

VoIP



GRANDSTREAM



IP PBX Appliances
Multimedia IP Phones
Enterprise/SMB IP Phones
Video Conferencing



Analog Telephone Adapters
Analog VoIP Gateways
DECT IP Phones

IP Video Phones

GXV3275

Android Video IP Phone with 7" Touchscreen LCD



- Android OS, Version 4.2
- Up to 6 SIP accounts
- 7" Capacitive touchscreen LCD & advanced megapixel CMOS camera
- Built-in WiFi (802.11b/g/n), Bluetooth, mini-HDMI, SD & dual USB
- HD audio (available on handset and speakerphone)
- HD video (720p available via mini-HDMI output)
- Dual Gigabit network ports with integrated PoE
- Hundreds of thousands of Google Play™ apps available, including: Skype®, Microsoft Lync® & Google Hangouts™

GXV3240

Android Video IP Phone with 4.3" Touchscreen LCD



- Android OS, Version 4.2
- Up to 6 SIP accounts
- 4.3" Capacitive touchscreen LCD & advanced megapixel CMOS camera
- Built-in WiFi (802.11b/g/n), Bluetooth, mini-HDMI, SD & USB
- HD audio (available on handset and speakerphone)
- HD video (720p available via mini-HDMI output)
- Dual Gigabit network ports with integrated PoE
- Optional extension module (GXP2200EXT)
- Hundreds of thousands of Google Play™ apps available, including: Skype®, Microsoft Lync® & Google Hangouts™

GXP2200 EXT



- Extension module used with GXV3240 & GXP2140
- Features 20 extension keys and 2 arrow keys for page switching with bi-color LEDs alongside a large LCD with 320x160 resolution
- Device powered by the host phone
- Up to 4 modules can be cascaded to provide display of up to 160 extensions

Specifications	GXV3275	GXV3240
Camera	Megapixel	
Network Ports	2 x 10M/100M/1000M Network Ports, PoE	
WiFi and Bluetooth	Internal (802.11b/g/n) and Bluetooth	
Peripheral Ports	2 x USB, SD, mini-HDMI, Stereo Headset with Microphone, RJ9 headset jack	USB, SD, mini-HDMI, RJ9 headset jack
Voice, Video and Audio Codecs	Support for G.711μ/a, G.722 (wide-band), G.726-32, iLBC, Opus (pending), in-band and out-of-band DTMF (In audio, RFC2833, SIP INFO)	
Security and Provisioning	256-bit AES Encrypted Configuration File, TLS, SRTP, HTTPS, 802.1x Media Access Control	
Multi-language	English, German, Italian, French, Spanish, Portuguese, Russian, Chinese, Korean, Japanese, and more	
Universal Power Supply	12V, 1.5A, A/C Adapter included	

Enterprise IP Phones

GXP2160

Enterprise HD IP Phone with Multi-Keys



- Linux based
- 6 lines, 6 SIP accounts, 5 soft keys, 5-way conference
- 4.3" 480x272 TFT color LCD
- HD audio (handset, speakerphone & headset)
- Dual Gigabit network ports with integrated PoE
- Built-in Bluetooth, USB, EHS (Electronic Hook Switch)
- Multi-language
- 24 BLF/fast-dial keys

GXP2140

Enterprise HD IP Phone



- Linux based
- 4 lines, 4 SIP accounts, 5 soft keys, 5-way conference
- 4.3" 480x272 TFT color LCD
- HD audio (handset, speakerphone & headset)
- Dual Gigabit network ports with integrated PoE
- Built-in Bluetooth, USB, EHS (Electronic Hook Switch)
- Multi-language
- Optional extension module (GXP2200EXT)

GXP2130

Enterprise HD IP Phone with Multi-Keys



- Linux based
- 3 lines, 3 SIP accounts, 4 soft keys, 4-way conference
- 2.8" 320x240 TFT color LCD
- HD audio (handset, speakerphone & headset)
- Dual Gigabit network ports with integrated PoE
- Built-in Bluetooth
- Multi-language
- 8 BLF/fast-dial keys

