

SIP Trunk Compatibility Report

NEC is pleased to verify that:

BT Wholesale – WSIPT & One Voice Services

Has successfully met the standards for SIP Trunk compatibility with the NEC products listed below.

SV9100

NEC Enterprise Solutions has performed Interoperability Testing with the Switch(es) and Provider/Service listed above on the date specified.

Please always refer to the appropriate SIP Trunk Compatibility Report Index and the latest edition of a specific Compatibility Report on BusinessNet before considering connection.

If a Provider is no longer mentioned in the Index then the Compatibility Report has been withdrawn and connection will no longer be supported by NEC Enterprise Solutions.

Test Completion Date: 27/03/2015

Test Location: NEC UK

Name of Provider: BT Wholesale

Website: www.btwholesale.com

System Tested: SV9100 Software Version: V2.00.59

SIP Connection Mode: Networking Mode

Test Plan Version: 5.

Please refer to the following page(s) for further Information and Configuration Notes.

SIP Configuration Notes - BT Wholesale - WSIPT & One Voice Services

Use the specific configuration guide below as an example to configure an SV9100 PBX for connection to the service described above via SIP trunks.

Recommended Software Versions

SV9100:



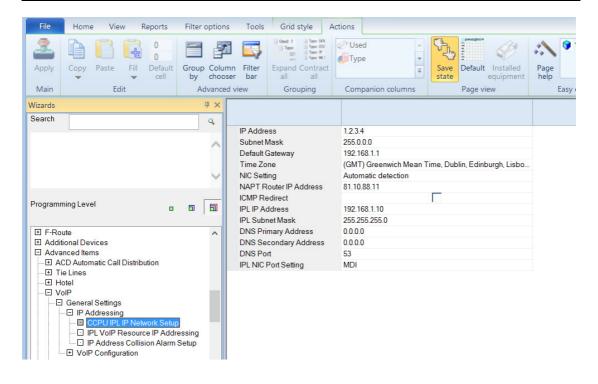
CCPU v.4.00.50

System Programming

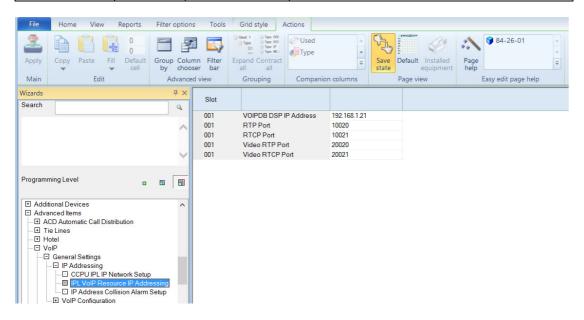
The following items should be changed – all other items are considered irrelevant and as such left as default. Screenshots are for example purposes only and will have been taken from the PBX under test but will apply to the other PBXs listed on the cover of the Compatibility Report.

Any differences in programming will be documented where necessary.

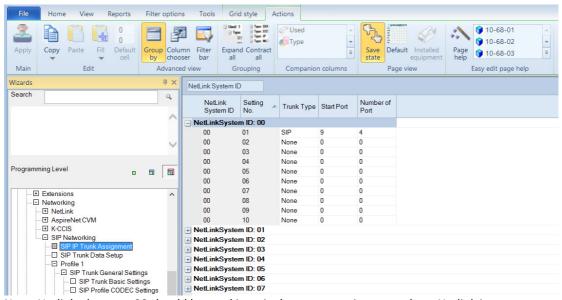
Advanced Edit	PRG	Item	Setting
	10-12-09 IPLE/VOIPDB IP		Set according to customers network
		Address	requirements
	10-12-10	IPLE/VOIPDB	Set according to customers network
Advanced		Subnet Mask	requirements
Items > VoIP >	10-12-03	Default	Set according to customers network
General		Gateway	requirements
Settings >	10-12-02	Default	Set according to customers network
CCPU IP		Gateway Subnet	requirements
Address		Mask	
	10-12-01	IP Address	Must be in a different network range to IPLE IP
			Address (10-12-09) Must not be set as 0.0.0.0



