



CounterPath Configuration of Bria 3.0

Updated: 09-14-2010

Note: Bria has instant messaging and video conferencing features built into the phone. Neither of these features are supported by FacetPhone, and if used, could cause undesirable effects.

First, setup the configuration on the FacetPhone side. Define the new IP phone with properties like the image included below from the FacetPhone GUI ip-phone sheet.

The settings particular to Bria 3.0 are the manufacturer type of "Generic SIP IP phone", "Dynamic IP address", and "Transmit digits as RFC 2833 packets" checked.

You do not need to fill in the MAC address, or any of the other settings on this properties sheet.

IP Phone <New>

Use next available station number

Station number: 110

Manufacturer: Generic SIP IP phone

IP phone description: Bria SoftPhone on PC

Dynamic IP address

IP address of the IP phone: N/A

MAC address of the IP phone:

Default route:

NAT address:

Branch where IP phone located: Dallas office

Operator group: FacetCorp

Dialing restriction: <none>

Default outgoing trunk group(s): 0

Digits transmitted in audio

Transmit digits as RFC 2833 packets

Route all audio through the server

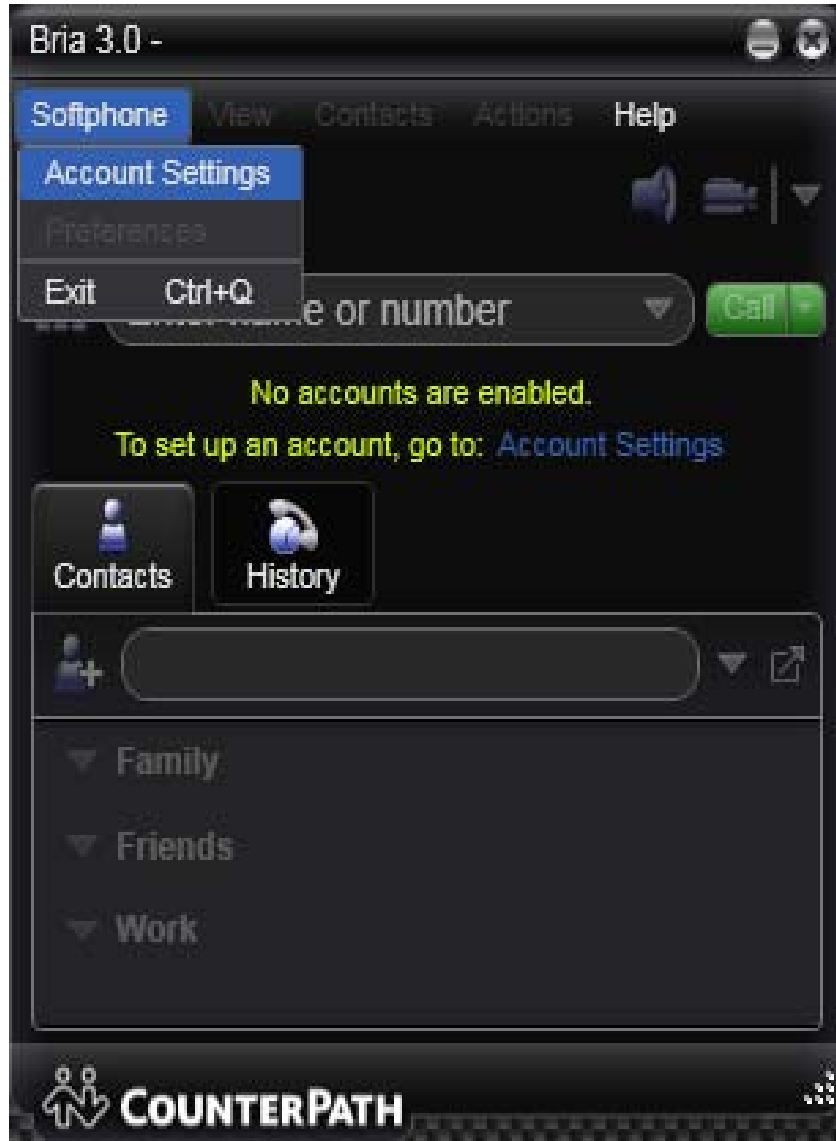
Route audio through the server for different branch

OK Cancel



Now startup the Bria 3 application. You can close the video window because we don't support that yet! Click "Softphone" and choose "Account Settings."

Fill in all of your information from the IP Phone section of the FacetPhone GUI.





Enter the account details as follows:

User Details:

Station###

FacetPhone IP Address

Password 1234

Display Name: Anything

Authorization Name: FacetPhone Username

Domain Proxy:

Register with domain and receive calls: Place a check-mark

Send outbound via: Proxy

Enter FacetPhone IP Address

Dial Plan is located on the FacetPhone server: [Server IPADDR]

http://xxx.xxx.xxx.xxx/facetphone/config_report.txt

SIP Account

Storage Transport Advanced

Account Voicemail Topology Presence

Account name: **station110**

Protocol: SIP

Use for

Call

IM / Presence

User Details

User ID: **station110**

Domain: **xxx.xxx.xxx.xxx (call agent)**

Password:

Display name: **Name or Description**

Authorization name: **FacetPhone User-Name**

Domain Proxy

Register with domain and receive calls

Send outbound via:

Domain

Proxy Address: **xxx.xxx.xxx.xxx (call agent)**

Dial plan: *001xxxxxxxx[2-9]xxxxxxxx[3x1xxxxxxxx[2-9]xxxxxxxx[1xx[2-9]x#1[1-2]1x

OK Cancel



Now click on the Voicemail tab.
Enter star key (*) as the Number to dial for checking voicemail.
Once you have made all of the appropriate changes, click "OK."

The screenshot shows the 'SIP Account' configuration window with the 'Voicemail' tab selected. The 'Check for voicemail' checkbox is checked. The 'Number to dial for checking voicemail' field is highlighted with a yellow box and contains the asterisk (*). Other fields include 'Number for sending calls to voicemail', 'Send calls to voicemail if unanswered for' (set to 0 seconds), and a 'Forwarding' section with 'Always forward to' and 'When on the phone, forward to' options.

Storage	Transport	Advanced
Account	Voicemail	Topology
	Presence	

Check for voicemail

Number to dial for checking voicemail: *

Number for sending calls to voicemail: []

Send calls to voicemail if unanswered for: 0 seconds

Forwarding

Always forward to: []

When on the phone, forward to: []

OK Cancel



Click on Softphone, followed by Preferences.





Click on Advanced, and now match the DTMF and RTP settings to the image below.

