











- Founded in 2002
- Over 500 employees
- Product Portfolio contains over 40 products: Video Conferencing System, Audio Conference Phone, 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



WWW.GRANDSTREAM.COM



Awards & Recognition

Global Presence



15 time winner





9 time winner



BEST

2014 BUSINESS WORLI

2 time winner





Over 10 Years of Growth and Innovation...

WWW.GRANDSTREAM.COM 4



Product Interoperability

IP-PBX and Softswitch





Product Interoperability









Grandstream's Long History of Open Source









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GXP1700 Series Mid-Range IP Phones

GDS3710 HD IP Video Door System

GWN7600 Wireless Access Point

GWN7000 Wireless Router





GXP1760 Mid-Range IP Phone

- 200 x 80 pixel back-lit LCD display
- Digital BLF Keys, 24 programmable keys
- 6 lines, 3 SIP accounts
- 5-way conferencing
- Automated provisioning
- Integrated PoE, USB Port, and EHS support









GXP1780/82 Mid-Range IP Phone

- 200 x 80 Pixel Back-Lit LCD Display
- Digital BLF Keys, 32 Programmable Keys
- 8 lines, 4 SIP Accounts
- 5-way Conferencing
- Automated provisioning
- Gigabit Network Ports on GXP1782 Only
- Integrated PoE USB Port, and EHS Support

broadsoft (Pending)







GWN7600

Entry Level Wireless Access Point

- Supports more than 250+ WiFi client devices.
- 1.167Gbps wireless throughput and 2x Gigabit wireline ports.
- 175-Meter coverage range for superior device mobility.
- Dual-band 2x2:2 MIMO technology
- Advanced Wi-Fi security features, including WEP, WPA/WPA2-PSK, WPA/WPA2-Enterprise
- Offers simultaneous dual band Wi-Fi signals.
- Embedded controller for auto-discovery, autoprovision, and management of up to 25 GWN7510s.







GWN7000

Enterprise Multi-WAN Gigabit VPN Router

- Embedded Provisioning. Can control up to 500 GWN76xx WiFi access points.
- Seven Gigabit ports, 2 WAN and 5 LAN.
- Hardware accelerated VPN including PPTP, L2TP/IPSec and OpenVPN with speeds up to 300Mbps.
- 1M packet/sec routing & 10Gbps aggregate switching power.
- 3G/4G-LTE wireless uplink via USB dongle.
- Web interface to centrally monitor/provision the entire network









GDS3710 HD IP Video Door System

- HD Resolution up to 1080p, 30 frames per second.
- Built-in RFID chip reader for keyless entry.
- SIP/VoIP support for streaming of video feeds to multiple smartphones, sip endpoints and software simultaneously.
- Hemispheric camera for 180 degree wall-to-wall coverage, supports motion detection.
- Weatherproof, vandal resistant, PoE enabled, built in microphone and speaker for intercom functionality.





NEW PRODUCTS









HT802/812/814 Analog Telephone Adapters

UCM6200 IP PBX Appliance



GWN7610 Wireless Access Point



New



GWN7610

Mid-Range Wireless Access Point

- Supports more than 250+ WiFi client devices.
- 1.75Gbps wireless throughput and 2x Gigabit wireline ports.
- 175-Meter coverage range for superior device mobility.
- Dual-band 3x3:3 MIMO technology
- Advanced Wi-Fi security features, including WEP, WPA/WPA2-PSK, WPA/WPA2-Enterprise
- Offers simultaneous dual band Wi-Fi signals.
- Embedded controller for auto-discovery, autoprovision, and management of up to 50 GWN7510s.





New

UCM6200 series IP PBX Appliance



- Up to 100 concurrent calls and up to 32 conference attendees
- Up to 800 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









New

HT802 Analog Telephone Adapter

- Supports 2 SIP profiles, with 2 FXS ports and one 10/100Mbps port
- Supports 3-way voice conferencing.
- T.38 Fax support for creating Fax-over-IP
- Advanced Telephony features, such as call transfer, multi-language prompts, flex dial plan, and more.
- Automated provisioning options include TR-069 and XML config files.