Specifications	GXP2160	GXP2140	GXP2130
Line Keys and SIP Accounts	6 Line keys, 6 SIP accounts	4 line keys, 4 SIP accounts	3 line keys, 3 SIP accounts
Network & Headset Ports	2 x 10M/100M/1000M (Gigabit) Network Ports, PoE		
Peripheral Interfaces	Integrated Bluetooth, USB, RJ9, Headset Jack		RJ9, Integrated Bluetooth
Voice Codescs	G.711, G.722 (wide-band), G.723 (pending) G.729A/B, G.726, iLBC, Opus (pending), in-band and out-of-band		
Security and Provisioning	TLS/SRTP, 802.1x, TR-069, Encrypted XML config file, TFTP/HTTP/HTTPS		
Personalized Web Apps	Weather, Currency SDK/API for custom application development		
Expansion Module/BLF Keys	24 BLF/speed dial keys	Supports 4 GXP2200EXT modules	8 BLF/speed-dial keys
Multi-language	English, German, Italian, French, Spanish, Portuguese, Russian, Turkish, Polish, Chinese, Korean, Japanese, Czech, Arabic, Hebrew, Croatian, Hungarian, Dutch, Slovenian and more		
Universal Power Supply	12V, 1.0A A/C Adapter included		12V, 0.5A A/C Adapter included

Small Business IP Phones

GXP1630

Small Business Gigabit HD IP Phone



- Linux based
- 3 lines, 3 SIP Accounts, 3 soft keys and up to 4-way conference
- 8 BLF/speed dial keys
- 132x64 graphic backlit LCD display
- HD audio (available on handset and speakerphone)
- Dual Gigabit network ports with integrated PoE
- Electronic Hook Switch (EHS) with Plantronics headsets
- TLS/SRTP/HTTPS, TR-069, auto-provisioning and multi-language

GXP1628

Small Business Gigabit HD IP Phone



- Linux based
- 2 lines, 2 SIP Accounts, 3 soft keys and up to 3-way conference
- 8 BLF/speed dial keys
- 132x48 graphic backlit LCD display
- HD audio (available on handset and speakerphone)
- Dual Gigabit network ports with integrated PoE
- Electronic Hook Switch (EHS) with Plantronics headsets
- TLS/SRTP/HTTPS, TR-069, auto-provisioning and multi-language

GXP1620/1625

Small Business HD IP Phone



- Linux based
- 2 lines, 2 SIP accounts, 3 soft keys and up to 3-way conference
- 132x48 backlit graphic LCD display
- HD audio capability (available on handset and speakerphone)
- Dual network ports (10/100Mbps) with integrated PoE (GXP1625 only)
- Electronic Hook Switch (EHS) with Plantronics headsets
- TLS/SRTP/HTTPS, TR-069, auto-provisioning and multi-language

GXP1610

Small Business HD IP Phone



- Linux based
- 2 line keys, 1 SIP account, 3 soft keys, 3-way conference
- 132x48 graphic LCD display
- Dual network ports (10/100Mbps)
- Electronic Hook Switch (EHS) with Plantronics headsets
- TLS/SRTP/HTTPS, TR-069, auto-provisioning and multi-language

Specifications	GXP1630	GXP1628	GXP1620/1625	GXP1610
Line Keys & SIP Accounts	3 Line Keys 3 SIP Accounts	2 Line Keys 2 SIP Accounts	2 Line Keys 2 SIP Accounts	2 Line Keys 1 SIP Account
Network Ports	2 Gigabit ports		2 10/100Mbps ports	
PoE	Yes		GXP1625 (yes) GXP1620 (no)	No
Headset	Yes			
Telephony Features	Hold, transfer, forward, 3-way conferencing (GXP1630: 4-way conferencing), call park/pickup, shared-call appearance (SCA) / bridged-line appearance (BLA), downloadable phonebook (XML, LDAP, up to 500 items), call waiting, call history (up to 200 records), flexible dial plan, hot desking, personalized ringtones, server redundancy & fail-over			
Voice Codescs	G.711, G.722, G.723 (pending), G.726-32, G.729 A/B, in-band and out-of-band DTMF (In audio, RFC2833, SIP INFO)			
Security & Provisioning	TLS/HTTPS/TR069, 802.1X, SRTP, Encrypted XML, TFTP/HTTP, Browser			
Multi-language	English, German, Italian, French, Spanish, Portuguese, Russian, Turkish, Polish, Chinese, Korean, Japanese, Czech, Arabic, Hebrew, Croatian, Hungarian, Dutch, Slovenian and more			
Universal Power Supply	5V, 600mA, A/C adapter included			

Video Conferencing



GVC3200

FHD Video Conferencing System



- Patent-pending built-in MCU supports up to 9-way video conferencing
- Remote controllable PTZ lens with 12x optical zoom
- Based on Android 4.4 and offers full access to video conferencing apps in the Google Play Store (such as Skype, Google Hangouts, etc.)
- Supports third-party SIP based video conferencing platforms
- Cross-protocol local conference mixing between SIP and Android apps
- Gigabit port, integrated dual-band WiFi, integrated Bluetooth
- 3x HDMI outputs for 3 separate HD monitors
- Share a computer screen via HDMI or VGA at 1080p for collaborative viewing of a presentation, document mark up and more
- Bluetooth multi-touch remote control included
- Includes everything needed for installation and use in the box, including external speaker/microphone, wall mount, HDMI cables, network cable, remote control and more

