

Spectrum Enterprise SIP Trunking Service 3CX Phone System Release 12.5 IP PBX Configuration Guide

About Spectrum Enterprise:

Spectrum Enterprise is a division of Charter Communications following a merger with Time Warner Cable and acquisition of Bright House Networks. Spectrum Enterprise is a national provider of scalable, fiber technology solutions. The Spectrum Enterprise portfolio includes networking and managed services solutions, including Internet access, Ethernet and Managed Network Services, Voice, TV and Cloud solutions. Our industry-leading team of experts works closely with clients to achieve greater business success.

About this document:

Spectrum Enterprise assures IP PBX compatibility by conducting interoperability testing to ensure any potential compatibility issues have been resolved prior to installation. Please review the IP PBX configuration instructions in this guide prior to your installation date.

Be advised that this document may contain references to Time Warner Cable Business Class. All references to Time Warner Cable Business Class, TWCBC or TWC should be read as Spectrum Enterprise.

Thank you,

Spectrum Enterprise

Document Purpose and Target Audience

This document will serve as a reference guide to configure the 3CX Phone System IP PBX to interoperate with Time Warner Cable (TWC) SIP Trunk Service.

This guide is not intended to be a replacement of the PBX manufacture's user or configuration guide. It is intended to provide additional guidance on configuring the PBX in preparation to receive voice service from the SIP Trunk. It provides detailed instructions and best practices for a successful installation with TWC SIP Trunks.

There are many options for establishing and maintaining service using the 3CX Phone System Series. This guide focuses on the minimum configurations essential for successful interoperability with Time Warner Cable Business Class SIP Trunks.

This configuration guide is based on:

Customer Premise Equipment:

Model	3CX Phone System
Release	12.5

TWC Network Equipment:

ESG	InnoMedia ESBC 9378-4B

SIP Trunk Components

The Time Warner Cable Business Class (TWCBC) SIP Trunks product is an IP-based, voice only trunk that uses Session Initiation Protocol (SIP) to connect an IP PBX to the PSTN. The IP PBX uses SIP to exchange signaling information with the service provider and to deliver and receive voice in IP packets.

The IP PBX is connected to the TWC Enterprise SIP Gateway (ESG), which provides network access for voice traffic. The customer is responsible for the LAN infrastructure and configuration, including the physical connection to the LAN port 2 on the ESG.

The ESG is the demarcation point to the TWC network. The ESG is connected to a dedicated router for SIP Trunks delivered over a fiber connection or to a cable modem when delivered over a DOCSIS connection.

SIP Trunk components located on the customer premise, including connections to the TWC network, are illustrated below.



All TWC SIP Trunk calls are routed over Time Warner Cable's IP network and are not routed over the public internet.

Getting Started

You will need to have the TWC "SIP Trunk Questionnaire" and "Business Class (BC) SIP Trunks: Customer Cut Sheet" in order to configure your IP PBX for TWC Business Class SIP Trunk service.

Confirm that your **IP PBX model number and software versions** recorded on the **Customer Cut Sheet** match those associated with your current equipment. If they do not, be sure to alert your TWC sales engineer or TWC project manager as this can impact how TWC designs your service configuration.

Example from Customer Cut Sheet for Cisco UC 560:

	SERVICE INFORMATION
PRODUCT	Business Class SIP TRUNK
IP-PBX MAKE	Cisco
IP-PBX MODEL	UC560
IP-PBX SOFTWARE VERSION	15.1(4)

While configuring your IP PBX for BC SIP Trunk service, you will need to know your Lead Telephone Number and the IP address of your IP PBX.

The Lead Number is confirmed on the Customer Cut Sheet as seen below:

			Trunk Groups	
TWC TRUNK Group ID	DID Range	Lead Number	Inbound Call Blocking	Outbound Call Blocking

The **IP Address** of the IP PBX was recorded on the **SIP Trunk Questionnaire**, Section 5. Signaling and Media as shown below:

5- Sig	naling and Media	leinen in der seinen s
IP Address for PBX or SBC	IP: xxx.xxx.xxx.xxx	
To setup LAN configuration for signaling of voice traffic to the ESG	Subnet: 255.255.xxx.xxx	
		TWC could provide IP address

This document is intended as an aid to help configure a customer's IP PBX for interoperability with TWCBC SIP Trunk Service.

