

FONET opsætning for Grandstream IP-telefon GXP2020.

071008

Via menu-knappen finder du IP-adressen. Eksempel: 192.168.000.104.

Du indtaster så følgende i din browser:

<http://192.168.0.104>. Derved får du følgende billede hvor du indtaster: admin – som password.

Du kommer så ind til nedenstående opsætning. Dine personlige parametre er:

SIP User ID: Brugernavn (findes på 'Mine Sider' under Lokalnumre)

Authenticate ID: Brugernavn (findes på 'Mine Sider' under Lokalnumre)

Authenticate Password: Adgangskode (findes på 'Mine Sider' under Lokalnumre)

Name: Dit navn

Caller ID Scheme: Kan sættes alt efter hvilken telefon du tilslutter

Husk!

Når du har foretaget ændringer, skal du gå nederst på siden og trykke på : Update

Gå derefter ind igen og tryk : Reboot

Grandstream Device Configuration								
STATUS	BASIC SETTINGS	ADVANCED SETTINGS	ACCOUNT 1	ACCOUNT 2	ACCOUNT 3	ACCOUNT 4	ACCOUNT 5	ACCOUNT 6
Account Active:	<input type="checkbox"/> No <input checked="" type="checkbox"/> Yes							
Account Name:	<input type="text" value=""/> (e.g., MyCompany)							
SIP Server:	<input type="text" value="gw 1.fonet.dk"/> (e.g., sip.mycompany.com, or IP address)							
Outbound Proxy:	<input type="text" value=""/> (e.g., proxy.myprovider.com, or IP address)							
SIP User ID:	<input type="text" value="Brugernavn"/> (the user part of an SIP address)							
Authenticate ID:	<input type="text" value="Brugernavn"/> (can be same or different from SIP UserID)							
Authenticate Password:	<input type="text" value="Kodeord"/> (not displayed for security protection)							
Name:	<input type="text" value="Dit navn"/> (optional, e.g., John Doe)							
Use DNS SRV:	<input checked="" type="checkbox"/> No <input checked="" type="checkbox"/> Yes							
User ID is phone number:	<input checked="" type="checkbox"/> No <input checked="" type="checkbox"/> Yes							

<i>SIP Registration:</i>	<input checked="" type="checkbox"/> No <input checked="" type="checkbox"/> Yes
<i>Unregister On Reboot:</i>	<input checked="" type="checkbox"/> No <input checked="" type="checkbox"/> Yes
<i>Register Expiration:</i>	<input type="text" value="60"/> (in minutes. default 1 hour, max 45 days)
<i>local SIP port:</i>	<input type="text" value="5060"/> (default 5060)
<i>SIP Registration Failure Retry Wait Time:</i>	<input type="text" value="20"/> (in seconds. Between 1-3600, default is 20)
<i>SIP T1 Timeout:</i>	<input type="text" value="1 sec"/> ▼
<i>SIP T2 Interval:</i>	<input type="text" value="4 sec"/> ▼
<i>SIP Transport:</i>	<input checked="" type="checkbox"/> UDP <input checked="" type="checkbox"/> TCP
<i>Use RFC3581 Symmetric Routing:</i>	<input checked="" type="checkbox"/> No <input checked="" type="checkbox"/> Yes
<i>NAT Traversal (STUN):</i>	<input checked="" type="checkbox"/> No <input checked="" type="checkbox"/> No, but send keep-alive <input checked="" type="checkbox"/> Yes
<i>SUBSCRIBE for MWI:</i>	<input checked="" type="checkbox"/> No <input checked="" type="checkbox"/> Yes
<i>PUBLISH for Presence:</i>	<input checked="" type="checkbox"/> No <input checked="" type="checkbox"/> Yes
<i>Proxy-Require:</i>	<input type="text"/>
<i>Voice Mail UserID:</i>	<input type="text"/> (UserID for voice mail system)
<i>Send DTMF:</i>	<input type="checkbox"/> in-audio <input checked="" type="checkbox"/> via RTP (RFC2833) <input type="checkbox"/> via SIP INFO
<i>Early Dial:</i>	<input checked="" type="checkbox"/> No <input checked="" type="checkbox"/> Yes (use "Yes" only if proxy supports 484 response)
<i>Dial Plan Prefix:</i>	<input type="text"/> (this prefix string is added to each dialed number)
<i>Delayed Call Forward Wait Time:</i>	<input type="text" value="20"/> (Allowed range 1-120, in seconds.)
<i>Enable Call Features:</i>	<input checked="" type="checkbox"/> No <input checked="" type="checkbox"/> Yes (if yes, call features using star codes will be supported locally)
<i>Call Log:</i>	<input checked="" type="checkbox"/> Log All Calls <input checked="" type="checkbox"/> Log Incoming/Outgoing only (Missed calls NOT recorded) <input checked="" type="checkbox"/> Disable Call Log
<i>Session Expiration:</i>	<input type="text" value="180"/> (in seconds. default 180 seconds)
<i>Min-SE:</i>	<input type="text" value="90"/> (in seconds. default and minimum 90 seconds)

<i>Caller Request Timer:</i>	<input type="checkbox"/> No	<input type="checkbox"/> Yes (Request for timer when making outbound calls)	
<i>Callee Request Timer:</i>	<input type="checkbox"/> No	<input type="checkbox"/> Yes (When caller supports timer but did not request one)	
<i>Force Timer:</i>	<input type="checkbox"/> No	<input type="checkbox"/> Yes (Use timer even when remote party does not support)	
<i>UAC Specify Refresher:</i>	<input type="checkbox"/> UAC	<input type="checkbox"/> UAS	<input type="checkbox"/> Omit (Recommended)
<i>UAS Specify Refresher:</i>	<input type="checkbox"/> UAC	<input type="checkbox"/> UAS (When UAC did not specify refresher tag)	
<i>Force INVITE:</i>	<input type="checkbox"/> No	<input type="checkbox"/> Yes (Always refresh with INVITE instead of UPDATE)	
<i>Enable 100rel:</i>	<input type="checkbox"/> No	<input type="checkbox"/> Yes	
<i>Account Ring Tone:</i>	<input type="checkbox"/> system ring tone <input type="checkbox"/> custom ring tone 1 <input type="checkbox"/> custom ring tone 2 <input type="checkbox"/> custom ring tone 3		
<i>Send Anonymous:</i>	<input type="checkbox"/> No	<input type="checkbox"/> Yes (caller ID will be blocked if set to Yes)	
<i>Anonymous Method:</i>	<input type="checkbox"/> Use From Header	<input type="checkbox"/> Use Privacy Header	
<i>Anonymous Call Rejection:</i>	<input type="checkbox"/> No	<input type="checkbox"/> Yes	
<i>Auto Answer:</i>	<input type="checkbox"/> No	<input type="checkbox"/> Yes	

Allow Auto Answer by Call-Info:

No Yes

Turn off speaker on remote disconnect:

No Yes

Check SIP User ID for incoming INVITE:

No Yes

Refer-To Use Target Contact:

No Yes

*Preferred Vocoder:
(in listed order)*

choice 1:

PCMA

choice 2:

PCMU

choice 3:

G.723.1

choice 4:

G.729A/B

choice 5:

G.726-32

choice 6:

iLBC

choice 7:

G.722 (wide band)

choice 8:

GSM

SRTP Mode:

Disabled Enabled but not forced Enabled and forced

eventlist BLF URI:

Special Feature:

Standard

Update