

## Broadvox Fusion SIP Trunks Configuration Guide

## **PBX Platform: KX-TDA50**



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### Overview

This document describes the configuration procedures required for the KX-TDA50 to make full use of the capabilities of Broadvox SIP Trunks Services.

The KX-TDA50 is one of Panasonic platforms that support SIP trunks functionality. It implements the Session Initiation Protocol (SIP) through a dedicated (Trunk) Gateway card (KX-TDA5450) which offers 4-Channels of SIP trunks. Capacity of the SIP trunks can be increased by installing an additional DSP daughter card (KX-TDA5451) which provides additional 4-Channels. The maximum numbers of SIP Gateway cards to be installed into one PBX is one card.

This guide describes the specific configuration items for the SIP Gateway card in addition to the PBX basic configuration related to SIP trunks functionality. It also describes basic Network configuration to familiarize dealers with the network setup for the SIP Gateway card. It does not describe the purpose and use of all configuration options on the SIP Gateway card. For those details, see the KX-TDA5450 Installation Manual and the KX-TDA50 Manuals available from Panasonic Reseller Website.



## **Architecture Overview**

The following diagram illustrates simple VOIP networks connecting the SIP-GW4 card: The Case of Broadvox Setup.





## **Basic SIP-GW Settings for Broadvox**

#### 1-Starting the SIP-GW4 Maintenance Utility

To start the SIP-GW4 Maintenance Utility, log in from your PC connected to the card by specifying the default IP address, username, and password.

1. Start Internet Explorer from the Start menu.

2. Specify the URL of the SIP-GW4 Maintenance Utility with the IP address *http://192.168.0.1:8000/Exp*.

3. a. The log-in screen is displayed.

Connect to 192.	168.0.1
expert	
<u>U</u> ser name:	
Password:	
	Remember my password
	OK Cancel

In the User name box, type INSTALLER.

b. In the Password box, type 1234.

c. Click OK.

Now the menu screen of the SIP-GW4 Maintenance Utility is displayed

#### 2-Network Settings

WAN Interface: (i.e DHCP Client)

Panasonic	SIP-GW4 Maintenance Utility
Operation Menu	WAN Interface
1. Programming 1.1 Network Settings <u>1.1.1 WAN Interface</u> <compulsory> <u>1.1.2 LAN Interface</u> 1.2 SIP Settings <u>1.2.1 Channel &amp; SIP Telephony</u> <compulsory> <u>1.2.2 Voice Communication</u> <u>1.2.3 NAT Traversal</u> <u>1.2.4 Options</u></compulsory></compulsory>	Select Connection         WAN Interface       IP Address         PPPoE       -         DHCP Client       -         Fixed IP Address       -         OK (Reboot)       -
1.2 SIP Settings <u>1.2.1 Channel &amp; SIP Telephony</u> <compulsory? <u>1.2.2 Voice Communication</u> <u>1.2.3 NAT Traversal</u> <u>1.2.4 Options</u> 1.3 Advanced Settings (&gt;&gt;&gt; Enable)</compulsory? 	DHGP Client     -     Set     OK (Reboot)

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#### Note:

#### Please check if the WAN port will get an IP address in the same subnet as your LAN port.

Contact the Network Administrator to confirm the WAN network configuration...

#### For example:

LAN port: IP address 192.168.0.1, Subnet mask 24 (255.255.255.0) WAN port: IP address 192.168.0.2, Subnet mask 24 (255.255.255.0) If you do so, disconnect the cables from the LAN and WAN ports and reboot the SIP-GW4 card. Then, connect the cable only to the LAN port, and reassign the IP address and subnet mask address for the WAN port.

You will need to reboot The Gateway card for these settings to be effective

#### 3- Channel & SIP Telephony

#### 1. Click 1.2.1 Channel & SIP Telephony.

Panasonic	SIP-GW4 Maintenance Utility Channel & SIP Telephony						
Operation Menu							
Programming 1.1 Network Settings	Channel Setti	ngs					
1.1   WAN Interface <compulsory></compulsory>	Channel	ITSP Name	Registration to SIP server	Channel Attribute			
L.L.A.L.SH.INCERSCE	Chl	-		Not Used	~	[Set]	
1.2 SIP Settings 1.2.1 Channel & SIP Telephony Scompulsory	Ch2	-	-	Not Used	~	[Set]	
1.2.2 Voice Communication	Ch3	-	- 1	Not Used	~	Set	
1.2.3 NAT Traversal 1.2.4 Options	Ch4	-		Not Used	~	Set	
2. 6 durance d Settinger ( NNS Eachie)	Ch5	-	-	Not Used	~	[Set]	
Contraction Contraction (Contraction of Contraction of Contractio	Che	-	-	Not Used	~	[Set]	
Configuration Management 2.1 Backup (Download from SIP-OW)	Ch7	-	-	Not Used	~	Set	
2 Restore (Upload to SIP-GW) 3 Initialization	Ch8	-	-	Not Used	~	Set	

#### 2. Specify Channel Attribute for each channel referring to the descriptions below.

Attribute	Description
Not Used (default)	The channel is not in use.
Basic channel	A subscriber channel. If you select this attribute, the "Set" button will become active, which allows you to configure SIP Telephony Basic Settings and SIP Telephony Detailed Settings.
Additional channel of ChN (N: 1–8)	Subordinative channel that can be added to <b>Basic channel</b> above. This setting is to be configured when several channels can be used for one subscription with an ITSP. Selects a basic channel number to which an additional channel is added.