3CX Configurations

The instructions provided in this section are intended to help the customer configure the 3CX PBX to connect to the ESG. They are not intended for advanced functionality setups. It is further assumed that the customer already have knowledge of 3CX operations.

Once logged into the 3CX GUI as an Administrator, follow these steps to configure SIP Trunk Service.

Navigate your web browser to the IP address of your 3CX SIP PBX.

- 1. Create Extension
- 2. Create SIP Trunk

Create a 3CX Extension

1. From your 3CX Administration console, click Add Extension. Referring to Figure 1

3CX Phone System Manage X				_ D ×
$\epsilon \rightarrow C$ [] localhost/man	agement/MainForm.wgx			= (~>
	agemeny mann orm.wgx			A
3CX Server	Manager 🕨 Add Extension		Manage Add View	Settings Help
 Server Ports/Trunks Status Extension Status System Extensions Status CXPhone Clients Remote Connections Phones Server Activity Log Server Event Log Server Event Log Server Event Log Services status Extensions WebRTC Gateway WolP/PSTN Gateways VolP/PSTN Gateways VolP/PSTN Gateways VolP/PSTN Gateways VolP/PSTN Gateways OutBound Rules Digital Receptionist Ring Groups Call Queues Fax Machines Settings Updates Links 	Manager > Add Extension Carbon Settings and click OK or Apply to save of General Voice Mail Forwarding Rules Phone Provis User Information Configure user information below Extension Number First Name Last Name Email address Mobile Number Authentication The authentication ID and Password are used by the p ID Password	hanges. ioning 3CXPhone Other Op 669 Name Last Name Last Name 669 50pw980	Manage Add View	Settings Help
> 😰 Help				
	4			Þ

Figure 1 Add Extension

Figure 2 Adding a new extension

2. Click **Apply** or **OK** button to save the configuration.

Create a VoIP Provider

1. Navigate to VOIP Providers.





- 2. Click the Add Provider button to create a new VOIP Provider.
- Under Name of Provider, enter TWC SIP Trunk (or whatever name you want). Select "Generic SIP Trunk"

OCV	VOIP Providers				
3CX	Add VOIP Provider Wizard				
- > 3CX Phone System	Add VOIP Provider Wizard				
Ports/Trunks Status Status Extension Status	Name of Provider	eSi	BC	0	
System Extensions Status Remote Connections Phones	Choose a Provider:				
Server Activity Log		0	3	3StarsNet	BE/NL
Services status		0		Actio.pl	PL
Extensions		0	0	Broadvox Fusion (IP Based)	US
STN devices		0	3	Broadvox Fusion (Register)	US
OIP Providers		0	C	CallCentric	US
Inbound Rules Reidese		0	G	Chevond	US
OutBound Rules		0	110	Cellip	SE
3 Digital Receptionist		0		FasyCall	GR
- 33 Ring Groups		0	0	Engin	ALL
- 3 ³³ Call Queues		0	4	G7Eleven	TE
Fax Machines				Ganaria SIP Tarrela	IL I
Securitys Securitys Securitys Security S				Canada Valla Bassidas	
E 👷 Links		0		Generic Volir Provider	
🗄 🤄 Help		0		Inmonex	worldwide
		0	1	IDDI	FR
		O	-	Masquevoz	ES
		0	N	nettel	DK
		0	9	Nexvortex	US



4. Click **Next**, Under VOIP Providers Details, enter the SIP server IP address as ESG LAN IP address. Note: Please check the SIP account information we send you, the SIP Server or IP address will be different from the IP address below. It is just an example

SIP server hostname or IP	172.16.1.20	0	
SIP Server port	5060	0	
Outbound proxy hostname or IP	172.16.1.20	0	
Outbound proxy port (default is 5060)	5060	0	

Figure 5 Setting SIP Server Address

5. Click **Next**. Enter your SIP account information here according to SIP UA account of ESG. Enter DID (the number we assigned to you) as External Number, Enter authentication ID, Enter Authentication Password. Enter the maximum simultaneous call. The number should be matching our system setting.

