



GRANDSTREAM

CONNECTING THE WORLD



AGENDA

1

COMPANY OVERVIEW

2

IP SOLUTIONS PORTFOLIO

3

GRANDSTREAM IP PBX UCM6100 SERIES

4

TOTAL UCM SOLUTION

5

GVC3200 VIDEO CONFERENCE SOLUTION

6

IP VIDEO TALK PRO



“ **Founded in 2002**

“ **Over 500 employees**

“ **Product Portfolio contains over 40 products: *Video Conferencing System, 20 different IP Phones, 5 ATAs, 10 IP Surveillance cameras, 5 Video Encoder/Decoders & 2 IP PBXs***

“ **Primarily serving small-to-medium size businesses (SMBs) and consumer markets**

US



Boston - Headquarters
Los Angeles, CA
Dallas, TX

China



Hanghou, Shenzhen
Hong Kong - Warehouse

Morocco



Casablanca - Support Center, EMEA

Venezuela



Venezuela- Support Center, LATM

Netherlands



Moerdijk - Warehouse

VoIP Product Lines

IP Multimedia Phones

Enterprise IP Phones

Small Business / Home Office IP Phones

Analog Gateways & ATAs

IP PBX Appliances

Video Conference System



Surveillance Product Lines

Box IP Cameras

Cube & Mini Dome IP Cameras

IP Video Encoders

Outdoor IP Cameras

Video Management Software

Network Video Recorder



UCM6100 & UCM6510 series IP PBX Appliances



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WWW.GRANDSTREAM.COM

Voice
Secure, clear, and reliable
Conferencing
Comprehensive Codec Support
Endless Customizable Voice features

Data
Phonebook files
CDR
Codec Transcoding
System Backup
Call Recording
Voicemail/fax to email



Mobility
Softphone apps
Monitor your business
Connect Multiple Offices

Video
Face to face video calls
Video Surveillance Integration
SIP Video
Video Codec Support



*Delivering high quality, secure and reliable voice,
video, data & mobility for **SMBs***



- Enterprise-grade features in an affordable, compact & easy-to-manage PBX designed specifically for the SMB market
- No licensing fees
- Fast and easy setup & management
- ALL hardware/software included as well as lifetime firmware updates





UCM6100 series

IP PBX Appliance

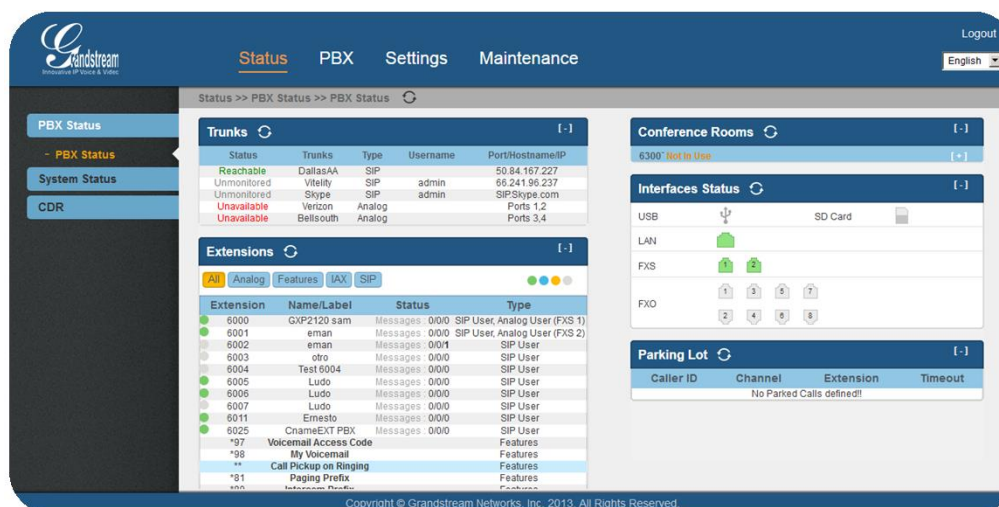


- " 2/4/8/16 FXO trunk port models, 2 FXS ports
- " Up to 60 concurrent calls and up to 32 conference attendees
- " Up to 500 SIP endpoints and up to 50 SIP trunk accounts
- " Dual Gigabit network ports, integrated PoE, USB and SD ports
- " Zero Configuration endpoint provisioning and no licensing fees



General Specifications

- Up to 500 extensions
- # of FXO ports & concurrent calls:
 - 2 - 30 (UCM6102)
 - 4 - 45 (UCM6104)
 - 8 - 60 (UCM6108)
 - 16 - 60 (UCM6116)
- Gigabit ports with PoE Plus
- Each bridge supports up to **25** (UCM6102 & UCM6104) or **32** (UCM6108 & UCM6116) conference attendees
- Zero-configuration provisioning
- Simple setup/management with Web UI

The screenshot displays the Grandstream Web UI for PBX management. The main content area is titled 'Status >> PBX Status >> PBX Status'. It features several panels:

- Trunks:** A table showing trunk configurations.

Status	Trunks	Type	Username	Port/Hostname/IP
Reachable	DallasAA	SIP		50.84.167.227
Unmonitored	Vitelty	SIP	admin	66.241.96.237
Unmonitored	Skype	SIP	admin	SIP:Skype.com
Unavailable	Verizon	Analog		Ports 1,2
Unavailable	Bellsouth	Analog		Ports 3,4
- Extensions:** A table listing individual extensions.

Extension	Name/Label	Status	Type
6000	GXP2120 sam	Messages: 0/00	SIP User, Analog User (FXS 1)
6001	eman	Messages: 0/00	SIP User, Analog User (FXS 2)
6002	eman	Messages: 0/01	SIP User
6003	otro	Messages: 0/00	SIP User
6004	Test 6004	Messages: 0/00	SIP User
6005	Ludo	Messages: 0/00	SIP User
6006	Ludo	Messages: 0/00	SIP User
6007	Ludo	Messages: 0/00	SIP User
6011	Ernesto	Messages: 0/00	SIP User
6025	CNAMEXT PBX	Messages: 0/00	SIP User
*97	Voicemail Access Code		Features
*98	My Voicemail		Features
**	Call Pickup on Ringing		Features
*91	Paging Prefix		Features
- Interfaces Status:** Shows the status of various interfaces: USB, LAN, FXS (ports 1-6), and FXO (ports 1-6).
- Parking Lot:** A table for managing parked calls, currently showing 'No Parked Calls defined!'.

