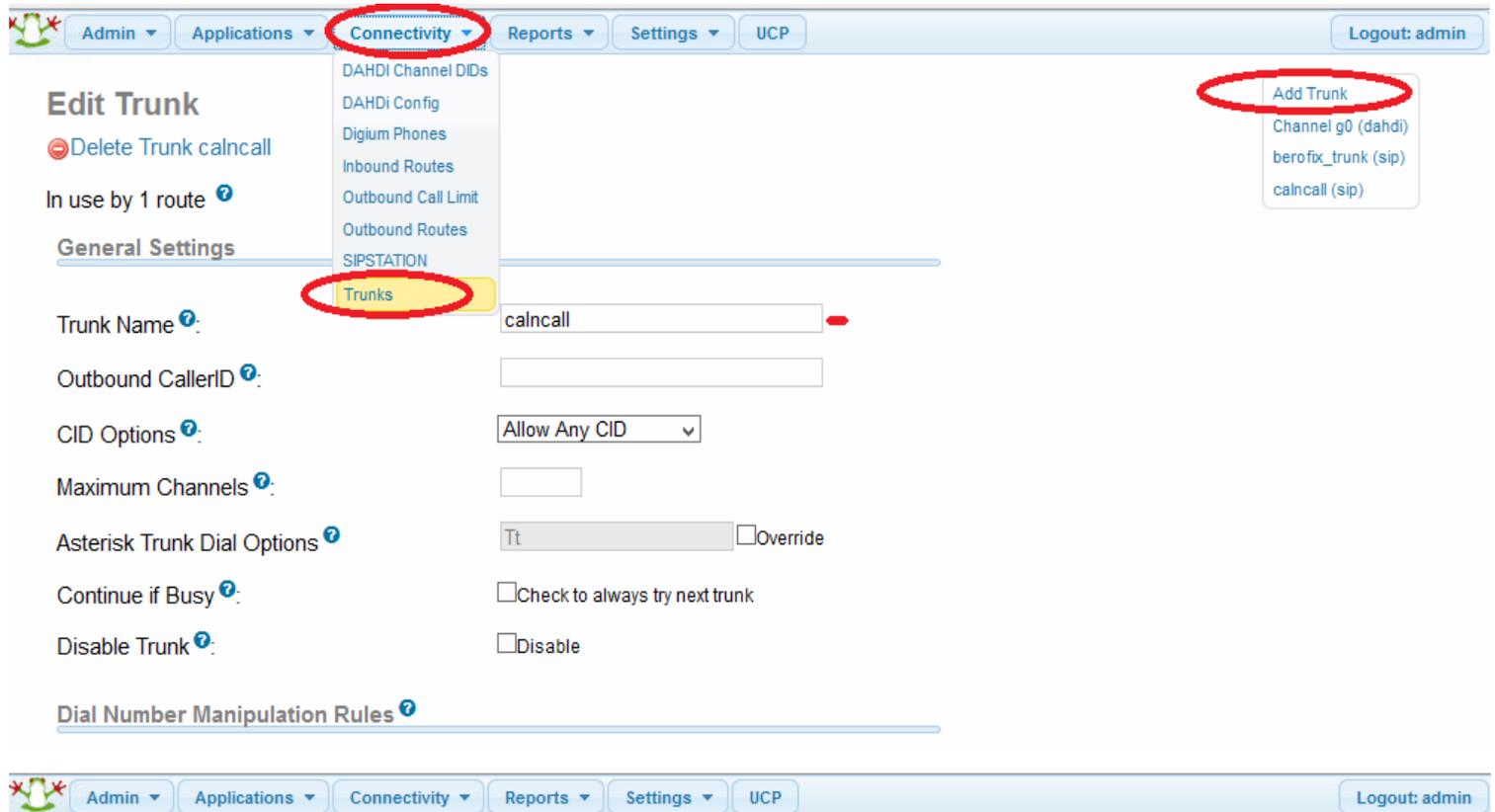


Asterisk Trunk configuration with Calncall SIP Trunk.

Please follow the screen short it will show you the configuration.



The screenshot shows the Asterisk management interface. At the top, there is a navigation bar with tabs: Admin, Applications, Connectivity, Reports, Settings, and UCP. The 'Connectivity' tab is selected and circled in red. Below the navigation bar, the 'Edit Trunk' page is displayed for the trunk named 'calncall'. The 'General Settings' section is visible, with the following fields:

- Trunk Name: calncall
- Outbound CallerID: (empty)
- CID Options: Allow Any CID
- Maximum Channels: (empty)
- Asterisk Trunk Dial Options: Tt Override
- Continue if Busy: Check to always try next trunk
- Disable Trunk: Disable

On the right side of the page, there is a 'Trunks' list with the following entries: Add Trunk, Channel g0 (dahdi), berofix_trunk (sip), and calncall (sip). The 'Add Trunk' button is circled in red.