External Number	2404983506	0	
Authentication ID (aka SIP User ID)	2404983506		
Authentication Password	•••••	0	
3 Way Authentication ID			
nultaneous Calls			
Maximum simultaneous calls	10	0	

Figure 6 Configuring SIP account

6. Click Next. Here you are required to setup the behavior of 3CX when receiving SIP trunk incoming call. You can connect the call to certain extension, or you can connect to Digital Receptionist (Auto Attendant), provided that you already have recorded the voice message. For initial testing purpose, we recommend you to connect the call to extension, so you can test the incoming call after setup.

Office Hours			
Configure where calls should be routed during office hours.			
End Call			
Connect to Extension	3506	*	0
Connect to Queue / Ring Group		~	0
Connect to Digital Receptionist		~	0
Voicemail box for Extension	3506	*	0
Forward to Outside Number			0
Send fax to email of extension	email of extension 8888	~	0
Same as Out of Office hours			
			< Back Next >

Figure 7 Assign DID to an extension

7. Click Next. Here you need to setup the outgoing call behavior. In general, to distinguish the internal call between extension and outgoing call to outside number, you can setup a prefix so 3CX know how to route the call through SIP trunk. For example, you can add Calls to numbers starting with Prefix with "9". When you want to dial out from extension, simply dial 9+10 digital number you want to dial.

🤣 Create an Outbound Call Rule to configure on which PSTN port, VOIP provider or bridge an outbound calls should be placed on

Dude Menne						
Rule Name			R	ule for eSBC	0	
pply this rule to th	hese calls					
fine to which out	tbound calls the	e rule must apply				
Calls to number	rs starting with	(Prefix)	9		0	
Calls from exte	nsion(s)		Ē		0	
Calls to Number	rs with a lengt	n of	Ē		0	
Calls from exte	nsion group			Select	2	
ake outbound call	ls on					
ake outbound cal nfigure up to 3 r	ls on outes for calls.	The second and third	route will be used as	backup. For each	route, digits can be stripp	ed or added.
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ake outbound cal onfigure up to 3 m Route Route	ls on outes for calls. 1 2	The second and third	route will be used as	backup. For each Strip Digits 1 V	route, digits can be stripp Prepend	ed or added.
lake outbound cal onfigure up to 3 r. Route Route Route	ls on outes for calls. 1 2 3	The second and third	route will be used as	Strip Digits	route, digits can be stripp Prepend	ed or added.

Figure 8 Configuring Outbound Call Rule

8. Click **Finish** to compete the initial setup. You can observe if the trunk or extension is registering correctly by looking at Port/Trunks Status or Extension Status.

2	Ports/Trunks Status						
JUX	🞇 Disconnect Cal						
JCK Phone System Ports/Trunks Status Status System Extension Status System Extensions Status System Extensions Status Server Activity Log Server Activity Log Server Status Server Status	Status Connected Registered (die) Registered (die)	Virtual Extension Number 10000 10001 10002	Type Provider Provider Provider	Name e58C-3506 e58C-3507 e58C-3508	OUT	Caler ID 2404983506 2404983507 2404983508	Desthation 3506 3507 3508

Figure 9 Ports/Trunk Status Screen That completes the configuration of adding ESG SIP Trunk to 3CX.

TWC Turn-up Testing Procedure

To ensure proper service between the IP PBX and the TWC network, test calls from the IP PBX will be made. Typically, the following call types will be used (call testing varies depending on service configuration)

- 1. Outbound/Inbound call to a local number
- 2. Outbound/Inbound call to a long distance number
- 3. Calls to 411 and 611
- 4. Outbound calls to a blocked number to verify call blocking settings
- 5. Other calls based on customer request , e.g. FAX testing using T.38 or calls to an auto-attendant to verify DTMF

Questions

If you have questions, please contact your Time Warner Cable Business Class Account Executive.