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UCM6510

IP PBX Appliance

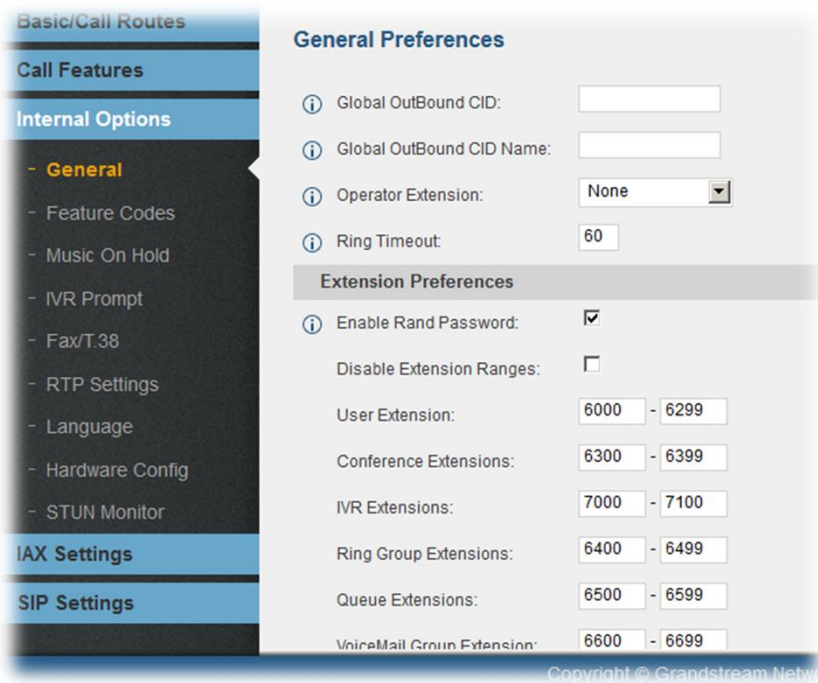


- “ E1/T1/J1 Interface
- “ 2 PSTN trunk FXO ports, 2 FXS ports with lifeline capability
- “ Up to 2000 SIP endpoints and up to 50 SIP trunk accounts
- “ Dual Gigabit network ports, integrated PoE, USB and SD ports
- “ Zero Config endpoint provisioning and no licensing fees



UCM6100 Series - Voice

Delivering high quality, secure and reliable voice communications



The screenshot displays the configuration interface for the UCM6100 Series. The left sidebar shows a navigation menu with categories: Basic/Call Routes, Call Features, Internal Options (with 'General' selected), IAX Settings, and SIP Settings. The main content area is divided into two sections: 'General Preferences' and 'Extension Preferences'. The 'General Preferences' section includes fields for Global OutBound CID, Global OutBound CID Name, Operator Extension (set to 'None'), and Ring Timeout (set to 60). The 'Extension Preferences' section includes a checked checkbox for 'Enable Rand Password', a 'Disable Extension Ranges' checkbox, and several extension range fields: User Extension (6000 - 6299), Conference Extensions (6300 - 6399), IVR Extensions (7000 - 7100), Ring Group Extensions (6400 - 6499), Queue Extensions (6500 - 6599), and VoiceMail Group Extension (6600 - 6699). A copyright notice 'Copyright © Grandstream Network' is visible at the bottom of the interface.

Customizable

- ◆ Call Routing
- ◆ Auto-Attendant
- ◆ IVR
- ◆ Call Forwarding
- ◆ Call Retrieval
- ◆ Music on Hold
- ◆ Transfer
- ◆ Ring Group/Hunt Group

Basic Features	Auto Attendant (IVR)	Call Forward	Black/White List	Attended/Blind Transfer
	Call Record (CDR)	Call Queue	Callback	Music On Hold
	Caller ID	Call Park	Intercom/Paging	Voicemail
	Conference Bridge	Call Pickup	Ring Group	Call Forwarding
	Do Not Disturb (DND)	Call Waiting	Multi-language	ACD
Advance Features	Record Server	LDAP Server	Busy Lamp Field (BLF)	802.1X Access
	Zero Config	Video Codec Negotiation	DID	Mobile Extension
	Firewall/Router	Fax Server	Fax-to-Email	TLS
	Media Security (SRTP)	FXO Automatic Detection	3-way Video Conference	Billing Interface (3rd party)



Conference Server

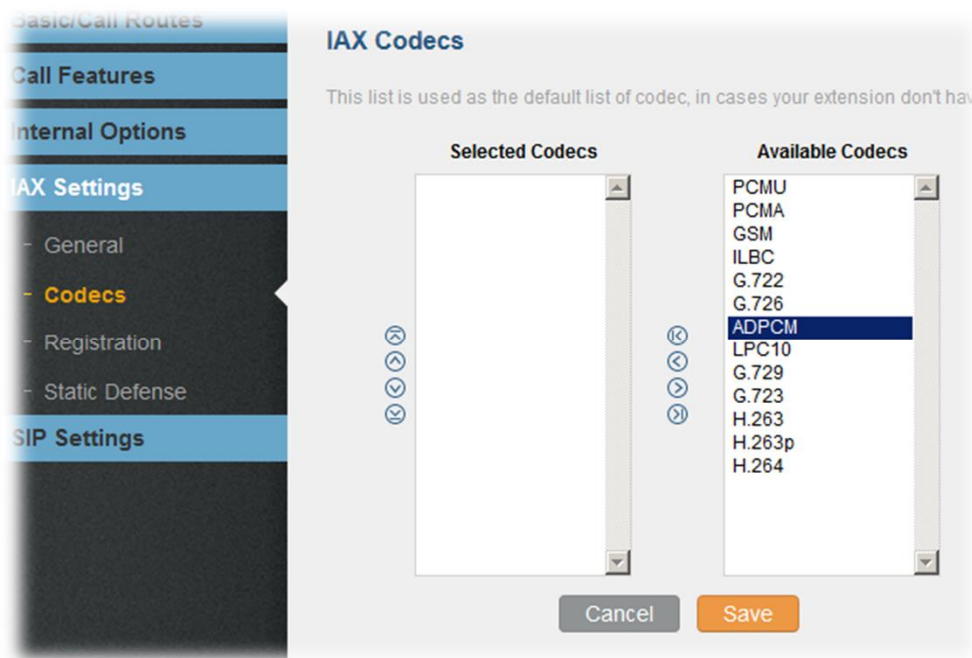
- “ Built-in conference bridge feature on UCM6100 series and UCM6510
- “ Support invitation via conference administrator or web GUI
- “ Support PIN code to access conference bridge
- “ Conference IVR menu options include inviting other parties, mute, lock conference room, kick off participants and etc.