Advanced Edit	PRG	Item	Setting
Advanced	84-26-01	VoIP Gateway IP	Set according to customers network
Items > VoIP >		Address	SV9100 requirements. IPLE requires 1 x static IP
IP Addressing			address for the DSP resources.
>VoIP Resource			
IP Address			

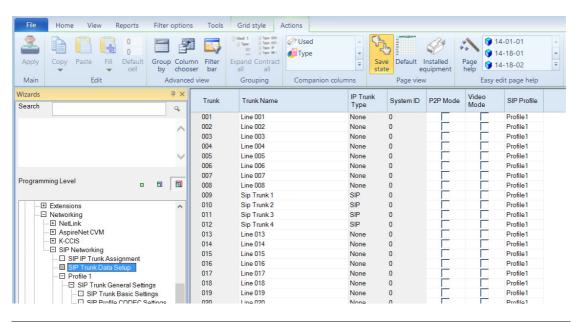


Advanced Edit	PRG	Item	Setting
Advanced Items	10-68-01	Trunk Type	Set to SIP
> VoIP			
>Networking>SIP	10-68-02	SIP Trunk Start	Enter the Trunk starting port. This may vary
Networking > SIP		Point	depending on other Trunk cards installed and
IP Trunk			must be from an un-allocated port.
Assignment	10-68-03	Number of	Enter the number of SIP Trunks required.
_		ports	

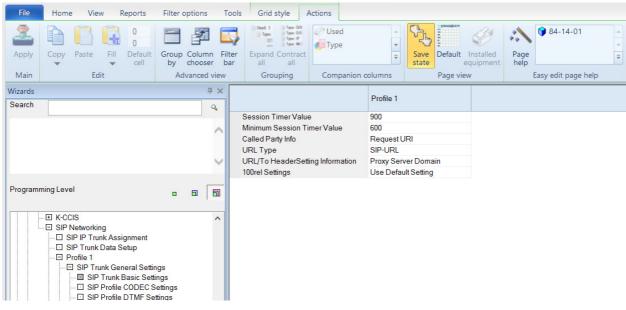


Note: Netlinked system 00 should be used in a single system environment where Netlink is not activated. If Netlink has been deployed then use netlinked system ID 01 to 50 depending where the SIP Trunks are connected.

Advanced Edit	PRG	Item	Setting
Advanced Items	14-18-01	Trunk Type	Should be set to SIP
> VoIP			
>Networking>SIP	14-18-02	System ID	Should match the netlink system ID as per 10-68
Networking > SIP			
Trunk Data Set	14-18-05	SIP Profile	Set to Profile 1 or alternative profile if multiple
Up			SIP Carriers are required.

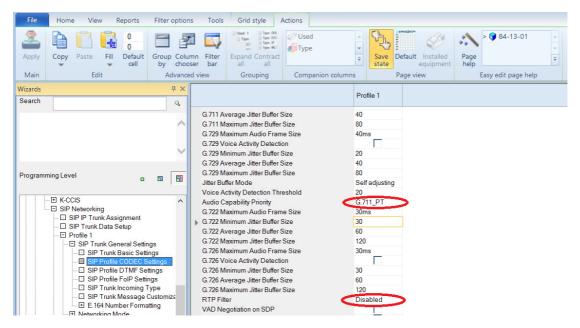


Advanced Edit	PRG	Item	Setting
Advanced Items >	84-14-07	Session timer	Set to 900
VoIP		value	
>Networking>SIP	84-14-08	Minimum	Leave set as 600
Networking >		session timer	
Profile 1>SIP Trunk		value	
General	84-14-09	Called Party	Leave set as Request URI
Settings>SIP Trunk		Info	
Basic Settings	84-14-15	100rel	Use Default Setting
		Settings	