HT812



HT812 & HT814 FXS Port Analog Adapter

- HT812 2 FXS ports with 2 SIP profiles
- HT814 4 FXS ports with 4 SIP profiles
- Dual Gigabit network ports and integrated NAT router
- Features T.38 Fax, 3-way audio conferencing, and a wide range advanced telephony features
- Automated provisioning options including TR-069 and XML config files
- Supports ZeroConfig to use with the UCM series of IP PBXs



Networks (Pending)





GVR3550

Network Video Recorder

- Up to 24 channels 720p HD, or 16 channels of 1080p FHD video/audio real-time recording
- Up to 4 SATA 3.5" internal hard drives with maximum 16TB storage and RAID 0 or RAID 1 configurations
- 1 HDMI and 1 VGA output, as well as USB, eSATA and alarm in/out ports
- Compatible with all Grandstream surveillance devices, as well as all ONVIF compliant IP cameras









GVR3552 Network Video Recorder

- Up to 8 channels 1080p FHD, or 16 channels of 720p HD video/audio recording
- Up to two 2.5" internal hard drives with up to 4TB storage
- HDMI, USB and VGA interfaces
- Compatible with all Grandstream surveillance products, and all ONVIF compliant IP cameras (pending)







GXV3672 series 720p/1080p Outdoor Tube Camera

- GXV3672_36: 1.2MP, 3.6mm lens
- GXV3672_FHD: 3.1M camera, 8mm lens
- H.264 ONVIF compliant infrared
- Day/night mode with light sensor and IR_CUT filter







GXV3674 HD/FHD 720p/1080p Outdoor Tube Camera

- 2.8mm-12mm Vari-focal lens, 1.2MP
- H.264 ONVIF compliant infrared 720p HD, 1080p FHD
- IP66 weatherproof









GXV3610 HD/FHD 720p/1080p Outdoor Dome Camera

- H.264 ONVIF compliant infrared 720p HD, 1080p FHD
- 1.2MP, 3.6mm lens
- IP66 weatherproof
- Day/night mode



IP Surveillance





GXV3610FHD 1080p Outdoor Dome Camera

- IP66 vandal & weather resistant
- H.264 ONVIF compliant
- GXV3662_HD: 1.2MP, 3.3-12mm lens
- GXV3662_FHD: 3.1MP, 3.3-12mm lens
- Integrated PoE









GXV3611IR_HD

Indoor Infrared Fixed Dome HD IP Camera

- Integrated IR-CUT for day/night mode
- Integrated microphone and speaker
- H.264 ONVIF compliant HD (720p)
- 2MP, 2.8mm lens
- Integrated PoE
- Supports Time-lapse photography







GXV3500 Video Encoder/Decoder

- 1 channel video encoder or decoder
- Rich peripheral interfaces
- H.264 ONVIF compliant



Surveillance Features

Define the phone number to call from the camera

Phone List		
Phone Number	Remark Name	Remove
		Add
102	VideoPhone	
	Check All	Remove
Note: You must restart t	he device to apply the changes.	

IP Camera SIP Settings

	General Phone Settings		
•	Registered:	Online	
Settings	Unregister On Reboot:	V	
m			
& Audio	SIP Settings		
rking	Account Name:	104	0
	SIP Server:	192.168.60.2	0
	Outbound Proxy:		0
	SIP User ID:	104	θ
iced Settings	Authenticate ID:	104	0
	Authenticate Password:		0
	STUN Server:		0
	Stream:	Secondary 💌	
	Preferred Vocoder:	PCMU 💌	
	Register Expiration(Second):	3600 0	
	local SIP Port:	5060	
	local RTP Port:	5004	
	Auto on Hook Timer:	300 0	
	Disable Audio in SIP Call:	0	
	Enable Keep Alive		
	Accept Direct IP Call:	V	
	Enable White List Number Filter:	0	
	SIP Proxy Compatibility Mode:	0	
	Enable Dual-way Audio Warning Mode:	V 0	
	Note: You must restart the device	e to apply the changes.	





