

# **Technical Brief**

### Configuring ProSBC with Twilio Elastic SIP Trunking

For complete details and an interactive step-by-step guide, visit: <u>https://docs.telcobridges.com/tbwiki/ProSBC:Twilio</u>

# Introduction

<u>Twilio Elastic SIP Trunking</u> is a cloud-based solution that provides connectivity for IP-based communications infrastructure to connect to the PSTN for making and receiving telephone calls to the rest of the world via any broadband internet connection. Twilio's Elastic SIP Trunking service automatically scales, up or down, to meet your traffic needs with unlimited capacity. In just minutes you can deploy globally with Twilio's easy-to-use self-service tools without having to rely on slow providers.

Sign up for a free Twilio trial and learn more about configuring your Twilio Elastic SIP Trunk

**ProSBC** is connects your IP-PBX to Twilio Elastic SIP Trunking. This article describes how to connect ProSBC to Twilio Elastic SIP Trunking.

## Prerequisites

- ProSBC must be installed as described in their respective installation guides.
- Twilio Elastic SIP Trunking Account. Please visit the following link for more details about Twilio Elastic SIP Trunking

# ENTERPRISE IP-PBX LAN Fax Server

## **Typical Use Case for SIP Trunking with Twilio**

### Configuration Steps:

(Step 1) Create IP Network	(Step 2) Create Protocol Stack	(Step 3) Create Call Route	(Step 4) Create Twilio Elastic SIP Trunk
<ul> <li><u>Configuring</u> <u>an IP Virtual</u> <u>Port</u></li> <li><u>Configuring</u> <u>a VLAN</u></li> <li><u>Configuring</u> <u>IP Interfaces</u></li> <li><u>Create an IP</u> <u>port range</u></li> </ul>	<ul> <li><u>Create an IP port</u> range</li> <li><u>Create a SIP</u> stack</li> <li><u>Create a SIP</u> transport server</li> <li><u>Modify profile</u></li> <li><u>Allocate a SIP</u> <u>NAP</u></li> </ul>	• <u>Create a</u> <u>first call</u> <u>route</u>	<ul> <li><u>Create</u> a new IP ACL</li> <li><u>Create a</u> <u>New</u> <u>Trunk</u></li> <li><u>Associate</u> <u>Phone</u> <u>numbers</u> <u>on your</u> <u>Trunk</u></li> </ul>

# NAP Configuration for Twilio Elastic SIP Trunking

#### See <u>Create SIP NAP</u>

In this example, we will use telcobridges.pstn.twilio.com trunk information:

Configuration	Status		
List			
Editing NAP:			
Name		TWILIO	]
Enabled		$\checkmark$	
Default Profile		default 🗸	
Use Proxy Address			
Proxy address		telcobridges.pstn.twilio.com	]
Proxy port		5060	]
Filter by proxy port			
Poll Remote Proxy		<b>V</b>	
Proxy Polling Interval		1	minutes 🗸
Accept only authorized use	ers		
Registration Parameter	<u>s</u>		
Authentication Paramet	ters		
Network Address Trans	lation (NAT)		
SIP-I Parameters			
Advanced Parameters			
Save			

\*\*NOTE: Connection to Twilio Elastic SIP Trunking is available in multiple geographic edge locations. If you wish to manually connect to a specific geographic edge location that is closest to the location of your communications infrastructure, you may do so by pointing your communications infrastructure to any of the following localized Termination SIP URIs:

- {example}.pstn.ashburn.twilio.com (North America Virginia)
- {example}.pstn.umatilla.twilio.com (North America Oregon)
- {example}.pstn.dublin.twilio.com (Europe Ireland)
- {example}.pstn.frankfurt.twilio.com (Europe Frankfurt)
- {example}.pstn.singapore.twilio.com (Asia Pacific Singapore)
- {example}.pstn.tokyo.twilio.com (Asia Pacific Tokyo)
- {example}.pstn.sao-paulo.twilio.com (South America São Paulo)
- {example}.pstn.sydney.twilio.com (Asia Pacific Sydney)

Click here for more information on Twilio Elastic SIP Trunking IP Addresses

### **Route Configuration for Twilio Elastic Trunking**

• See <u>Create a first call route</u>

In this example, we will create an outgoing and incoming routes for Twilio Elastic Trunk

### Route for outgoing to Twilio

ROUTE_2_TWILIO
Help
Help
IPPBX 🗸
Help
Help
TWILIO
(same as NAP)
(same as NAP)
<b>v</b>

### Route for incoming from Twilio

UTE_FROM_TWILIO	
	Help
	Help
/ILIO 🗸	
	Help
	Help
PBX 🗸	
me as NAP) 🗸	
me as NAP) 🗸 🗸	
2	

# **Twilio Elastic SIP Trunking Configuration**

From your <u>Twilio Console</u>, navigate to the <u>Elastic SIP Trunking</u> area (or click on the sip icon on the left vertical navigation bar).



### **Create a new IP-ACL**

Click on Authentication in the left navigation, and then click on IP Access Control Lists.



In this example an IP-ACL is created with the name: "MyExampleACL". Add your SBCs IP addresses to the IP-ACL you create.

IVIYEAAAA	bickel		
Properties			
FRIENDLY NAME	MyExampleACL		
IP-ACL SID	ALf 17000000000000000000000000000000000000		
ASSOCIATED SIP TRUNKS			
ASSOCIATED SIP DOMAINS			
IP Address Ran	ges		
			IP Access Control Lists may have up to 100 IP addresses.
IP ADDRESS RA	NGE	FRIENDLY NAME	
<b>19</b> 2.1 192.168.0.10 - 1	192.168.0.10		×

### **Create a new Trunk**

MyExampleACL

For each geographical region desired (e.g., North America, Europe), create a new Elastic SIP Trunk.