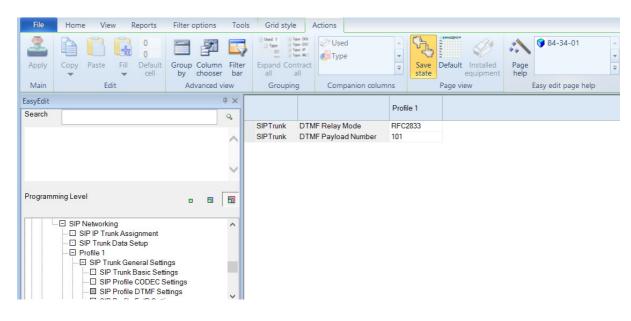


Page 4 of 12

Advanced Edit	PRG	Item	Setting
Advanced Items >	84-13-28	Audio	Set to customer's requirements (G.711 or
VoIP		Capability	G.729only)
>Networking>SIP		Priority	
Networking > Profile	84-13-49	RTP Filter	Set to disabled
1>SIP Trunk General			
Settings>SIP Profile			
Codec Settings			

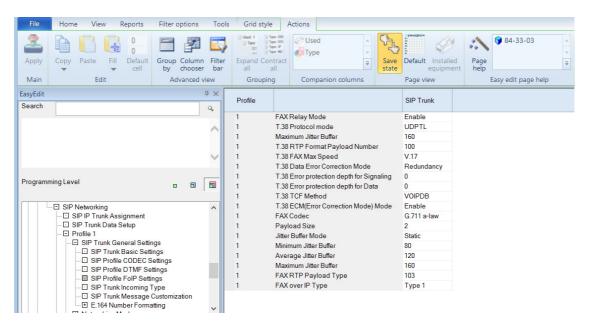


Advanced Edit	PRG	Item	Setting
Advanced Items > VoIP	84-34-01	DTMF	Set to RFC2833
>Networking>SIP		Relay	
Networking > Profile		Mode	
1>SIP Trunk General	84-34-02	DTMF	Set as 101
Settings>SIP Profile		Payload	
DTMF Settings		Size	

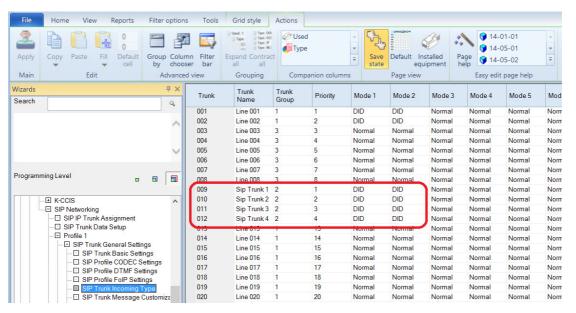


Page 5 of 12

Advanced Edit	PRG	Item	Setting
Advanced Items >	84-33-01	Fax Relay	Enable if required to support T.38 Fax
VoIP		Mode	
>Networking>SIP			
Networking > Profile	84-33-02	T.38	Set to UDPTL
1>SIP Trunk General		Protocol	
Settings>SIP Profile		Mode	
FoIP Settings			



Advanced Edit	PRG	Item	Setting
Advanced Items > VoIP	14-01-01	Trunk Name	Label Trunks
>Networking>SIP Networking > Profile 1>SIP Trunk General	14-05-01	Trunk Group/Priority	Assign to Trunk Groups and set priority. You must remove all unused Trunks from the group selected.
Settings>SIP Trunk Incoming Type	22-02	Incoming Call Trunk Setup	Set to DID

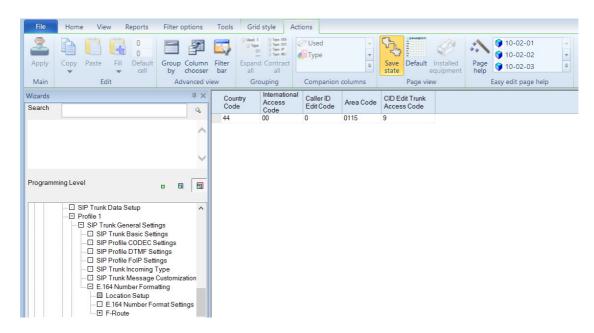


Page 6 of 12

e.164 Number Formatting Support

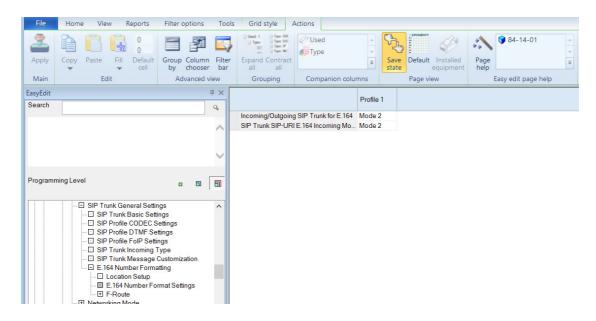
BT Wholesale require that the caller ID presentation to be converted as e.164 format +44 for example. Please configure as below to meet the requirements as addition to the settings above for commands 84-14-13 & 84-14-16 for SIP Profile 1.

