3CX Trunk Configuration for Spark SIP Connect Trunks

Create Trunk

First Create a new Trunk in 3CX using the "Generic" Template

Add SIP Trunk/VoIP Provider	×
Select Country	
🕋 Generic	•
Select Provider in your Country	
Generic SIP Trunk	~
Main Trunk No	
Lead Number as Specified by Spark	
	OK Cancel

Specify The Main Trunk No. as the Lead Number specified by Spark.

General

Enter the details on the General Tab as shown below

Spark SIP Trunk ox Cancel	🕲 Help
General DIDs Caller ID Options Inbound Parameters Outbound Parameters	
Trunk Details	
Enter name for Trunk	
Spark SIP Trunk (Description of your choice)	
Registrar/Server/Gateway Hostname or IP	
sip-net.spark.co.nz (SIP Realm as specified by Spark) 5060 Auto Discovery	
Outbound Proxy	
ric0102-p01.spark.co.nz (Outbound Proxy as specified by Spark) 5060 Auto Discovery	
Number of SIM Calls	
10	

Authentication
Type of Authentication
Register/Account based 🗸
Authentication ID (aka SIP User ID)
35437015 (The Authentication ID as specified by Spark Note the lack of the leading 0)
Authentication Password
Authentication Password as specified by Spark
3 Way Authentication Password

DDI's

On the DDI Tab, enter any DDI numbers or test numbers that have been assigned to the SIP Trunk by Spark

Spark SIP Trunk OK Cancel	🕲 Help
General DIDs Caller ID Options Inbound Parameters Outbound Parameters	
DIDs	
+ Add Single DID X Delete	
DID/DDI Number	
0354370	×
0354370	×
0354370	×

Caller ID

On the Caller ID Tab specify the Default Outbound Caller ID for the trunk (Note individual Outbound Caller ID's can be specified per extension and will override this Caller ID)

Spark SIP Trunk ок Cancel	🕲 Help
General DIDs Caller ID Options Inbound Parameters Outbound Parameters	
Reformat Incoming or Outgoing Caller Identification numbers by configuring matching patterns. Click here for more information.	
Default caller ID	
Configure Outbound Caller ID	
0354370	

Options

On the Options tab Scroll to the bottom and remove all but G711-a Law codec

	() Help
General DIDs Caller ID Options Inbound Parameters Outbound Parameters	
Call options	
Allow inbound calls	
Allow outbound calls	
Disallow video calls	
Advanced	
PBX Delivers Audio	
Supports Re-Invite	
Support Replaces	
Put Public IP in SIP VIA Header	
Alternative Proxy	
Alternative Development	
Alternative Proxy	
Alternative Proxy SRTP Be-Berister Timeout	
Alternative Proxy SRTP Re-Register Timeout 180	
Alternative Proxy SRTP Re-Register Timeout 180 Select which IP to use in 'Contact' (SIP) and 'Connection' (SDP) fields	
Alternative Proxy SRTP Re-Register Timeout 180 Select which IP to use in 'Contact' (SIP) and 'Connection'(SDP) fields Use Default Settings	~
Alternative Proxy SRTP Re-Register Timeout 180 Select which IP to use in 'Contact' (SIP) and 'Connection'(SDP) fields Use Default Settings Transport Protocol Transport Protocol	~
Alternative Proxy SRTP Re-Register Timeout 180 Select which IP to use in 'Contact' (SIP) and 'Connection' (SDP) fields Use Default Settings Transport Protocol Any	~
Alternative Proxy SRTP Re-Register Timeout 180 Select which IP to use in 'Contact' (SIP) and 'Connection'(SDP) fields Use Default Settings Transport Protocol Any IP Mode Any	~
Alternative Proxy SRTP Re-Register Timeout 180 Select which IP to use in 'Connection'(SDP) fields Use Default Settings Transport Protocol Any IP Mode Any	~ ~
Alternative Proxy SRTP Re-Register Timeout 180 Select which IP to use in 'Contact' (SIP) and 'Connection'(SDP) fields Use Default Settings Transport Protocol Any IP Mode Any Codec Priority	v v
Atternative Proxy SRTP Re-Register Timeout 180 Select which IP to use in 'Contact' (SIP) and 'Connection'(SDP) fields Use Default Settings Transport Protocol Any IP Mode Any	~ ~ ~
Atternative Proxy SRTP Re-Register Timeout 180 Select which IP to use in 'Contact' (SIP) and 'Connection'(SDP) fields Use Default Settings Transport Protocol Any IP Mode Any Codee Priority Codee Priority 3 More Up 3	✓ ✓ ✓