Specifications	GVC3200
Camera	1/3" 2 Megapixel CMOS 1920x1080 @30fps
Lens	12x optical zoom, +/- 23 degrees tilt, +/- 90 degrees pan
Network Interfaces	Gigabit port, integrated dual-band WiFi (b/g/n)
Bluetooth	Integrated Bluetooth 4.0+ EDR, includes multi-touch Bluetooth remote
Video Inputs / Outputs	Output: 3x HDMI with resolution up to 1080p / Input: 1x HDMI and 1x VGA with resolution up to 1080p
Voice Codecs	G.711μ/a, G.722 (wide-band), G.726-32, iLBC (pending), Opus, G.722.1, G.722.1c (pending) in-band and out-of-band DTMF (In audio, RFC2833, SIP INFO)
Video Codecs	H.264 BP/MP/HP, video resolution up to 1080p, frame rate up to 30fps, bitrate up to 4Mbps
Embedded MCU	Up to 4-way 1080p conference, 5-way 720p, 9-way at VGA resolution
Security	User and administrator level passwords, MD5 and MD5-session based authentication, 256-bit AES encrypted configuration file, TLS
Multi-Language	English, German, Italian, French, Spanish, Portuguese, Russian, Turkish, Polish, Chinese, Korean, Japanese and more
Package Content	GVC3200 video conference system, external USB speaker/MIC, remote control, lens cover, universal power supply, network cable (1.5 meters), USB extension cable (5 meters), mounting kit, 4 HDMI cables (one 1.5 meter cable, two 3 meter cables and one 5 meter cable), 2 AAA batteries, quick installation guide, brochure, GPL license 128/256-bit SRTP, HTTPS, 802.1x media access control

Analog VoIP Gateways & Analog Telephone Adapters

HT502/503

2 FXS / 1 FXS & 1 FXO Analog Telephone Adapter

- 2 FXS ports (HT502), 1 FXS and 1 FXO port (HT503)
- Up to 2 SIP profiles
- Integrated router and QoS capability
- Advanced security protection with TLS/SRTP/HTTPS
- Auto-provisioning

HT701/HT702/HT704

1/2/4 FXS port Analog Telephone Adapter

- 1 FXS port (HT701), 2/4 FXS ports (HT702/704)
- 1 SIP profile (HT701), Up to 2 SIP profiles (HT702/704)
- Advanced security protection with TLS/SRTP/HTTPS
- Auto-provisioning

				
HT701		HT702/HT704		
Specifications	HT502	HT503	HT701	HT702/704
Telephone Interfaces	2 FXS	1 FXS, 1 FXO	1 FXS	2 FXS (HT702), 4 FXS (HT704)
PSTN Failover Life-line Port	N/A	Yes	N/A	N/A
Network (10/100Mbps) Ports	2 (with integrated router)		1	1
Short/Long Haul Ring Load	REN 3, Up to 3,000 ft (1km)		REN 5, Up to 3,00 ft (1km)	REN 3, Up to 3,00 ft (1km)
Voice and Fax Codecs	G.711, G.723.1, G.729A/B, G.729E (HT502 and HT503 only), G.726, iLBC, T.38			
Provisioning	Encrypted XML, TFTP/HTTP, TR-069, HTTPS			
Telephony Features	Hold, transfer, forward, 3-way conference, call waiting, DTMF via SIP INFO/RFC2833/In-audio, dial plan			
Universal Power Supply	12V, 500mA (1A for HT704); A/C adapter included			

GXW4004/4008 and GXW4104/4108

4/8 port FXS Gateway and 4/8 port FXO Gateway

- GXW4004/4008: 4/8 FXS ports with PSTN failover life-line
- GXW4104/4108: 4/8 FXO ports
- Integrated router with QoS capability
- Multiple SIP profiles
- Advanced security protection with SRTP

GXW4216/4224/4232/4248

15/24/32/48 port FXS Gateway

- 16/24/32/48 port models
- Multiple SIP profiles
- Advanced security protection with TLS/SRTP/HTTPS
- Gigabit network port
- 128x32 backlit graphic LCD and multi-language