Advanced Edit	PRG	Item	Setting
Trunks>Location Set	10-02-01	Country	Enter the country code as 44 for UK
Up		Code	
	10-02-02	International	Enter as 00
		Access Code	
	10-02-03	Caller ID Edit	Enter 0 if required
		Mode	
	10-02-05	CID Edit TAC	Enter the digit used for Trunk Access –
			Normally 9 or 0



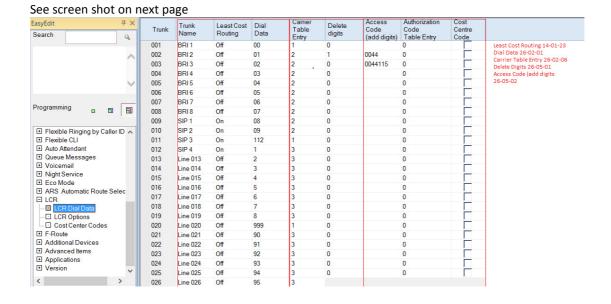
Advanced Edit	PRG	Item	Setting
Advanced Items > VoIP >Networking>SIP	84-14-13	Incoming/Outgo ing SIP Trunk for	Set to Mode 2
Networking > Profile 1>SIP Trunk General		E.164	
Settings>E.164 Number Formatting>E.164 Number Format Settings	84-14-16	SIP Trunk SIP- URI E.164 Incoming Mode	Set to Mode 2

Continues on next page



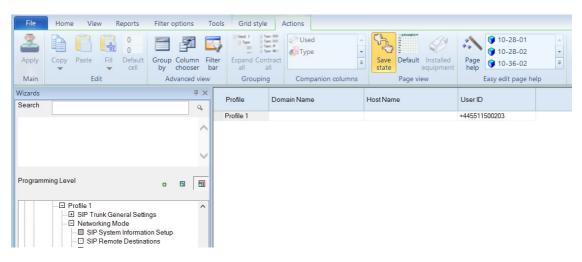
Also configure LCR to convert the dialed digits to add 0044 infront of the number so the customer can still dial using the national format when dialing. LCR also converts numbers dialed from the Speed Dial tables & Caller History from key sets. Please set LCR as per the table below and enable Trunks required.

Advanced Edit	PRG	Item	Setting
	14-01-23	Least Cost	Enable each required Trunk (SIP Trunk
		Routing	only)
	26-02-01	Dial Data	Enter digits to be modified. See table
			below and add further digits as required.
Advanced Items >	26-02-06	Carrier Table	Enter the required Carrier Table for the
LCR>LCR Dial Data			modification for each dialled digit.
	26-05-01	LCR Delete Digits	Enter the amount of digits required for
			deletion per carrier table. See table below
	26-05-02	Access Code	Use this table to add digits per carrier
			table. See table below.

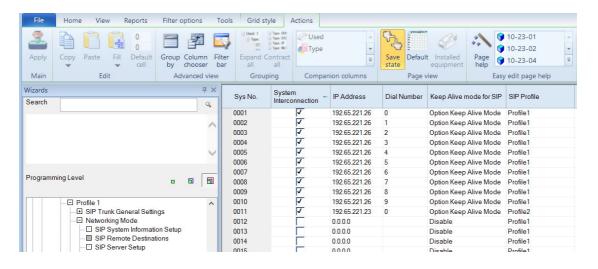


Page 8 of 12

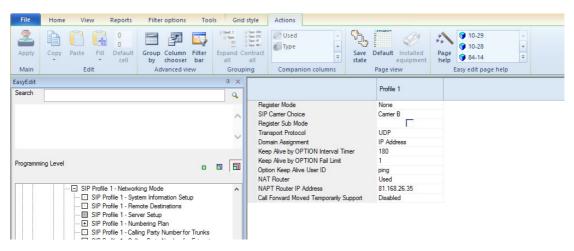
Advanced Edit	PRG	Item	Setting
Advanced Items > VoIP >Networking>SIP	10-28-01	Domain Name	Set according to customers network requirements or leave blank if not required
Networking > Profile 1>Networking Mode>SIP System Information Setup	10-28-02	Host Name	Set according to customers network requirements or leave blank if not required
	10-36-02	User ID	Enter the user ID supplied by BT Wholesale (normally the main number that must be entered in international format)



