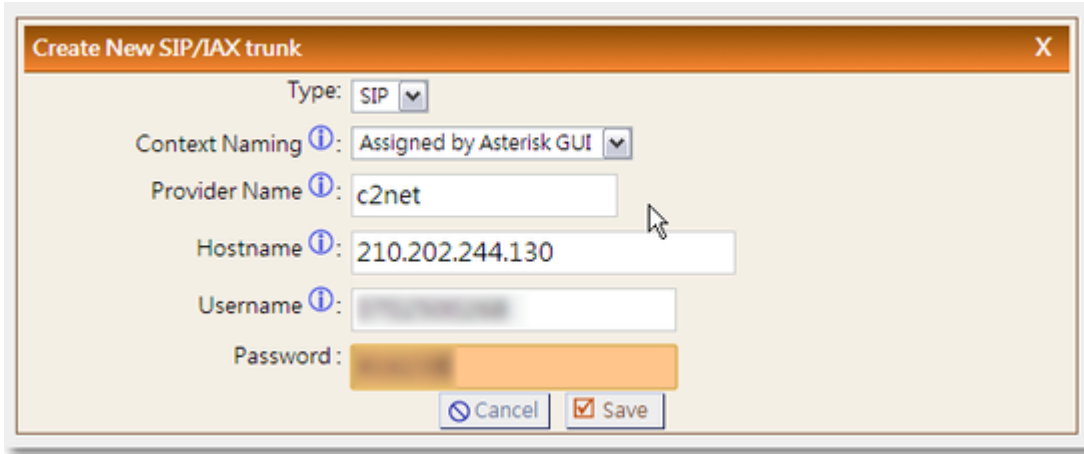


IP PBX ????????????????

1. Trunk ?? : ? IP PBX ??? SIP Provider ?????????????? C2NET ???
2. Outgoing Calling Rules ????
3. Dialplan ????
4. Users ????
5. Incoming Calling Rules ????

### 1. ?? Trunk

Web-UI > Trunks > VOIP Trunks > New SIP/IAX Trunk



The screenshot shows a dialog box titled "Create New SIP/IAX trunk" with a close button (X) in the top right corner. The form contains the following fields:

- Type: SIP (dropdown menu)
- Context Naming (i): Assigned by Asterisk GUI (dropdown menu)
- Provider Name (i): c2net (text input)
- Hostname (i): 210.202.244.130 (text input)
- Username (i): [blurred] (text input)
- Password: [blurred] (password input)

At the bottom of the dialog, there are two buttons: "Cancel" and "Save" (with a checkmark icon).

Web-UI > Trunks > VOIP Trunks > Edit c2net

- FromUser: ??? SIP ???? username ??
- insecure: ????? very??????????????

The screenshot shows a configuration window titled "Edit SIP trunk trunk\_1". It contains the following fields and options:

- Provider Name: c2net
- Hostname: 210.202.244.130
- Username: [blurred]
- Password: [blurred]
- Codecs: First: u-law, Second: a-law, Third: GSM, Fourth: G.726, Fifth: [empty]
- CallerID: [empty]
- FromDomain: [empty]
- FromUser: [blurred] (highlighted with a red box)
- AuthUser: [empty]
- insecure: no
- Outbound Proxy: [empty]
- Enable Remote MWI:

Buttons: Cancel, Save

????? C2NET ? SIP Server ?? port 7890????????? port 5060????????????????????? SIP Server ? 5060 ????????????????

Web-UI > File Editor(????) > users.conf >?? [trunk\_1]

- ?? port=7890
- ?? registersip=no

Web-UI > File Editor(????) > sip.conf > ?? [general]

- ????? register=sip\_number:sip\_pass@210.202.244.130:7890

????????????????? Apply Changes

## 2. ?? Outgoing Calling Rules - ????

????????? 011 + <C2NET ??>

Web-UI > Outgoing Calling Rules > New Calling Rule

- Calling Rule Name: ??????? c2net
- Pattern: ?????????????? 011????? \_011.???????????????
- Use Trunk: ???? Trunk
- Strip: ?????? 3 ?????????? 3



- DialPlan: ?????????? internal
- SIP: ??
- SIP/IAX Password: SIP ??
- NAT: ????? NAT???????????? LAN?????

**Create New User**

General:

Extension: 6000 CallerID Name: alang DialPlan: internal

Internal CallerID: 6000 CallerID Number:

Enable Voicemail for this User

VoiceMail Access PIN code: Email Address:

Technology

SIP  IAX Analog Station: None flash: 750 rxfash: 1250

Codec Preference: First: u-law Second: GSM Third: None Fourth: None Fifth: None

VoIP Settings

MAC Address: Line Number: 1 LineKeys: 1

SIP/IAX Password: 1234 IAX: Require Call Token:

IAX: Max Call Numbers:

NAT:  Can Reinvite:  DTMF Mode: RFC2833 insecure: no

Other Options

3-Way Calling (analog)  In Directory  Call Waiting (analog)

ADA User  Is Agent Pickup Group: 1

Cancel Update

### 5.?? Incoming Calling Rules

Web-UI > Incoming Calling Rules > New Incoming Rule

???????????? 6000

- Trunk: ??? SIP Trunk ??
- Time Interval: ???????
- Pattern: ????????? s
- Destination: ???????

The image shows a 'New Incoming Rule' dialog box with the following fields and values:

- Trunk: c2net
- Time Interval: None (no Time Intervals matched)
- Pattern: s
- Destination: User Extension -- 6000

Buttons: Cancel, Update

TIPs?

???Pattern ??? DID ?????????? Trunk ??? DID ??????????????????

????????? WEb-UI ??? Trunk ? Incoming Calling Rules????????? sip.conf?? extensions.conf

??