Scenario 1: Smoke Detection (GXV3611IR_HD only)

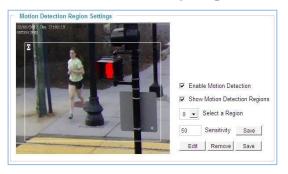
	Alarm Event	+ DI Alarm Time Sch	nedule		
	Motion Detection Syslog	Record Video From Voice Alarm to SIP Alarm Output Upload to Alarm Cer Upload to Alarm HT Email and FTP uploe	configured extension		
Set the time when Smoke alarm will be enabled. In this example, it's enabled 24/7.		Save Smoke Alarm Time Date Everyday Everyday	e Schedule Start Time 00:00 00:00	End Time 23:59 23:59 Check All	Add Delete





Scenario 2: Motion Detection (GXV3611IR_HD only)

Step1: Enable Motion detection and setup regions



Step2: Enable Motion detection and setup regions

Alarm Action
Record Video From Pre Alarm 0 v seconds to After Alarm 1 v seconds
Voice Alarm to SIP Phone
Upload to Alarm Center
Upload to Alarm HTTP Server
Email and FTP upload JPEG
Save

Note: When **Upload to Alarm Center** is checked, the Motion Detection region and icon (in Home page) will blink in red if your camera detects any motion.

Step3: Setup Motion detection Time Schedule

- Motion De	tection Time Schedu	е		
Region ID	Date	Start Time	End Time	
0 💌	Everyday 💌	00:00	23:59	Add
		E		



Scenario 3: Open Door Feature (DTMF)

ave		
SIP Open Door Settings		
Enable SIP Open Door:		
Key to Open the Door:	••••	
Delay Lock Time(Second):	1 💌	
Save		
Phone List		
Phone Number	Remark Name	Remove
6179114110	Local Police	Add
Note: You must restart the o	levice to apply the changes.	
Note: You must restart the o	levice to apply the changes.	
	levice to apply the changes.	Remove
White List	levice to apply the changes.	Remove Add

- Enable SIP Open Door: This will close the DO connections in the back to close the circuit connected to it.
- Key to Open the Door: DTMF key (Currently only SIP INFO DTMF is supported) to close the circuit. Digits 0-9 only.
- **Delay Lock Time(Second):** This is the time in seconds that the circuit will remain closed when this function is triggered
- Phone List/Phone Number: Callee or call receiver number when alarm call trigged.
- White List/Phone Number: Phone numbers allowed to call into the camera.

Notes:

- OpenDoor feature is not available in following models: GXV3615/GXV3615WP_HD/GXV3672HD/FHD
- OpenDoor feature can be used in combination with Alarm IN/OUT using a Door Bell connected in Alarm IN and Door lock connected in Alarm OUT





Scenario 4: Alarm In

Step1: Configuring Alarm Input on the camera







Scenario 4: Alarm In

Step2: Enable Alert Event

Configuration > Advanced Settings > Alert Event

Alarm Action	
Record Video From Pre Alarm 0 vectors to After Alarm 1 vectors	
Voice Alarm to SIP Phone	
Alarm Output	
Upload to Alarm Center	
Upload to Alarm HTTP Server	
Email and FTP upload JPEG and Store it to SD driver	

Step3: Configure Alarm Schedule

Configuration > Advanced Settings > Alert Event

- Alarm Event Ti	me Schedule		
Date	Start Time	End Time	
Everyday 💌	00:00	23:59	Add

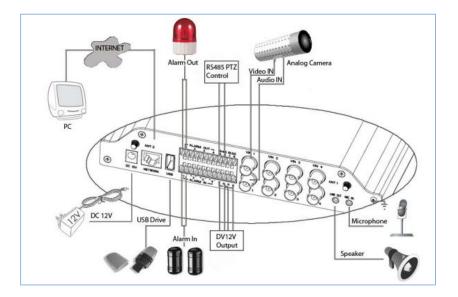
In this scenario, when Alarm IN triggers an alarm, a SIP call will be triggered to preconfigured SIP phone and Alarm Output circuit will be closed/opened (depending on initial state)





Scenario 5: Encoders

With two encoders: GXV3504 supporting up to 4 analog cameras, and GXV3500 supporting one analog camera



A video encoder digitizes analog video signals and sends digital images directly over an IP network, such as a LAN, intranet or Internet.