	UCM6102/6 104	UCM6108/6 116	UCM6510
Max. Number of Conference Bridge	3	6	8
Max. Number of Participants	25	32	64



UCM6100 Series - **Voice**

Delivering high quality, secure and reliable voice communications



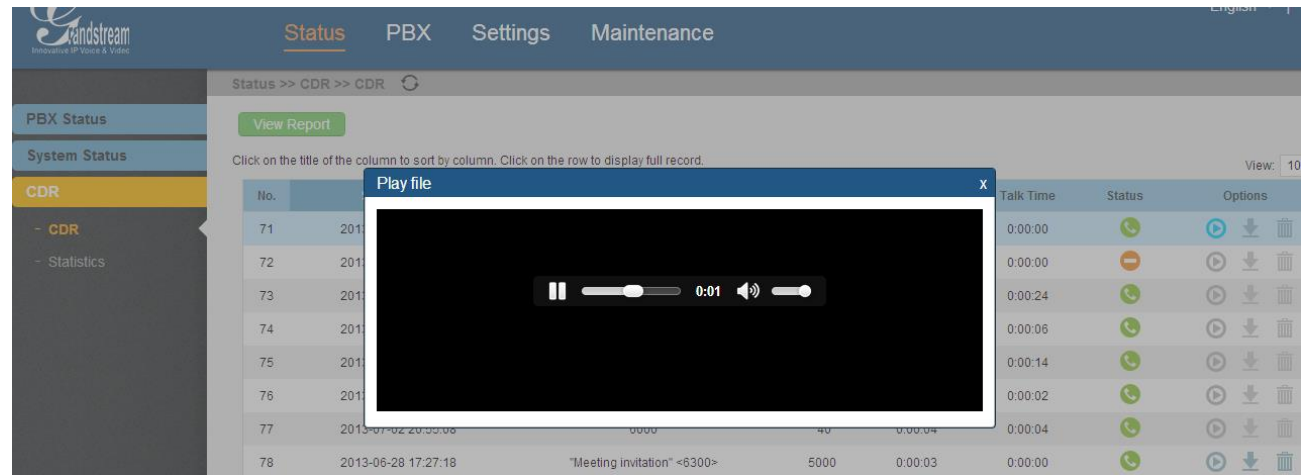
Supports all major voice codecs

- ◆ G.711
- ◆ **G.722 (HD audio)**
- ◆ G.723.1
- ◆ G.726
- ◆ **G.729**
- ◆ **iLBC**
- ◆ GSM
- ◆ PCMU
- ◆ PCMA
- ◆ GSM
- ◆ ADPCM
- ◆ LPC10



Recording Server

- Support 500,000 minutes call recording in built-in memory
- Recording length can be extended via USB flash drive
- Recording file can be played and downloaded from UCM6100 series or UCM6510 web GUI
- Automatic recording (coming soon)




The screenshot displays the Grandstream web GUI interface. The top navigation bar includes 'Status', 'PBX', 'Settings', and 'Maintenance'. The main content area shows 'Status >> CDR >> CDR' with a 'View Report' button. A table lists call records with columns for 'No.', 'Talk Time', 'Status', and 'Options'. A 'Play file' dialog box is overlaid on the table, showing a play button, a progress bar at 0:01, and a volume icon.

No.	Talk Time	Status	Options
71	0:00:00	📞	🔍 ⬇️ 🗑️
72	0:00:00	📞	🔍 ⬇️ 🗑️
73	0:00:24	📞	🔍 ⬇️ 🗑️
74	0:00:06	📞	🔍 ⬇️ 🗑️
75	0:00:14	📞	🔍 ⬇️ 🗑️
76	0:00:02	📞	🔍 ⬇️ 🗑️
77	0:00:04	📞	🔍 ⬇️ 🗑️
78	0:00:03	📞	🔍 ⬇️ 🗑️



Security

- ~ Embedded system design with system stability and antivirus capabilities
- ~ Network layer security: LLDP, 802.1x
- ~ TLS and SRTP support
- ~ Embedded firewall with static defense and dynamic defense
- ~ Blacklist/Whitelist support
- ~ AES-128 Encryption Algorithm