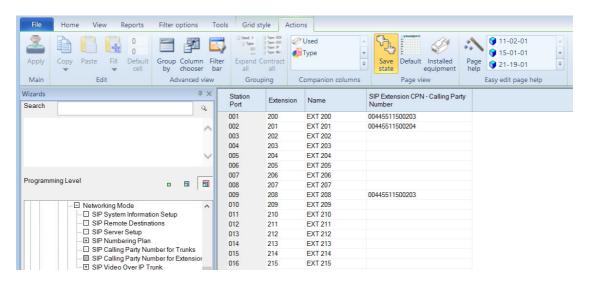
Advanced Edit	PRG	Item	Setting
Advanced Items > VoIP	10-23-01	System	Set to Enabled
		Interconnection	
	10-23-02	IP Address	Set to SIP server IP Address supplied by BT
			Wholesale
>Networking>SIP Networking >	10-23-04	Dial Number	The first digit(s) that will be dialled
Profile	10-23-05	Keep Alive Mode	Set to option keep alive mode required for
1>Networking			SIP Proxy Fall back to profile 2. See
Mode>SIP Remote			comments for further information on this
Destinations			requirement.
	10-23-06	SIP Profile	Set to Profile 1 or alternative profile if
			multiple SIP Carriers are required.



Advanced Edit	PRG	Item	Setting
	10-28-05	Domain Assignment	Set to IP Address
	84-14-17	Call Forward Moved	Not supported. Use standard
		Temporarily Support	forwarding methods if external divert
Advanced Items >			is required
VoIP	10-29-05	SIP Carrier Choice	Set to carrier format B (See comments
>Networking>SIP			for more information why this format
Networking > Profile 1>Networking Mode>SIP Server			is required).
	10-29-16	Register Sub Mode	Set to disabled
	10-12-06	NAPT Router	Enable if required
Setup	10-29-21	NAPT Router IP	Enable if Router if configured for port
		Address	forwarding to modify the contact
			header from private to public IP
			address



Advanced Edit	PRG	Item	Setting
Advanced Items > VoIP	21-19	SIP Extension	Enter the full number including the
>Networking>SIP		CPN Number	area code
Networking > Profile 1>			
Carrier Mode (Domain			
Name)>SIP Calling Party			
Number for Extensions			



Network Configuration

If Public IP addresses are assigned to the SV9100 IPLE and VoIP Gateways, then there should be no network configuration required.

If there is one public IP address assigned, and NAT is used, it is necessary to configure Port Forwarding on the router and we recommend that SIP ALG is disabled:

- Port 5060 should be forwarded to the IPLE IP address as per command 10-12-09. If using
 multiple SIP Carriers then port 5062 should be used for the second profile in command 8414-06. BT Wholesale WSIPT & One Voice Services has not been tested in this mode. SIP
 Carrier does not need to change their port as 5060 is still correct.
- Port 10020 10533 should be forwarded to the VoIP Gateway IP addresses as per command 84-26-01
- NAT Router should be enabled in PRG10-29-21 and the Public IP address should be entered into PRG10-12-07.

Licensing

The SV9100 system should be licensed for each SIP Trunk required. Four SIP Trunk licenses are included with new SV9100 systems. Set command 10-54-01 (Blade License Set Up) for Slot GCD-CP10 to the amount of DSP Licenses as per the feature activation page as per code 5103 (don't include the 8 that comes as default with IPLE).

