1.cpl 046 wi00833760 ISS1:10F1 01/12/2011 NO 047 wi00641671 ISS1:10F1 01/12/2011 NO p30596_1 p30596_1.cpl 048 wi00839645 ISS1:10F1 01/12/2011 NO p30807_1 p30807 1.cpl ISS1:10F1 01/12/2011 049 wi00862916 NO p30676_1 p30676_1.cpl 050 wi00845667 ISS1:10F1 01/12/2011 NO 051 wi00852317 ISS1:10F1 p30176 1 01/12/2011 p30176 1.cpl NO 052 wi00686889 ISS3:10F1 p30074_3 01/12/2011 p30074 3.cpl NO 01/12/2011 p30306 1.cpl 053 WI00853478 ISS1:10F1 p30306 1 NO 054 Q02147525 ISS1:10F1 p30072 1 01/12/2011 p30072 1.cpl NO 055 wi00855050 ISS1:10F1 p30731 1 01/12/2011 p30731 1.cpl YES ISS1:10F1 p30709⁻1 01/12/2011 056 WI00865566 p30709_1.cpl YES p30719_1 057 wi00879820 ISS1:10F1 01/12/2011 p30719_1.cpl NO ISS1:10F1 058 wi00688110 p30305_1 01/12/2011 p30305_1.cpl NO 059 wi00864908 ISS1:10F1 p30825_1 01/12/2011 p30825 1.cpl NO 060 wi00688114 ISS1:10F1 p30319_1 01/12/2011 p30319_1.cpl NO p30625_1 p30654_1 WI00853186 ISS1:10F1 01/12/2011 p30625 1.cpl 061 NO 01/12/2011 p30654 1.cpl 062 wi00869693 ISS1:10F1 NO p30787_1.cpl p30787_1 01/12/2011 063 wi00861072 ISS1:10F1 NO p30548_1 p30548 1.cpl 064 wi00834381 ISS1:10F1 01/12/2011 NO p30740 1.cpl p30740_1 01/12/2011 065 wi00853658 ISS1:10F1 NO 066 wi00853781 ISS1:10F1 p30416 1 01/12/2011 p30416 1.cpl NO 067 wi00688225 ISS1:10F1 p30295 1 01/12/2011 p30295 1.cpl NO

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0.00		T001 1001		01/10/0011	20056 1	220
068	wi00869468	ISS1:10F1	p30856_1	01/12/2011	p30856_1.cpl	NO
069	WI00836290	ISS1:10F1	p30554_1	01/12/2011	p30554_1.cpl	NO
070	wi00848801	ISS1:10F1	p30336_1	01/12/2011	p30336_1.cpl	YES
071	wi00865433	ISS1:10F1	p30828_1	01/12/2011	p30828_1.cpl	NO
072	WI00844778	ISS1:10F1	p30641_1	01/12/2011	p30641_1.cpl	NO
073	wi00848697	ISS1:10F1	p30621_1	01/12/2011	p30621_1.cpl	NO
074	wi00858461	ISS1:10F1	p30564_1	01/12/2011	p30564_1.cpl	YES
075	wi00848515	ISS1:10F1	p30677_1	01/12/2011	p30677_1.cpl	NO
076	wi00853388	ISS1:10F1	p30065_1	01/12/2011	p30065_1.cpl	NO
077	wi00853431	ISS1:10F1	p29935_1	01/12/2011	p29935_1.cpl	NO
078	wi00604003	ISS1:10F1	p29726_1	01/12/2011	p29726_1.cpl	NO
079	wi00874210	ISS1:1 OF 1	p30880_1	01/12/2011	p30880_1.cpl	NO
080	WI00851975	ISS1:10F1	p30312_1	01/12/2011	p30312_1.cpl	NO
081	wi00853837	ISS1:10F1	p30172_1	01/12/2011	p30172_1.cpl	NO
082	wi00688037	ISS2:10F1	p29376_2	01/12/2011	p29376_2.cpl	NO
083	WI00853745	ISS1:10F1	p29841_1	01/12/2011	p29841_1.cpl	YES
084	wi00861414	ISS1:10F1	p30791_1	01/12/2011	p30791_1.cpl	NO
085	wi00862909	ISS1:10F1	p30809_1	01/12/2011	p30809_1.cpl	NO
086	wi00839916	ISS1:10F1	p30593_1	01/12/2011	p30593_1.cpl	NO
087	wi00857960	ISS1:10F1	p30768_1	01/12/2011	p30768_1.cpl	NO
088	wi00857493	ISS1:10F1	p30766_1	01/12/2011	p30766_1.cpl	NO
089	WI00827391	ISS1:10F1	p30477_1	01/12/2011	p30477_1.cpl	NO
090	wi00687324	ISS1:10F1	p16376_1	01/12/2011	p16376_1.cpl	NO
091	wi00826074	ISS1:10F1	p30452_1	01/12/2011	p30452_1.cpl	NO
092	wi00852510	ISS1:10F1	p30357_1	01/12/2011	p30357_1.cpl	NO
093	WI00836333	ISS1:10F1	p30481_1	01/12/2011	p30481_1.cpl	NO
094	wi00868063	ISS1:10F1	p30848_1	01/12/2011	p30848_1.cpl	NO
095	wi00836181	ISS1:10F1	p30450_1	01/12/2011	p30450_1.cpl	NO
096	wi00856160	ISS1:10F1	p30750_1	01/12/2011	p30750_1.cpl	NO
097	wi00827512	ISS1:10F1	p30479_1	01/12/2011	p30479_1.cpl	NO
098	wi00686928	ISS2:1of1	p29899_2	01/12/2011	p29899_2.cpl	NO
099	wi00701008	ISS2:10F1	p30381_2	01/12/2011	p30381_2.cpl	NO
100	wi00837793	ISS1:10F1	p30573_1	01/12/2011	p30573_1.cpl	NO
101	wi00688204	ISS1:10F1	p30197_1	01/12/2011	p30197_1.cpl	NO
102	wi00854255	ISS1:10F1	p30124_1	01/12/2011	p30124_1.cpl	NO
103	wi00880384	ISS1:10F1	p30977_1	01/12/2011	p30977_1.cpl	NO
104	wi00824288	ISS1:10F1	p30461 1	01/12/2011	p30461 1.cpl	NO
105	wi00843569	ISS1:10F1	p30627_1	01/12/2011	p30627 1.cpl	NO
106	wi00828961	ISS2:10F1	p30492_2	01/12/2011	p30492 2.cpl	NO
107	wi00853753	ISS1:10F1	p30064_1	01/12/2011	p30064 1.cpl	NO