The screenshot shows the 'Add a Trunk' page in the Asterisk management interface. The navigation bar is the same as in the previous screenshot. The 'Add a Trunk' section is visible, with the following options:

- Add SIP (chan_sip) Trunk
- Add DAHDI Trunk
- Add IAX2 Trunk
- Add ENUM Trunk
- Add DUNDi Trunk
- Add Custom Trunk

On the right side of the page, there is a 'Trunks' list with the following entries: Add Trunk, Channel g0 (dahdi), berofix_trunk (sip), and calncall (sip).



General Settings

Trunk Name [?]:

Outbound CallerID [?]:

CID Options [?]:

Maximum Channels [?]:

Asterisk Trunk Dial Options [?]: Override

Continue if Busy [?]: Check to always try next trunk

Disable Trunk [?]: Disable

Dial Number Manipulation Rules [?]

() + |

[+ Add More Dial Pattern Fields](#)

[Clear all Fields](#)

Dial Rules Wizards [?]:

Outbound Dial Prefix [?]:

Outgoing Settings

Trunk Name [?]:

Outgoing Settings

Trunk Name [?]:

PEER Details [?]:

```
host=siptrunk.calncall.com  
username=***SIP user name***  
secret=***SIP password***  
type=peer
```

Incoming Settings

USER Context [?]:

USER Details [?]:

```
host=siptrunk.calncall.com  
type=peer  
context=from-trunk
```

Registration

Register String [?]:

SIP user name:SIP password@siptrunk.calncall.com/SIP user name

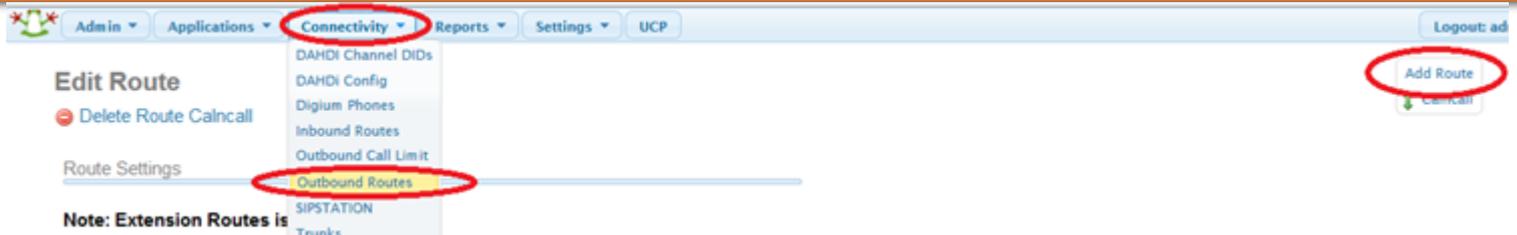
[Submit Changes](#)

[Duplicate Trunk](#)

FreePBX Outbound configuration with CalnCall SIP Trunk.

Follow the below steps to configure outbound rule

Step1: Goto → connectivity -> outbound Routes once you click outbound routes you can get below screenshot



Step2: You have to give Route name for your identification. Need to give dial Patten for this outbound routes default is X.

Edit Route

[-] Delete Route Calncall

Route Settings

Note: Extension Routes is not registered

Route Name [?]:

Route CID: [?] Override Extension [?]

Route Password: [?]

Route Type: [?] Emergency Intra-Company

Music On Hold? [?]

Time Group: [?]

Route Position [?]

Additional Settings

Note that the meaning of these options has changed. [Please read the wiki for futher information on these changes.](#)

Call Recording [?]:

PIN Set [?]:

Dial Patterns that will use this Route [?]

() + prefix / CallerID

() + prefix / CallerID

Dial patterns wizards [?]:

Step3: Need to select trunk sequence (which trunk you want use to make outgoing calls)

[+ Add More Dial Pattern Fields](#)

Dial patterns wizards [?]:

Export Dialplans as CSV [?]: [Export](#)

Trunk Sequence for Matched Routes [?]