Inbound

On the Inbound Parameters unselect the Option "Call Source Identification"

park SIP Trunk ок Cancel		🕲 Helj
General DIDs Caller ID Options Inbound Parameters Outbound Parameters		
Caller Number/Name Field Mapping:		
Review the SIP header of the INVITE and specify where the following values should be present within the	INVITE:	
"CalledNum" number that has been dialed (default: To->user)	To : User Part	~
"CallerName" caller's name (default: From->display name)	From : User Part	~
"CallerNum" caller's number (default: From->user)	From : User Part	~
Call Source Identification		
Configure this option only when the SIP Trunk is IP based (peering), or does not support automatic inbox you might need to toggie this option (on(off) and see what configuration works best for this SIP Trunk	und call detection. If you have multiple trunks from the same vendor or issues with incoming ca	lls,
Request Line URI : User Part		~
"EnforcedOriginatorCallerID" To be used when you want to send Anonymous via PAI		~
Use both "Call Source Identification" rules and "Caller Number/Name -> CalledNum" field mapping:	(Note: Disables catch all routing capability)	

Outbound

On The outbound Parameters you can Leave most of the settings as they are in the Generic Template except for the PAI (P-Asserted Identity) The P-Asserted Identity : User Part should be set to AuthID and the P-Asserted Identity : Host Part should be set to GWHostPort as shown below (Scroll down to the bottom of the outbound Parameters)

Without these parameters set Spark will not Authenticate or accept your Outbound Calls and you will receive a 404 Not found error when trying to make calls.

Seneral DIDs Caller ID Options Inbound Para	meters Outbound Parameters	
Outbound Parameters		
Assign SIP header fields to 30Y Call Variables Requires advance	of SIP knowledge. Misconfiguration will cause your PRY to malfunction	
cin field	Variable	Custom Value
Request Line URI : User Part	"CalledNum" number that has been dialed (default: To->r	Custom Value
Request Line URI : Host Part	"GWHostPort" gateway/provider host/port	
Contact : User Part	"OutboundCallerId" Outbound caller Id taken from Exten 🗸	
Contact : Host Part	"ContactUri" usually, content of Contact field	
To : Display Name	"CalledName" name that has been dialed (default: To->d 💙	
To : User Part	"CalledNum" number that has been dialed (default: To-> 🗸 🗸	
To : Host Part	"GWHostPort" gateway/provider host/port	
From : Display Name	"OutboundCallerId" Outbound caller Id taken from Exten 🗸	
rom : Host Part	"GWHostPort" gateway/provider host/port	
ser Agent : Text String	Leave default value	
emote Party ID - Called Party : Display Name	Leave default value	
emote Party ID - Called Party : User Part	Leave default value	
emote Party ID - Called Party : Host Part	Leave default value	
emote Party ID - Calling Party : Display Name	"OutboundCallerid" Outbound caller Id taken from Exten 🗸	
emote Party ID - Calling Party : User Part	"OutboundCallerid" Outbound caller Id taken from Exten 🗸	
temote Party ID - Calling Party : Host Part	"GWHostPort" gateway/provider host/port	
-Asserted Identity : Display Name	Leave default value	
-Asserted Identity : User Part	"AuthiD" authentication	
-Asserted Identity : Host Part	"GWHostPort" gateway/provider host/port.	
Preferred Identity : Display Name	Leave default value	
Preferred Identity : User Part	Leave default value	