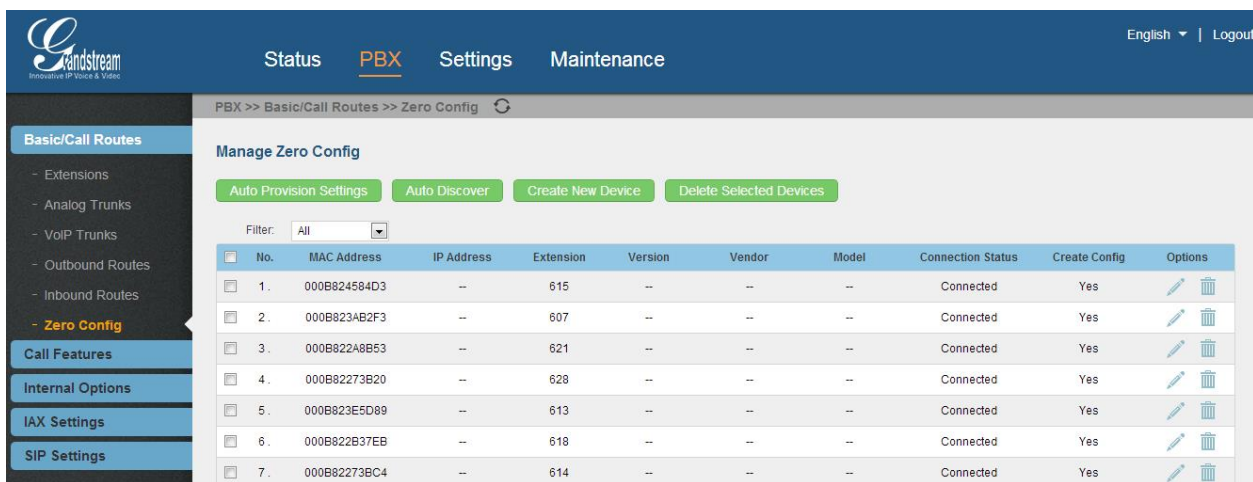
Network Layer	Support LLDP on data link layer authentication; Support 802.1X authentication for network access; Support firewall/ACL access control.
System Configuration	Inbound route/outbound route with privilege assigned; Log printout; Automatic system backup.
Application 	Support HTTPS for secure remote access; Support AES-128 encrypted data transmission; Random generated password for SIP extensions; Support authentication in SIP registration and calls; Support HTTP Digest (MD5), SIP Digest Algorithm; Support TLS Signaling encryption (UCM6510 to end-point); Support SRTP real-time media encryption (UCM6510 to end-point).

















Grandstream IP PBX - Auto-Provisioning

End-point Auto Provisioning

- Support auto-provisioning for Grandstream devices
- Support real-time configuration by refreshing the extension in UCM6100 series or UCM6510 web page
- Support Plug-and-Play (PnP) and batch configuration



The screenshot shows the Grandstream IP PBX web interface. The top navigation bar includes 'Status', 'PBX', 'Settings', and 'Maintenance'. The main content area is titled 'Manage Zero Config' and features several action buttons: 'Auto Provision Settings', 'Auto Discover', 'Create New Device', and 'Delete Selected Devices'. Below these buttons is a table with columns for 'No.', 'MAC Address', 'IP Address', 'Extension', 'Version', 'Vendor', 'Model', 'Connection Status', 'Create Config', and 'Options'. The table contains seven rows of device information, all with a 'Connected' status.

No.	MAC Address	IP Address	Extension	Version	Vendor	Model	Connection Status	Create Config	Options
1.	000B824584D3	--	615	--	--	--	Connected	Yes	 
2.	000B823AB2F3	--	607	--	--	--	Connected	Yes	 
3.	000B822A8B53	--	621	--	--	--	Connected	Yes	 
4.	000B82273B20	--	628	--	--	--	Connected	Yes	 
5.	000B823E5D89	--	613	--	--	--	Connected	Yes	 
6.	000B822B37EB	--	618	--	--	--	Connected	Yes	 
7.	000B82273BC4	--	614	--	--	--	Connected	Yes	 



UCM6100 Series - **Video**

Delivering high quality, secure and reliable video communications



Delivering high quality, secure and reliable video communications



Video Calling

- Any SIP video endpoint
- See customers and employees through real-time, face-to-face video communication
- When used with Grandstream video phones, create your own multi-user video conference



Delivering high quality, secure and reliable video communications

Video Surveillance Integration



- ◆ Receive alerts from anywhere
- ◆ Create a comprehensive solution to monitor and receive alerts from IP cameras
- ◆ Register IP cameras to the PBX
- ◆ Make video calls to IP cameras to view live feeds
- ◆ Speak through cameras with 2-way audio & video (door entry)
- ◆ IP cameras can be set to automatically call video phone when alert is triggered