Now click on Trunks again on the left vertical navigation bar, and create a new Trunk.

	Create A New SIP Trunk		
Name your new SIP	Frunk, then configure it in the following steps.		
FRIENDLY NAME			
	Cancel	Create	

Under the **General Settings** you can enable different features as desired.

#### **Features**

To learn more about SIP Trunking features, please see our user documentation.

#### Call Recording (i)



Calls will not be recorded.

#### Call Recording

Record from ringing

#### **Recording Trim**



Silence will not be trimmed from recording

#### Secure Trunking (i)



RTP must be used for media packets. SIP messages may be sent unencrypted or encrypted using TLS. Any SRTP encrypted calls will be rejected

#### Call Transfer (SIP REFER) (i)



Twilio will consume an incoming SIP REFER from your communications infrastructure and create an INVITE message to the address in the Refer-To header

#### Enable PSTN Transfer ()

Allow Call Transfers to the PSTN via your Trunk.

#### Symmetric RTP (i)



Twilio will detect where the remote RTP stream is coming from and start sending RTP to that destination instead of the one negotiated in the SDP

#### Additional Features

### In the Termination section, select a Termination SIP URI.

### **Termination URI**

Configure a SIP Domain Name to uniquely identify your Termination SIP URI for this Trunk. This URI will be used by your communications infrastructure to direct SIP traffic towards Twilio. Be sure to select a localized SIP URI to ensure your traffic takes the lowest latency path. If a localized version isn't selected, then your traffic will be sent to US1. Learn more about Termination Settings 7

TERMINATION SIP URI	telcobridges		.pstn.twilio.com
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Show Localized URIs

Click on "Show localized URI's" and copy and paste this information as you will use this on your SBC to configure your Trunk.

NORTH AMERICA VIRGINIA	telcobridges.pstn.ashburn.twilio.com
NORTH AMERICA OREGON	telcobridges.pstn.umatilla.twilio.com
EUROPE DUBLIN	telcobridges.pstn.dublin.twilio.com
EUROPE FRANKFURT	telcobridges.pstn.frankfurt.twilio.com
SOUTH AMERICA SAO PAULO	telcobridges.pstn.sao-paulo.twilio.com
ASIA PACIFIC SINGAPORE	telcobridges.pstn.singapore.twilio.com
ASIA PACIFIC TOKYO	telcobridges.pstn.tokyo.twilio.com
ASIA PACIFIC SYDNEY	telcobridges.pstn.sydney.twilio.com

or

Assign the IP ACL ("MyExampleACL" ) that you created in the previous step.

#### Authentication View all Authentication lists

The following IP ACLs and Credential Lists will be used to authenticate the INVITE for termination calls inbound to Twilio.



In the **Origination** section, we'll add Origination URI's to route traffic towards your TelcoBridges SBC. The recommended practice is to configure a redundant mesh per geographic region (in this context a region is one of North America, Europe, etc). In this case, we configure two Origination URIs, each egressing from a different Twilio Edge.

### Click on 'Add New Origination URI', we'll depict the configuration for North America:

	Add Origination URL	
ORIGINATION SIP URI	sip:telcobridges.com;edge=ashburn	
PRIORITY	10	
	Priority ranks the importance of the URI. Values range from 0 to 65535, where the lowest number represents the highest importance.	
WEIGHT	10	
	Weight is used to determine the share of load when more than one URI has the same priority. Its values range from 1 to 65535. The higher the value, the more load a URI is given.	
ENABLED	ON	
	Cancel Ad	d

Continue to add the other Origination URIs if desired. Origination URIs are listed as follows:

Ori	gination URIs					
Conf	Configure the IP address (or FQDN) of the network element entry point into your communications infrastructure (e.g. IP-PBX, SBC).					
Show	v more about provisioning for high service availability					
Ð	ORIGINATION URI	PRIORITY	WEIGHT	ENABLED		
	sip:telcobridges.com;edge=ashburn	10	10	~	$\times$	
	sip:telcobridges.com;edge=umatilla	10	10	~	$\times$	

In this example, Origination traffic is routed via Twilio's Ashburn edge, if that fails then we'll route from Twilio's Umatilla edge.

### **Associate Phone Numbers on your Trunk**

In the **Numbers** section of your Trunk, add the Phone Numbers that you want to associate with each Trunk. Remember to associate the Numbers from a given country in the right Trunk. For example, associate US & Canada Numbers with the North American Trunk and European Numbers with the European Trunk etc.

lumbers					View my Address
Emergency Calling select numbers to			be associated with an emerger time.	cy address with matching ISO	Country. Please
Number	~		Filter		hoose Action 🗸
NUMBER	FRIENDLY NAME	COUNTRY	EMERGENCY CALLING STATUS	EMERGENCY ADDRESS	
+18507004044	(850) 790-40 <b>44</b>	US	Enabled	375 BEALE ST 3rd floor suite,	SF, CA, 94105
+16800000000	(689) 220 30 <b>33</b>	US	Enabled	375 BEALE ST 3rd floor suite,	SF, CA, 94105
+176	(769) 210-5055	US	Disabled		

# Troubleshooting

Support resources, including access to the free TBwiki and TB Support Forum, can be found here: <u>https://freesbc.telcobridges.com/support/</u>