

Avaya IP Office Essential Edition

Extension Sends Its DID Number via SIP

Telquest Tech Support

The screenshot shows the 'User (25)' settings page. At the top, there are tabs for 'Hunt Group Membership', 'Announcements', and 'SIP'. A callout '1. Click here...' points to the 'User (25)' icon. Another callout '2. Click here...' points to the 'SIP' tab. A large blue box contains the text: 'This TAB will only appear if any SIP Lines "Local URI" is set to "Use Internal Data" See Page 3...'. Below the tabs, there are three rows of settings:

SIP Name	SIP-Name	From: "SIP-Display-Name" <sip:SIP-Name@callcentric.com>
SIP Display Name (Alias)	SIP-Display-Name	From: "SIP-Display-Name" <sip:SIP-Name@callcentric.com>
Contact	Contact-Name	Contact: "SIP-Display-Name" <sip:Contact-Name@192.

Callouts include 'IP Office Settings' pointing to the settings area and 'This is what is sent to the ITSP' pointing to the contact information in the third row.

Place the number you want to send in all 3 of the IP Office Settings. (SIP Name, SIP Display, Contact)
Different ITSP's use different areas for the number you want to send.

From: "SIP-Display-Name" <sip:SIP-Name@callcentric.com>

SIP Name area

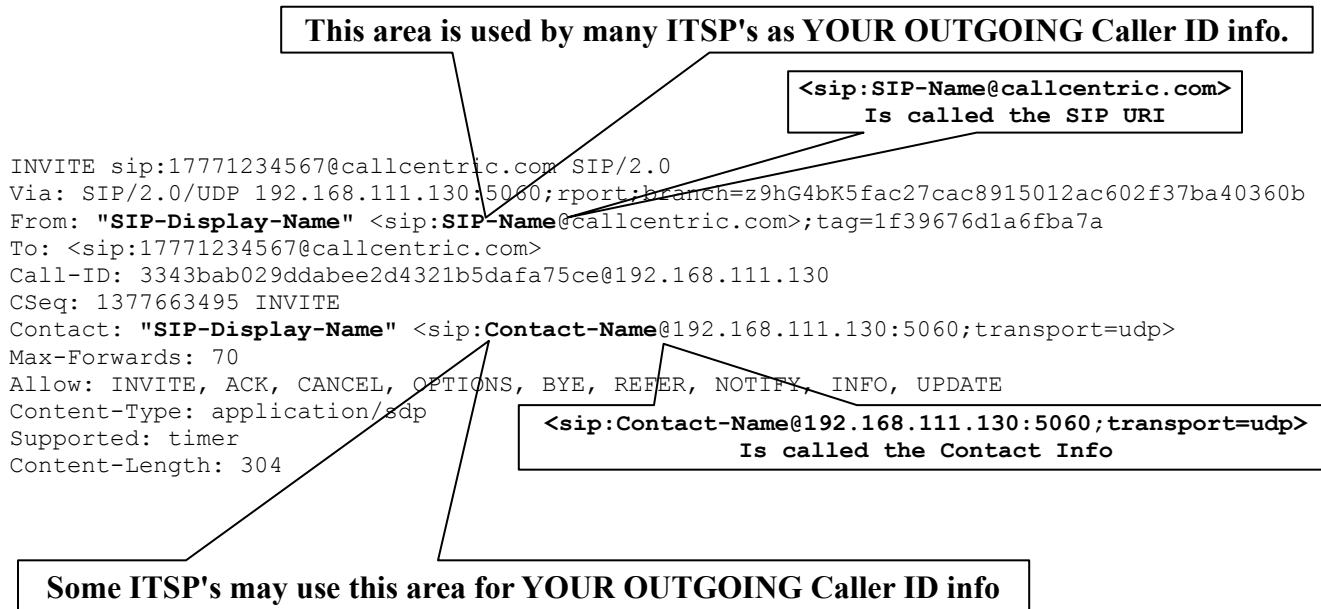
From: "SIP-Display-Name" <sip:SIP-Name@callcentric.com>

SIP Display Name (Alias) area

Contact: "SIP-Display" <sip:Contact-Name@callcentric.com>

Contact area

Here is the actual Invite Message that was sent. (SDP removed)



Note:

Even though you may change any of the settings, the ITSP may reject it and only allow you send info that is recognized by them as valid.

This means that if you have a bank of DID Numbers from the ITSP, you can ONLY send any of the numbers that you have subscribed to.

If you send a number that is not in your bank of DID's, then they will most likely reject it and the call may not go through.

Making changes in the Local URI of a SIP Line (aka SIP Trunk)

1. Click here...

2. Select or Create a SIP Line

3. Click here...

4. Click here...

5. Change to this...

FYI: "Local URI" = "SIP Name" in User settings...

Line	
Line Number	Line Type
1	Analogue Trunk
2	Analogue Trunk
3	Analogue Trunk
4	Analogue Trunk
5	PRI 24 (Univers
17	SIP Line
18	SIP Line
19	SIP Line

SIP Line Transport SIP URI VoIP T38 Fax SIP Credentials

Add...
Remove
Edit...

Edit Channel

Via <None>

Local URI Use Internal Data

Contact Use Credentials User Name

Display Name Use Credentials User Name

PAI None

Registration 0: <None>

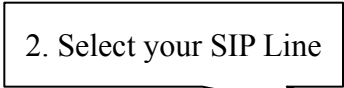
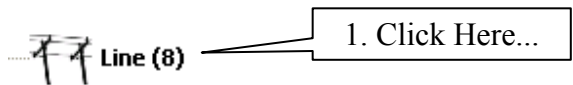
Incoming Group 0

Outgoing Group 0

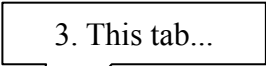
Max Calls per Channel 10

Note: "Use Internal Data" means to use the USERS SIP setting.

The SIP Line will send out the Users "SIP Name" (usually the DID Number) in the "From:" "SIP URI" portion of the SIP Header.



Line		
Line Number	Line Type	Line SubType
1	Analogue Trunk	
2	Analogue Trunk	
3	Analogue Trunk	
4	Analogue Trunk	
5	PRI 24 (Universal)	PRI
17	SIP Line	
18	SIP Line	
19	SIP Line	



SIP Line

Transport

SIP URI

VoIP

T38 Fax

SIP Credentials

Line Number

17

ITSP Domain Name

callcentric.com

In

Prefix

Use

National Prefix

0

Cal

Country Code

Ori

International Prefix

00

for

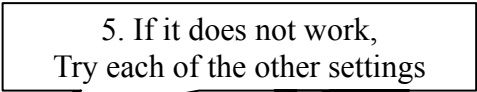
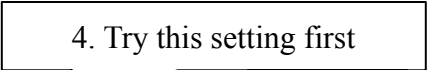
Send Caller ID

P Asserted ID

Nar

Association Method

By Source IP address



Send Caller ID

None

Diversion Header

P Asserted ID

Remote Party ID

None