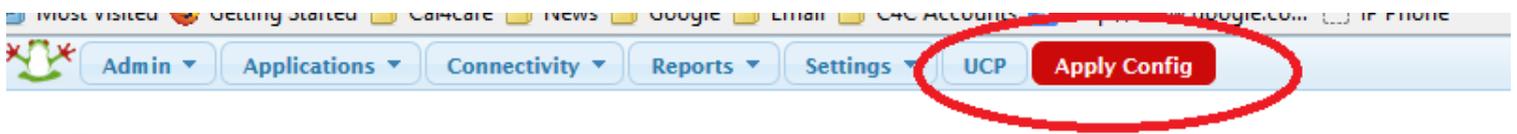
- 0
- 1

[Add Trunk](#)

Optional Destination on Congestion [?]

[Submit Changes](#) [Duplicate Route](#)

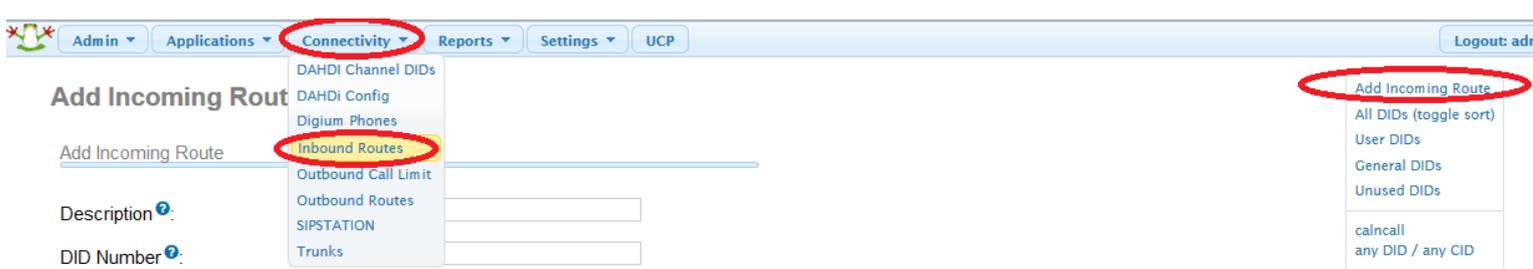
Once you submit you have to apply configuration on top.



FreePBX Inbound configuration with CalnCall SIP Trunk.

Follow the below steps to configure outbound rule

Step1: Goto → connectivity -> inbound Routes once you click inbound routes you can get below screenshort



Step2: You have to give Description for your identification. Need to give DID Number for this Inbound routes, it should be SIP Trunk User Name.

Route:

 Delete Route

 Edit Extension 102 (102)

Edit Incoming Route

Description : Calncall Acct. SIP User Name

DID Number : Calncall Acct. SIP User Name

CallerID Number :

CID Priority Route :

Options

Alert Info :

CID name prefix :

Music On Hold : ▾

Signal RINGING :

Reject Reverse Charges :

Pause Before Answer :

Privacy

Privacy Manager : ▾

Step3: Goto Set Destination Tab and Need to select where you want receive the call, it may be Ext., IVR, ring group

Call Recording

Note that the meaning of these options has changed. Please read the wiki for further information on these changes.

Call Recording [?]:

Force Yes **Don't Care** No Never

CID Lookup Source

Source [?]:

None ▾

Fax Detect

== choose one ==

Announcements **No** **Yes**

Call Flow Control

Call Recording

Callback

Conferences

Custom Applications

DISA

Directory

Extensions

Feature Code Admin

IVR

Languages

Misc Destinations [?]

Paging and Intercom

Phonebook Directory

Queue Priorities

Queues

Ring Groups

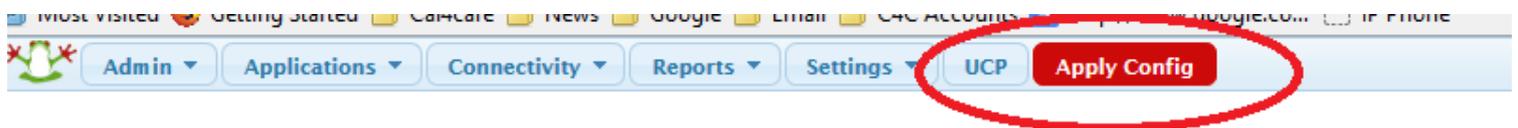
Sipstation

Extensions <102> 102

Submit

Clear Destination & Submit

Once you submit you have to apply configuration on top.



For any more clarification please contact our technical support

Email: support@cal4care.com.sg

Telephone: +65-63401006

Fax: +65-63401007

Web: www.cal4care.com.sg