#### <u>Note</u>

Channels 5 through 8 are only available when the SIP-DSP4 card is mounted on the SIP-GW4 card.

**3.** Click the **Reboot** button that appears below the table after all the required configuration, including the following SIP Telephony Basic Settings, is complete.



### -SIP Telephony Basic Settings

#### **1-System Level Configuration:**

This section describes system-wide configuration items that are generally required for each KX-TDA50 to work with Broadvox.

\*Operation Menu $\rightarrow$ 1.2 SIP settings $\rightarrow$ 1.2.1 Channel and SIP Telephony

1									
Operation Menu	Chanr	iel & SIP Tele	ephony						
1. Programming	SIP Telephony Settings - Basic	(Channel 1)							
1.1 Network Settings	Sorror Sottings								
<u>1.1.1 WAN Interface</u> <compulsory></compulsory>		nd01-03 fs br	oadvov net						
1.1.2 LAN Interface	Proxy Server <compulsory></compulsory>	1001-05.15.01							
	Registration	💿 Enable 🔘 Dis	able						
1.2 SIP Settings	Registration Timer(Offer)	180 sec (	20-86400)						
<u>1.2.1 Channel &amp; SIP Telephony</u> <cor< td=""><td>Registrar Server</td><td></td><td></td></cor<>	Registrar Server								
1.2.2 Voice Communication	Domain Name								
<u>1.2.3 NAT Traversal</u>	ITSP Name	Broadvov							
<u>1.2.4 Options</u>		DIGUNUX							
	DNS(SRV Record) Resolve Ability	🔵 Enable 💿 Dis	able						

Step	Command	Purpose
Step 1	<pre>Set Proxy Server<compulsory>. (i.e. nd01-03.fs.broadvox.net)</compulsory></pre>	Set the TDA50 SIP server to the Fully Qualified Domain Name (FQDN) for the Broadvox Application Server cluster.
Step 2	Enable Registration.(Default)	Specifies whether to register on the SIP server or not (if required).
Step 3	Set Registration Timer (Offer).(180 sec)	Specifies the length of time that the card offers for Registration to the SIP server (if required).
Step 4	Set Register Server. ( Blank)	Specifies the domain name or IP address of a Registrar server (if provided). If an ITSP provides both Proxy and Registrar Server information, specify the Registrar Server information.



Step	Command	Purpose
Step 5	Set Domain name. ( Blank)	Specifies the domain name (if provided besides proxy and registrar server).
Step 6	Set ITSP Name.(i.e Broadvox)	Specifies the ITSP name (optional).

#### 2-Subscriber Level Configuration Parameters

This section identifies the SIP service subscriber-specific parameters, including registration and authentication. These settings must be unique for each account. Provisioning a subscriber to register with Broadvox allows calls to terminate to the subscriber's line. Please locate this data from your Broadvox welcome Email/Letter.

Account Settings		
SIP Account <compulsory></compulsory>	1112223333	
Authentication ${\rm I\!D}$ <compulsory></compulsory>	welcome	
Authentication Password <compulsory></compulsory>	•••••	

Step	Command	Purpose
Step 1	Set SIP Account <compulsory>.</compulsory>	The register user ID must correspond with the line/port setting on Broadvox
Step 2	Set Authentication ID < <compulsory>.</compulsory>	If the Authentication service is configured on Broadvox, these parameters must be configured to match the Broadvox settings.
Step 3	Set Authentication Password <compulsory>.</compulsory>	Specifies the password (compulsory).

#### -Codecs priority and DTMF Settings

DTMF Sending Method = RFC2833.

Codec G729-AB can be chosen as first priority if you need to reduce the Bandwidth utilization by voice communication.