UCM6100 Series - **Data**

Delivering high quality, secure and reliable data communications



IP PBX Features

All-In-One Design

" FXS/FXO gateway, low cost and high stability

LCD Menu

" Easy installation and web access via IP address

Easy Configuration

" End-point deployment via Zero-Config and PnP

LDAP Server

" For daily office communication purposes

Recording/Fax

" 50,000 minutes of recording, 20,000 digital fax files

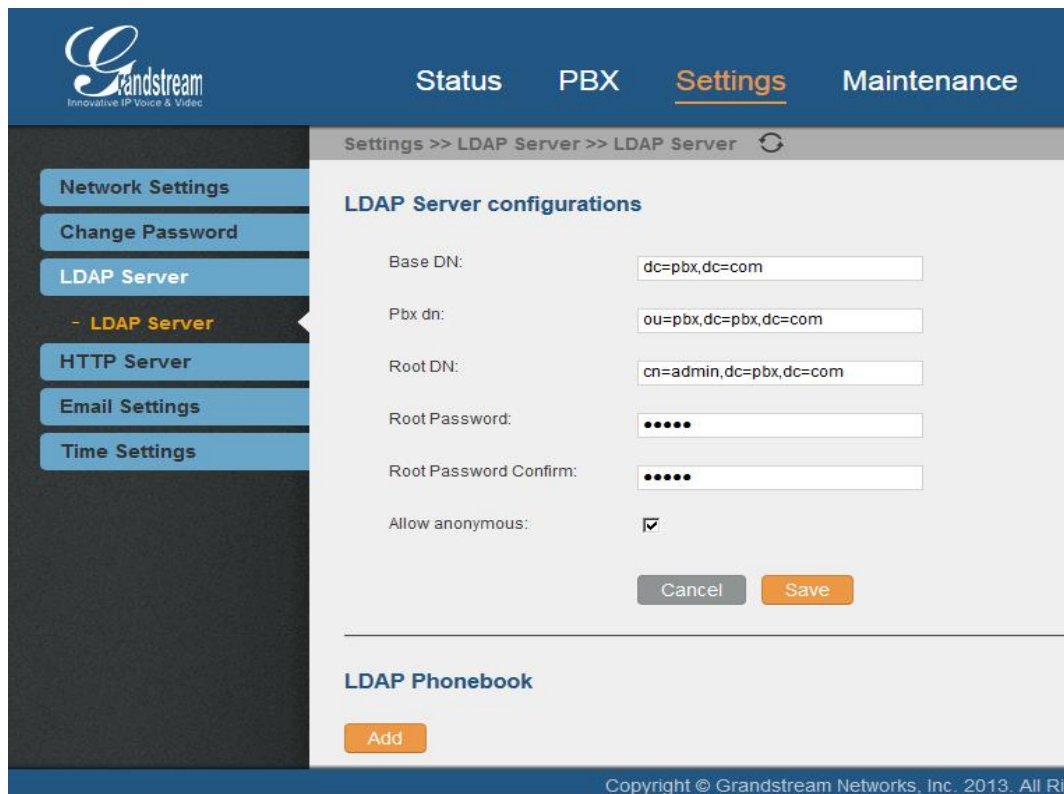
High Reliability

" ITU-T K.21 standard
" Telecom industry level IP PBX testing procedure



EmLDAP Sever edded LDAP Server

- Support 1000 phonebooks
- Default LDAP Phonebook generated for UCM series extensions
- Search contact number via department or name
- Grandstream phones can be auto-provisioned to use the UCM6100 series or UCM6510 LDAP server default phonebook



The screenshot displays the Grandstream web interface for configuring LDAP Server settings. The navigation menu includes Status, PBX, Settings (selected), and Maintenance. The breadcrumb trail is Settings >> LDAP Server >> LDAP Server. The main content area is titled "LDAP Server configurations" and contains the following fields:

Base DN:	<input type="text" value="dc=pbx,dc=com"/>
Pbx dn:	<input type="text" value="ou=pbx,dc=pbx,dc=com"/>
Root DN:	<input type="text" value="cn=admin,dc=pbx,dc=com"/>
Root Password:	<input type="password" value="....."/>
Root Password Confirm:	<input type="password" value="....."/>
Allow anonymous:	<input checked="" type="checkbox"/>

Buttons for "Cancel" and "Save" are located below the form fields. Below the configuration section, there is a section titled "LDAP Phonebook" with an "Add" button. The footer of the interface reads "Copyright © Grandstream Networks, Inc. 2013. All Rig".



Grandstream IP PBX - FXS ports for Analog Phone & Fax

1. Two FXS
Ports

Network management

2. T.38 Fax over
IP

Low cost

3. Over-voltage/over-
current protection
Lifeline capability during
power outage

Easy installation and
deployment

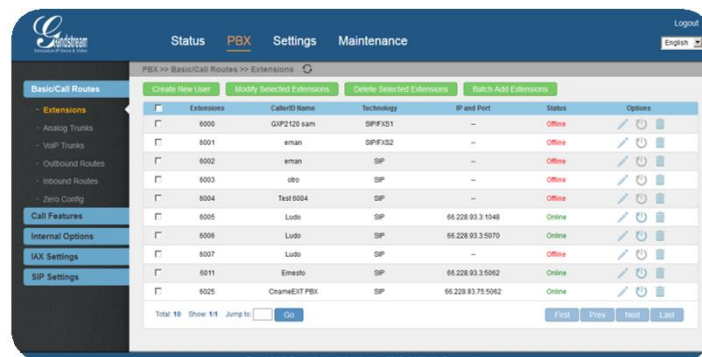


Fax Server

- “ 2 FXS ports
- “ T.38 support
- “ Fax can be converted to PDF file, and sent out via Email
- “ 20,000 digital Fax files
- “ Performance test done by Spirent testing equipment to guarantee stability



Fax



Extension	CallerID Name	Technology	IP and Port	Status	Options
6000	GMP120 sam	SIPFXS1	-	Offline	/ [Power] [Mute]
6001	aman	SIPFXS2	-	Offline	/ [Power] [Mute]
6002	aman	SP	-	Offline	/ [Power] [Mute]
6003	oto	SP	-	Offline	/ [Power] [Mute]
6004	Test 0004	SP	-	Offline	/ [Power] [Mute]
6005	Ludo	SP	66.228.93.1048	Online	/ [Power] [Mute]
6006	Ludo	SP	66.228.93.5070	Online	/ [Power] [Mute]
6007	Ludo	SP	-	Offline	/ [Power] [Mute]
6011	Emesto	SP	66.228.93.5062	Online	/ [Power] [Mute]
6025	ChameEXT PBX	SP	66.228.93.75.5042	Online	/ [Power] [Mute]



Email



Voicemail



*Delivering high quality, secure and reliable **mobility***



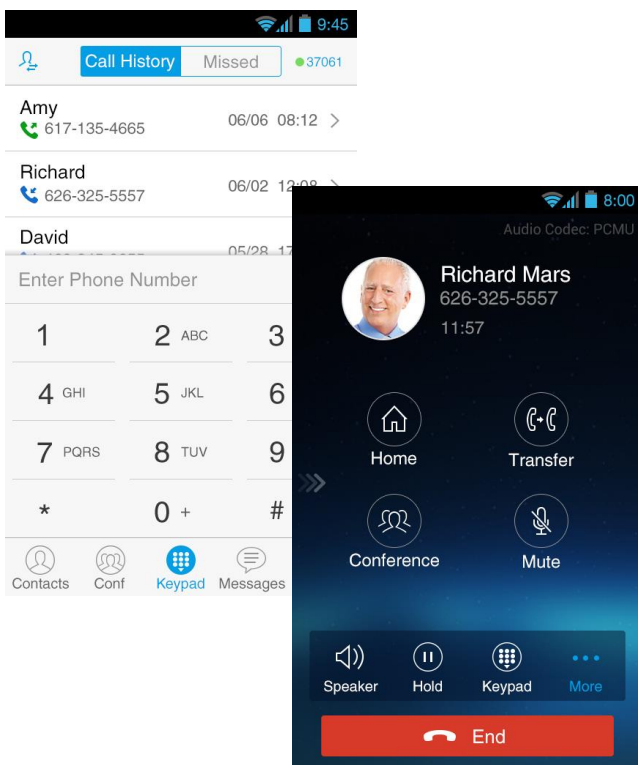
Delivering high quality, secure and reliable mobility



