

ایجاد ترانک SIP و یا IAX

طرف اول : IP04A

افزودن کاربر 100 جهت استفاده به عنوان ترانک که در ترانک B وارد می شود.

Extension	Full Name	Port	SIP	IAX	DialPlan	OutBound CID			
<input type="checkbox"/>	100	100	--	Yes	Yes	DialPlan1	100	<input type="button" value="Edit"/>	<input type="button" value="Delete"/>
<input type="checkbox"/>	101	101	--	Yes	Yes	DialPlan1	none	<input type="button" value="Edit"/>	<input type="button" value="Delete"/>
<input type="checkbox"/>	102	--	4	--	--	DialPlan1	none	<input type="button" value="Edit"/>	<input type="button" value="Delete"/>

Edit User Extension - 100

General :

Extension: Name: DialPlan:
CallerID: OutBound CallerID:

Enable Voicemail for this User

VoiceMail Access PIN code: Mailbox: Email Address:

Technology

SIP IAX Analog Station: flash: rxflash:
Codec Preference : First : Second : Third : Fourth : Fifth :

VoIP Settings

MAC Address : Line Number : SIP/IAX Password:
NAT: Can Reinvite: DTMF Mode: insecure:

Other Options

3-Way Calling In Directory Call Waiting CTI Is Agent
 Enable Call Record Pickup Group:

افزودن ترانک ToB

Manage SIP & IAX trunks

Analog Trunks Service Providers **VOIP Trunks** T1/E1 BRI Trunks

+ New SIP/IAX Trunk

Provider Name	Type	Hostname/IP	Username	
ToB	SIP	192.168.16.91	202	Edit Delete

Edit SIP trunk 202

Provider Name: ToB

Hostname: 192.168.16.91

Username: 202

Fromuser:

Fromdomain: 202

Password: 202

Contact Ext.: s

Insecure Type: very

Codecs: First: u-law Second: a-law Third: GSM

Fourth: G.726 Fifth: None

CallerID:

Enable Remote MWI

[Cancel](#) [Add](#)

افزودن Outgoing rule

Calling Rule	Pattern	Trunk	Failover Trunk	
pstn	_0X.	pstn	None Selected	Edit Delete
ToB	_2X.	ToB	None Selected	Edit Delete
toBpstn	_29X.	ToB	None Selected	Edit Delete

تنظیمات انتقال خط به B

The screenshot displays the ATCOM web interface for editing calling rules. The main window is titled "Edit Calling Rule" and features a sidebar menu on the left with categories like System Status, Configure Hardware, Trunks, and Outgoing Calling Rules. The central area shows a "Manage Calling Rules" section with a "New Calling Rule" button and a list of existing rules. A modal dialog box is open, allowing the user to edit a rule named "toBpstn" with the pattern "_29X". The dialog includes three main sections: "Send to Local Destination" (disabled), "Send this call through trunk:" (active), and "Use FailOver Trunk:" (disabled). In the active section, the "Use Trunk" is set to "ToB", "Strip" is "2", and "Prepend these digits" is "9". The "Use FailOver Trunk" section shows "fail over Trunk" set to "ToB", "Strip" is empty, and "Prepend these digits" is empty. The dialog has "Cancel" and "Save" buttons at the bottom. The browser's address bar shows the URL "http://192.168.16.90/static/config/index.html". The Windows taskbar at the bottom shows the start button and several open applications, including "Edit Calling Rule...", "SEEEEEEEEEEp...", "TeamViewerPort...", "ATCOM IP04-Us...", "eyeBeam", "Document1 - Mic...", and "untitled - Paint". The system clock indicates the time is 1:20 PM.

طرف بعدی IP04 B

کاربران

<input type="checkbox"/>	Extension	Full Name	Port	SIP	IAX	DialPlan	OutBound CID		
<input type="checkbox"/>	200	200	--	Yes	Yes	DialPlan1	200	Edit	Delete
<input type="checkbox"/>	201	--	--	Yes	Yes	DialPlan1	none	Edit	Delete
<input type="checkbox"/>	202	202	--	Yes	Yes	DialPlan1	202	Edit	Delete

Edit User Extension - 200

General :

Extension: Name: DialPlan:

CallerID: OutBound CallerID:

Enable Voicemail for this User

VoiceMail Access PIN code: Mailbox: Email Address:

Technology

SIP IAX Analog Station: flash: rxfash:

Codec Preference : First : Second : Third : Fourth : Fifth :

VoIP Settings

MAC Address : Line Number : SIP/IAX Password:

NAT: Can Reinvite: DTMF Mode: insecure:

Other Options

3-Way Calling In Directory Call Waiting CTI Is Agent

Enable Call Record Pickup Group:

افزودن ترانک ToA

Manage SIP & IAX trunks

Analog Trunks Service Providers VOIP Trunks T1/E1 BRI Trunks

+ New SIP/IAX Trunk

Provider Name	Type	Hostname/IP	Username	
ToA	SIP	192.168.16.90	100	Edit Delete

Edit SIP trunk 100

Provider Name: ToA

Hostname: 192.168.16.90

Username: 100

Fromuser:

Fromdomain: 100

Password: 100

Contact Ext.: s

Insecure Type: very

Codecs: First: u-law Second: a-law Third: GSM

Fourth: G.726 Fifth: None

CallerID:

Enable Remote MWI:

Cancel Add

افزودن Outgoing rule

Manage Calling Rules

+ New Calling Rule Restore Default Calling Rules Outgoing Calling Rules

An outgoing calling rule pairs an extension pattern with a trunk used to dial the pattern. This allows different patterns to be dialed through different trunks (e.g. "local" 7-digit dial through a PRI but "long distance" 10-digit dials through a low-cost SIP trunk). You can optionally set a failover trunk to use when the primary trunk fails. Note that this panel manages only individual outgoing call rules. See the Dial Plans section to associate multiple outgoing calling rules to be used for User outbound dialing.

Calling Rule	Pattern	Trunk	Failover Trunk	
pstn	_9X.	pstnb	None Selected	Edit Delete
ToA	_1X.	ToA	None Selected	Edit Delete