\*Operation Menu→1.2.2 Voice Communication

#### **Codec Settings**

Channel 1

Codec Priorit	у	1 G.729-AB 2 G.711-Mu 3 none		Set		
Packet Interv	al	20ms 💌				
VAD		🔘 Enable	⊙ Disable			
DTMF Relay	Sending metho	⊙ RFC28	33 O SIP INFO 🔘 Inband			
	Receive	RFC28	33 🗹 SIP INFO			
QoS	SIP	⊙ ToS	Priority 5 💌 Normal 💌			
		O DSCP	40 (0 - 63)			
		O HEX	a0 (0 - FF)			
	RTP / RTCP	⊙ ToS	Priority 5 💌 Normal 💌			
		O DSCP	40 (0 - 63)			
		○ HEX	a0 (0 - FF)			
Fax Detection	1	🔘 Enable	⊙ Disable			
Fax Detection	ı Signal	⊙ CNG/CED ○ CNG ○ CED				

#### - NAT Traversal

Broadvox supports SBC therefore the NAT traversal method can be disabled..

Keep Alive for NAT binding, RTP Type= Blank UDP

\*Operation Menu→1.2.3 NAT Traversal

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Operation Menu	NAT Traversal										
1. Programming	Channe	l Settings (	(only for Basic (	Channel)							
1.1 Network Seturgs <u>1.1.1 WAN Interface</u> <compulsory> <u>1.1.2 LAN Interface</u></compulsory>	Channel	ITSP Name	NAT Traversal	STUN Server	Keep Alive for I Packet Type	VAT Binding Interval(10-60)	rport				
	Ch1	Broadvox	Disable		Blank UDP 💌	20 sec	Enable 💌				
1.2 SIP Settings 1.2.1 Channel & SIP Telephony <cor< td=""><td>Ch2</td><td>Broadvox</td><td>Disable 💌</td><td></td><td>Blank UDP 💌</td><td>20 sec</td><td>Enable 💌</td></cor<>	Ch2	Broadvox	Disable 💌		Blank UDP 💌	20 sec	Enable 💌				

# Essential PBX Configuration: To Assign the DID numbers destinations, Go to 10-3.CO & Incoming Call→ DID table

📕 KX-TDA50 Maintenance Con	isole -	[10.	CO 8	t Incoming Call	- 3. DID Table]											
File(E) Disconnect(D) Tool(T) U	tility(U)	Viev	v(⊻)	Window( <u>W</u> ) Helpi	Э											
🕑 🔳 🖳 🧣																
System Menu 🛛 🗙	10.3	DID T	able													
1.Configuration						4										
2.System		OK	0	Cancel(C)	Apply(A)	]										
3.Group		0.4		Deviaturation (11)	Name Or		ND	1			Destin	-Kan Callin			1	
4.Extension		Aut	omatic	Registration(U)		nerate(	N	J	L		Desun	ation Settir	ig( <u>s</u> )		J	
5.Optional Device		1 - 100	) 1	01 - 200 201 -	- 200 201 - 300 301 - 400			501 -	600	601 - 700	701 - 80	0 801	801 - 900 901 - 1000			
6.Feature				1	1	-	DID D.	.PP		7		1	01			
7.TRS		- 19-2	No	DID	DID Name	-	DID De	stination		Tenant VM Trunk t Number Group No.		u	I Ring for DID	1		
8.ARS			140.	(32 digits)	(20 characters)	Day	Lunch	Break	Night		r Group No	Day	Lunci	h Break	Night	1
9.Private Network		T	1	<sup>1234567890</sup> E	Broadvox	101	500	500	500	1	1	Disable	Disabl	le Disable	Disable	-
0.CO & Incoming Call		1	2						1	1	1	Disable	Disabl	le Disable	Disable	
🔆 1.CO Line Settings			3							1	1	Disable	Disabl	le Disable	Disable	
🚭 2.DIL Table & Port Settings			4							1	1	Disable	Disabl	le Disable	Disable	
🚭 3.DID Table			5							1	1	Disable	Disabl	le Disable	Disable	
A.Miscellaneous			6					ľ.		1	1	Disable	Disabl	le Disable	Disable	
			7							1	1	Disable	Disabl	le Disable	Disable	

Important note:

Make sure that you provide the 10 digits DID number in the DID number field to be able to make use of the additional DID's assigned by the Provider.



### Minimum System Software requirements:

UPCMC	V7.4.1.0 (Latest as of 02/25/13)
SIP-GW card	V 4.100
TDA50 MPR	V 6.000

### **Useful Hints:**

- 1- Update your system software once the update is available through the Dealer website to make use of added features and improved functionality
- 2- Update your PCMC (Maintenance Tool) with the latest version available on the BTS website and through Panasonic Tech support department
- 3- Back up your configuration file with the Good known settings
- 4- Consult with your network administrator prior to installation to guarantee a smooth setup for your system over the existing network

### **Important:**

1- <u>E911</u>:

Please work closely with Broadvox to configure and Test E911..

#### 2- Fax and Modem communication:

Fax or Modem communication requires additional POTS to be connected to the Panasonic PBX in order to provide service reliability. Most of Broadvox VOIP services are provided over Best Effort internet which can adversely affects Fax /Modem Transactions that are time sensitive and depend on accurate Tone detection for successful operation.