Monitor your business from anywhere

- View live feeds & receive alerts from IP cameras on any device with an internet connection
- Speak through IP cameras

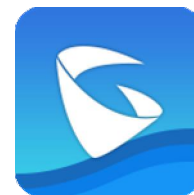




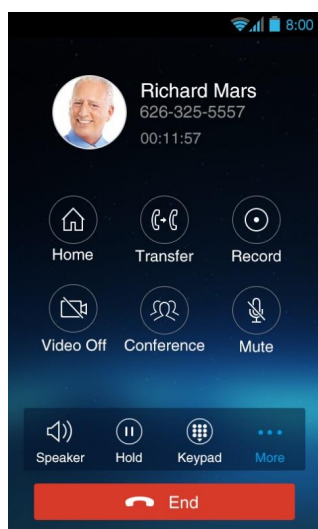
Grandstream Wave

Softphone Application For Android™

- “ Requires Android 4.0+
- “ Supports 6 SIP accounts, 6-way audio conferencing
- “ Features include call transfer, LDAP phonebook, virtual BLF keys and more
- “ Native integration with mobile devices including contacts, call history, and ringtones



Delivering high quality, secure and reliable mobility



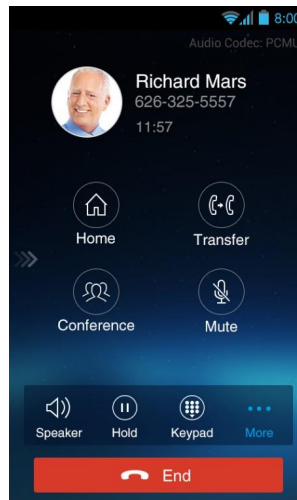
Make and receive calls on a smartphone & laptop

- Compatible with SIP smartphone & computer applications, including Grandstream Wave, Bria, and Zoiper
- Utilize the extension for the user, rather than their desk at the office
- Supports both video and audio calls



BYOD/Mobile Office SIP End-point Compatibility

- “ 3rd party softphone can register to UCM6100 series and UCM6510
- “ From softphone installed on your mobile devices, you can check Fax and voicemail remotely
- “ Support DECT cordless IP phone to register to the system



Grandstream Wave



DP715



DP710

SIP Softphone

Grandstream Wave, 3CX, EyeBeam, Bria, Zoiper, or other 3rd party softphone (many are free) supported on Android™, IOS or other mobile platform



GVC3200 Video Conferencing System



This Changes Everything



GVC3200 Video Conferencing System



The GVC3200 combines the power of SIP, the expandability of a cloud platform and the unrivaled flexibility of Android

- “ **Easy Installation** . This open standard SIP system requires a 2-step installation without any extra/proprietary equipment
- “ **Easy to Use** . Plug and play connection laptops for screen-sharing
- “ **Easy Interoperability** . GVC3200 can communicate with third-party SIP video conferencing platforms as well as utilize Android apps
- “ **State-of-the-art hardware** . No external MCUs or on-site servers required, as the MCU is built-in to the camera unit
- “ **Advanced Video Support** . 1080p Full-HD video, support for 3 monitor outputs and a PTZ camera with 12x zoom
- “ **Introduce video conferencing to your business or expand your current SIP platform with the GVC3200**



Powered by Android™

- “ Android 4.4 based . with full access to the ever expanding Google Play Store
- “ Custom applications available through the Android SDK
- “ Support for popular video conferencing applications (via Android), including: Skype for Business, Skype, Google Hangouts, GoToMeeting, BlueJeans & many more



GVC3200 Video Conferencing System

*Video Conferencing Has Never Been This **Easy***



Video	1080p resolution – 4 way video 720p resolution– 5 way video VGA resolution – 9 way video
Camera	PTZ camera (controlled by remote control) Full 1080p HD (1920x1080) @ 30fps
Lens	12x optical zoom
Network	1 Gigabit port, integrated WiFi
Bluetooth	Yes, integrated
Video Inputs	1 HDMI and 1 VGA
Video Outputs	3 HDMI
Mic/Speaker	Includes external Mic/speaker Microphone built-in to camera unit
Remote Control	Bluetooth remote control with multi-touchpad
Video Codecs	H.264
Application Bridging	Bridges SIP calls with any Android apps (Skype, Hangouts, etc.)



Easy Video Conferencing

The GVC3200 is the ideal Video Conferencing Solution for Small Businesses

Small Businesses



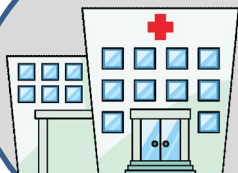
- ✓ Keep in touch with clients
- ✓ Impress perspective clients
- ✓ Keep in touch with vendors/suppliers
- ✓ Keep in touch with remote employees

Schools

- ✓ Hold multi-classroom events
- ✓ Video conference with other schools
- ✓ Bring in guest-speakers through video conferencing
- ✓ Meet with outside specialists
- ✓ Video conference with parents



Doctor's Offices



- ✓ Call other doctors
- ✓ Consult through video conferences
- ✓ Meetings/consultations with specialists, hospitals, Emergency Rooms, etc.

- ✓ **Easy Installation**
- ✓ **Easy to get Stated**
- ✓ **Easy Interoperability**
- ✓ **Up to 9-way video**

Small Law Offices

- ✓ Keep in touch with clients
- ✓ Hold face-to-face meetings with perspective clients
- ✓ Meet with remote or travelling teams



Video Conferencing has never been this *Easy*



Easy to Install

-Just connect the GVC to HD monitors, internet & power...that's it!