It essentially turns an analog video system into a network video system and enables users to view live images using a Web browser or a video management software such as Gsurf_Pro or else on any local or remote computer on a network.

It allows authorized viewers from different locations to simultaneously access images from the same analog camera, as well as network cameras if they are added to the system.





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





Delivering high quality, secure and reliable voice, video, data & mobility for <u>SMBs</u>



- 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 & UCM6510 series IP PBX Appliances



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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











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





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

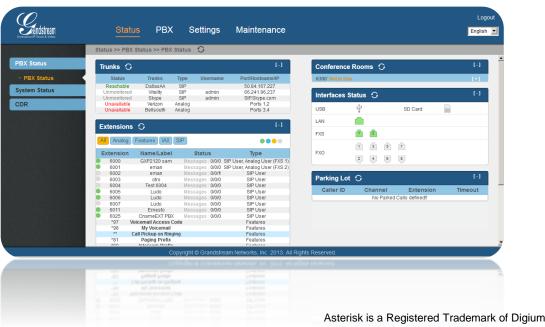
Video
Face to face video calls
Video Surveillance Integration
SIP Video
Video Codec Support



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







Easiest Possible Setup...

	X Status >> PBX S	status 🕑									L	
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Unavailable	Verizon A	Analog	Ports 1,2 Ports 3,4	USB	∯ SD Card							
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6005	Ludo	Messages : 0/0/0	SIP User	Caller	- Extensions		Extensions	CallerID Name	Technology	IP and Port	Status	Options
6007	Ludo	Messages : 0/0/0	SIP User		- Analog Trunks		6000	GXP2120 sam	SIP/FXS1	-	Offline	/ 也 💼
6025	CnameEXT PBX	Messages : 0/0/0	SIP User		- VolP Trunks		6001	eman	SIP/FXS2	-	Offline	/ 心 🏛
*98	My Voicemail		Features		- Outbound Routes		6002	eman	SIP	-	Offline	/ 心 🏛
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Video Calling

- Any SIP video endpoint
- See customers and employees through real-time, face-to-face video communication













Video Calling

 When used with Grandstream video phones, create your own multi-user video conference







Video Surveillance Integration

- 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
- Receive alerts from anywhere





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







Delivering high quality, secure and reliable mobility





Make and receive calls on a smartphone & tablet

- 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





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



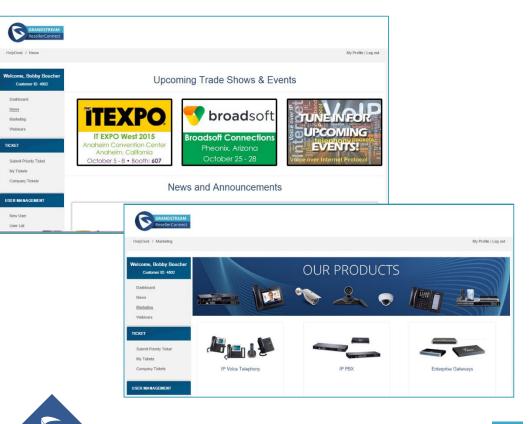


Nevrs

TICKET

ResellerConnect – Grandstream's Exclusive Reseller Portal

- Priority technical support tickets
- Exclusive product webinars
- Advance notification
- Access to high-resolution product images & graphics
- Weekly site updates & announcements
- Notification of feature & firmware updates
- Qualification as a Grandstream Certified Reseller





Thank you

Any Questions?





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