Known Limitations/Comments

- SIP calls are sent "en bloc". This means that the External Call Interdigit timer (PRG21-01-03) must expire before the call is set up. This can be reduced, but will have an impact on ISDN trunks also. The user can dial # to indicate "end of dialling" instead if required.
- Command 20-25-02 to provide ring back tone for SIP to SIP calls.
- BT SIP Accounts have a WEB portal where the SIP Settings can be amended as required. You will need to change the Caller ID presentation to use the From Header instead of the P-Asserted Identification field (default setting). This is required if the customer requires flexible CLIP using command 21-17 (Trunk Level) or 21-19 (Extension Level) to send an alternative CLI for outbound calls other than the base number. CLIP should be entered in international format 00441923610000 for example.
- SIP Carrier B format in 10-29-14 must be used for the BT WSIPT service as this format does
 not send the Privacy header that can override settings that can be applied using the BT
 web portal for example withheld CLIP at network level. Note comment above.
- Fax over SIP Trunks has been tested using T.38 for both inbound and outbound. Please note
 that we cannot always guarantee fax over IP as this can fail due to the Fax Machines used or
 bandwidth / packet loss on the network. Command 15-03-03 must be set to special & 15-03-18
 set to Fax for the SLT extensions used for Fax and Fax Relay Mode must be enabled in 84-13-50.
- Minimum and maximum session expiry timers need to be set as 600 for the minimum expiry timer (84-14-08) & 900 for the maximum (84-14-07) as required by BT Wholesale.
- Call Forward Moved Temporally with T302 return not supported with BT. If Call defection via SIP is required then Trunk to Trunk external forwarding will need to be configured.
- BT Wholesale provides a secondary SIP Server IP Address that can be used for back-up if their
 primary connection fails also referred to as SIP Proxy Fallback. This feature is not currently
 supported as the delay for fall-back is too long. We are looking into supporting this at a later
 date when the fall-back timer is more flexible but for now if required command 10-23-02
 could be amended to their back up SIP Server address if the primary SIP Server not
 contactable.

NEC Enterprise Solutions cannot be held responsible for any regulatory non-compliance resulting from the use of this service.

Document History

Version	Date	Description
5.0	17/04/2015	Certification tests were completed using BT Wholesales SIP Certification Plan. This report is now released for the SV9100.
5.1	28/04/2015	Added comment regarding SIP Proxy Fallback
5.2	06/01/2016	Updated certification regarding Carrier Format B to be used instead of E so that the privacy header is not sent in the invite for outgoing calls. Comments added that the BT SIP Account will need to be updated to use the From Header instead of the PAID header for outgoing CLIP requirements.

TO THE EXTENT NOT PROHIBITED BY LAW, IN NO EVENT SHALL NEC ENTERPRISE SOLUTIONS BE LIABLE FOR PERSONAL INJURY OR ANY INCIDENTAL, SPECIAL, DIRECT, INDIRECT OR CONSEQUENTIAL **INCLUDING** DAMAGES WHATSOEVER, WITHOUT LIMITATION, DAMAGES FOR LOSS OF PROFITS. LOSS OF DATA. BUSINESS INTERRUPTION OR ANY OTHER COMMERCIAL DAMAGES OR LOSSES ARISING OUT OF OR RELATED TO YOUR USE OF THIS REPORT, HOWEVER CAUSED REGARDLESS OF THE THEORY OF LIABILITY (CONTRACT, TORT OR OTHERWISE) AND EVEN IF NEC ENTERPRISE SOLUTIONS HAS BEEN ADVISED OF THE POSSIBILITY OF SUCH DAMAGES, DUE TO THE POSSIBILTY OF PROVIDERS MAKING SYSTEM CHANGES WHICH PROCEED THE COMPATIBILITY TESTING DETAILED IN THIS REPORT.

IN NO EVENT SHALL NEC ENTERPRISE SOLUTIONS TOTAL LIABILITY TO YOU FOR ALL DAMAGES (OTHER THAN AS MAY BE REQUIRED BY APPLICABLE LAW IN CASES INVOLVING PERSONAL INJURY) EXCEED THE AMOUNT OF SEVENTY FIVE POUNDS (£75).

THE FOREGOING LIMITATIONS WILL APPLY EVEN IF THE ABOVE STATED REMEDY FAILS OF ITS ESSENTIAL PURPOSE.