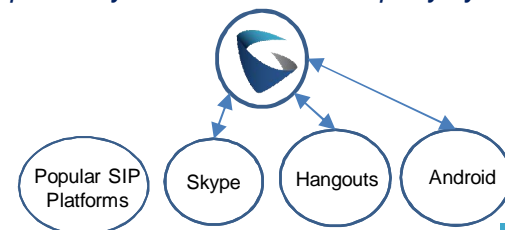
Easy to Get Started

-Full access to our cloud platform, IPVideoTalk Pro



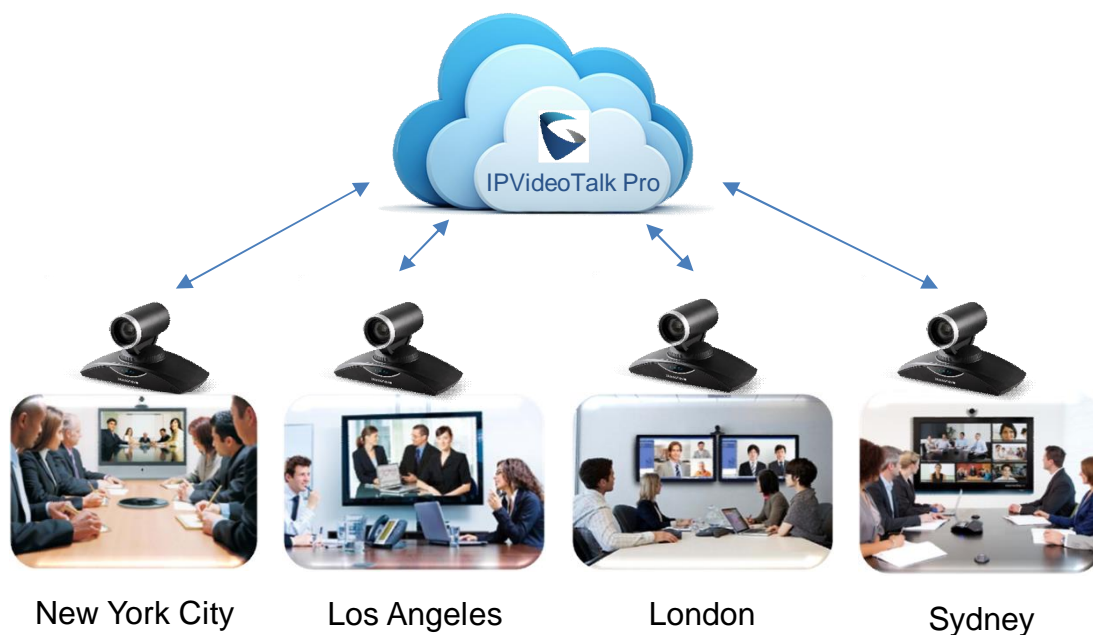
Easy Interoperability

- Never be locked into one vendor*
- Interoperable with popular 3rd party SIP platforms*
- Expand any current or future 3rd party system*



Easy to Get Started and *Easy* to Use

Cloud Video Conferencing with No Initial Contracts



The power of our
IPVideoTalk Pro
cloud video conference platform allows
plug and play video conferencing so you
can easily keep in touch with clients,
customers and co-workers worldwide.



Easy Interoperability

Use the GVC3200 in a Variety of Ways



- “ No vendor lock-in
- “ Fully interoperable with popular third-party SIP platforms
- “ GVC3200 hardware built on open standards (SIP)
- “ Expand your current or future third-party system
- “ Communicate with any SIP-based or Android platform



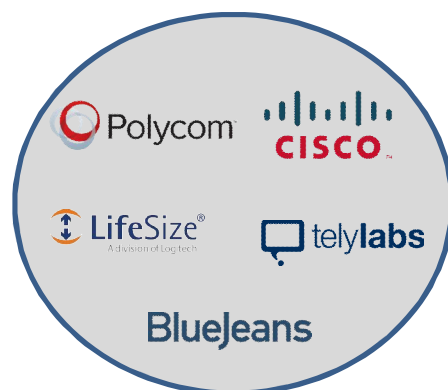
Easy Interoperability

Use the GVC3200 in a Variety of Ways



1. IPVideoTalk Pro

- Use with Grandstream's cloud service for plug and play use to experience up to 9-way video conferencing



2. Third-Party SIP Systems

- Use to communicate with or expand any SIP-based third-party video conferencing solution from any vendor



3. Android apps

- The GVC3200 is the only video conferencing system currently on the market that runs the Android operating system. This allows it to make video conferences using any Android app, including Skype, Google Hangouts and more

Extension 102

4. Video Conference Bridge

- Add the GVC3200 as an extension on your IP PBX or hosted/cloud PBX for use as a video conferencing bridge



Easy Interoperability

Use the GVC3200 in a Variety of Ways

Interoperable with a wide variety of video conferencing devices and applications, including:



- “ Direct IP calls between multiple GVC3200s
- “ Video calls between GVC3200 and SIP video phones such as the GXV3240 and GXV3275
- “ Video calls with Android applications such as Skype for Business, Google Hangouts, Skype & GoToMeeting
- “ Video calls between other VCS, such as Cisco, Polycom and LifeSize · via H.264 standard
- “ Bridge video calls from multiple VCS platforms and Android applications, all in once conference*

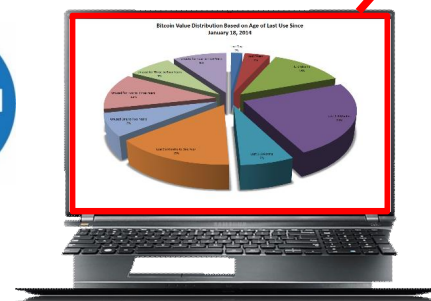
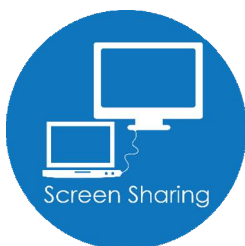