Maxnet Logicall Control.ini

[General] LogFile=/export/logicall/avaya_test_com/AvayaComTest.log Speech_Main_Path=/export/logicall/avaya_test_com/wav/main/ [BlindTransfer] BlindTransferDefault=sip:2600@47.166.92.207:5060 [SupervisedTransfer] SupervisedTransferOriginator1=5060 SupervisedTransfer1=sip:2600@47.166.92.207:5060 [Testing] TestNos Enabled=no TestCLI=0297949600 TestDNIS=2907 [SpeechResources] RTPDestIpAddress=47.166.92.15 RTPAltDestIpAddress=N/A RTPSrcIpAddress=47.166.92.16 RTPSrcPort ASR=2002 RTPSrcPort_TTS=2100 [CmdRequestListener] CmdReqListenerPort=30007 * Trace to File - Set to FALSE - Print Screen. * Debug Notes - Trace_Status - Should be FALSE or TRUE *
* Debug Notes - Trace Level - DEBUG, ERROR, EXTENDED * [Debug] TraceToFile=/export/logicall/avaya test com/AvayaComTest.log TraceLevel=EXTENDED

Maxnet Logicall MYSIP.cfg

#_____ General Demo Parameters # "general.board = 0 # NMS board number to use with ADI/VCE
general.slot = 1 # NMS timeslot to use with ADI/VCE and SIP-NCC general.stream=general.protocol=general.autoStart=0 general.autoRelease = 0 general.autoSDP = 1 SIP Parameters #-----sip.from = sip:6000@47.166.92.15:5060 sip.registrar sip.contact = #-----_____ SIP-SDP Parameters #sip.sdp.connection.networkType = IN sip.sdp.connection.addressType = IP4 sip.sdp.connection.address = 47.166.92.16 # IP address of CG board as sip.sdp.connection.port = 8004 # IP address of CG board as # configured in CG cfg file sip.sdp.origin.userName = nmsSip sip.sdp.origin.sessionId = 01234567890 sip.sdp.origin.networkType = IN sip.sdp.origin.addressType = IP4 sip.sdp.origin.address = 47.166.92.16 # IP address of CG board sip.auth.user = sip.auth.user sip.auth.password _____ MSPP Parameters -- Use Fusion, not HMP # #-----mspp.hmp = 0 # Do not use HMP => use Fusion since # nomedia (next parameter) is false. = 0 # There is media used in the application mspp.nomedia = 30 # NMS Time slot used by CG based Fusion mspp.slot # for DS0 endpoint. MUST be differen from
timeslot specified by 'general.slot' _____ Voice Play Parameters # _____ #----voice.play.file = play.vox voice.play.type = 2 # 0 means VCE_FILETYPE_VOX voice.play.encoding = 10 # 2 means NMS_24 Voice Record Parameters # #----voice.record.file = record.vox voice.record.type = 0 voice.record.encoding = 2 # 2 means NMS_24, consistent with shipped # fusion configuration file

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