Easy and Powerful Video Conferencing

Content Sharing

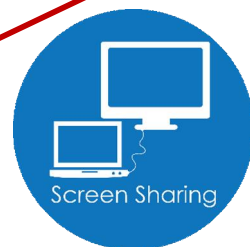
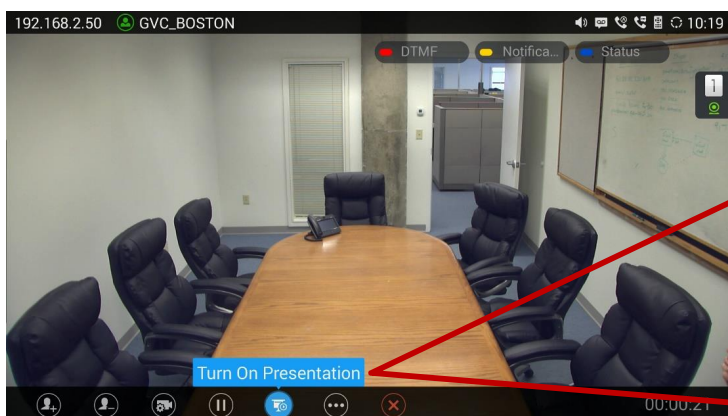
Share any computer or tablet screen with other attendees:

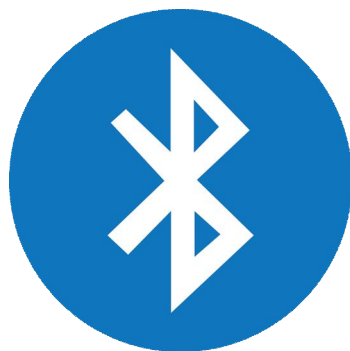
- ” Show presentations
- ” Review and markup documents
- ” View web content
- ” 1 HDMI and 1 VGA input



Easy Screen Sharing for Enhanced Collaboration

The GVC3200 features a VGA/HDMI input to connect a presentation and screen share with conference attendees directly through the video stream.





Bluetooth Remote

Included is a Bluetooth touchpad remote .
featuring an easy to use Android interface

- " Bluetooth 4.0 compliant
- " 3-point touch pad
- " 5 navigation keys, 3 Android keys, 4 functional keys, 3 multi-purpose keys
- " 0-9, #, * DTMF keys
- " Supports OTA firmware upgrade
- " Powered by 2 AA batteries

IPVideoTalk

Transform Your Video Conferencing Experience

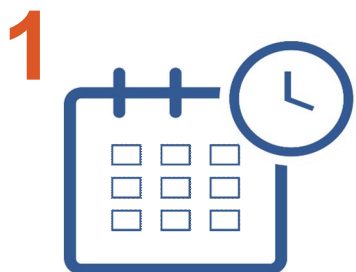


- Cloud based video conference system to be used with Grandstream's GVC3200, GXV32xx series of video phones, Grandstream Wave, WebRTC-capable browsers, and any phone
- Hold meetings and conferences with up to 100 attendees, 8 presenters, HD audio and video, and screen sharing
- Includes meeting scheduler, recordings, conference room integration and more



IPVideoTalk

Unified Video Conferencing & Web Collaboration Service



Plan or Start a Meeting

Schedule a future meeting, set a recurring meeting, or meet right away directly from the IPVideoTalk online interface



Invite Participants

Participants can join from Grandstream® GVC3200, GXV32xx series of video phones, WebRTC-capable browsers, or any phone



Meet & Collaborate

Host meetings with up to 100 attendees and 8 presenters with features such as HD audio & video, and screen sharing



IPVideoTalk Features



Supports All Devices



Full HD Video Quality



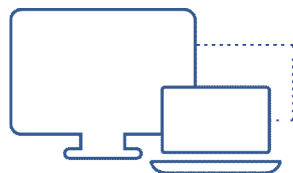
Full HD Audio Quality



Join a Meeting in 1-Click



Built-in Scheduler



Screen Sharing



Conference Room
Integration with GVC3200



Recording



Supports All Devices



GVC3200 series
Video Conference System



PC or Mac



WebRTC
Supported Browser



GXV32xx series
Android Video Phones



Tablet or Smart Phone

Conference Room Integration



Use Grandstream's GVC3200 series Video Conference Solution to integrate IPVideoTalk with your conference room

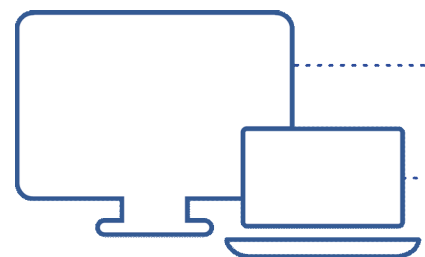
- Host video conferences through the cloud with IPVideoTalk
- Allow remote employees to join the conference from their PC, MAC or mobile device



Screen Sharing



Share content and presentations with attendees through the screen sharing feature



Schedule a Meeting

IPVideoTalk | Grandstream Networks, Inc.

Meetings | Contacts | Devices | Plans | System

Schedule Meeting +

Upcoming 1
History 1033

Schedule Meeting

* Subject: 1-80 characters

Time Zone: (GMT-05:00) Eastern Time (US and Canada) ▼

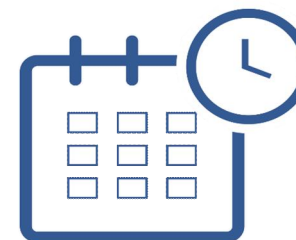
* Start Time: 12/07/2015 ▼ 04 ▼ : 00 ▼ PM ▼

* End Time: 12/07/2015 ▼ 05 ▼ : 00 ▼ PM ▼


* Host Device: Enter Linked IPVT Account/Internal Number/Alias ▼

Participants: Enter Participant Email/IPVT Account

Use the built-in scheduler to schedule meetings ahead of time, invite participants, and set reminders.



Join a Meeting in 1-Click



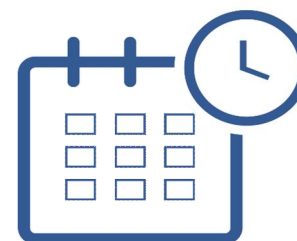
IPVideoTalk

*Improve Business
Productivity with Full HD
Video Conferences that
can be joined from
anywhere*

Join A Meeting

Joining a meeting is as easy as one click.

- Join from an email invitation with one click, or enter the meeting ID, your name and email, and join immediately





Thank you!

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