Specifications		GXW4004/4008	GXW4104/4108	GXW4216/4224/4232/4248
Telephone Interfaces	FXS	4/8	N/A	16/24/32/48
	Telco 50-pins	N/A	N/A	1 (GXW4216/4224) 2 (GXW4232/4248)
	FXO	N/A	4/8	N/A
Network Ports	10/100/1000Mbps	2 (10/100Mbps)	2 (10/100Mbps)	1 (10/100/1000Mbps)
	Integrated Router	Yes	N/A	N/A
PSTN Failover Life-line Port		1	N/A	N/A
Short and Long Haul	REN	2	N/A	2
	With 24 AWG lines	Up to 3,000 ft (1km)	N/A	Up to 6,000 ft (2km)
Voice and Fax Codecs		G.711, G.723.1, G.729A/B, G.726, iLBC, T.38		
Provisioning		Encrypted XML, TFTP/HTTP, Browser		
Telephony Features		Hold, transfer, forward, 3-way conference, call waiting (GXW40xx and GXW42xx only) DTMF via SIP INFO/RFC2833/In-audio, dial plan, hunt group (GXW40xx and GXW42xx only)		
Universal Power Supply		12V, 1.5A	12V, 1 A	12V, 5 A (GXW4248: 24 VDC, 6.25A)

IP PBXs & DECT Cordless IP Phones

UCM6510

T1/E1/J1 IP PBX



- 2 PSTN trunk FXO ports, 2 analog telephone FXS ports with life-line capability
- Up to 2000 SIP endpoint and up to 50 SIP trunk accounts
- Up to 200 concurrent calls with up to 64 conference attendees
- Dual Gigabit network ports with integrated PoE; USB and SD ports; integrated NAT router with advanced QoS support
- 128x32 backlit graphic LCD
- Zero Configuration endpoint provisioning & no licensing fees

UCM6102/6104/6108/6116

IP PBX for Small to Medium Businesses



- 2/4/8/16 FXO trunk ports and 2 FXS ports with life-line capability
- Up to 500 SIP endpoints and up to 50 SIP trunk accounts
- Up to 30/45/60/60 concurrent calls, up to 25/32 conference attendees
- Gigabit network ports with PoE; USB and SD ports
- 128x32 backlit graphic LCD
- Zero Configuration endpoint provisioning and no licensing fees

Specifications		UCM6510	UCM6108/6116	UCM6102/6104
Telephone Interfaces		2 FXS, 2 FXO, 1 T1/E1/J1	2 FXS, 8/16 FXO	2 FXS, 2/4 FXO
Network Ports	10/100/1000Mbps	2	1	2
	Integrated Router	Yes	N/A	UCM6102 only
PSTN Failover Life-line Port		Yes		
PoE		Yes (802.3 at PoE+)		
Voice, Video and Fax Codecs		G.711, G.722, G.723, G.726, G.729A/B, iLBC, GSM, H.264, H.263, H.263+, T.38		
Provisioning		Zero Configuration provisioning for Grandstream endpoints		
Telephony Features		Call park, call forward, call transfer, DND, ring/hunt group, conference bridge, paging/intercom		
Caller ID		Bellcore/Telcordia, ETSI-FSK, ETSI-DTMF, SIN 227-BT, NTT Japan		
Polarity Reversal/Wink		Yes		
Universal Power Supply		12V, 1.5A; AC Adapter Included		



DP715

DP710

DP715/DP710

DECT Cordless IP Phone System

- Integrated base station and handset (DP715)
- Charging unit and handset (DP710)
- 1.7" 102x80 backlit graphic LCD
- Each base station supports up to 5 registered DECT handsets, and up to 4 concurrent calls
- Up to 10-hour talk time, up to 80-hour standby time and multi-language support

Specifications	DP715/DP710
Frequency Range	1880 ~ 1900 MHz (Europe), 1920 ~ 1920 MHz (USA), 1910 ~ 1920 (Brazil)
Phonebook & Call History (by handset)	200 records, 10 dialed call records, 30 retrieved call records
Network Interfaces	Single 10/100Mbps ethernet port
Voice Codecs	G.711, G.723.1, G.726-32, AAL2, G.729A/B, iLBC
SIP Accounts	Up to 5 distinct SIP accounts per system, one or multiple handsets per SIP account
Security and Provisioning	TLS/SRTP/HTTPS, TFTP/HTTP, TR-069
Universal Power Supply	6V, 500mA for Base Station (DP715); 7V, 420mA for Charger Unit (DP710)



GRANDSTREAM

CONNECTING THE WORLD



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Shenzhen, China 518057

Analog VoIP Gateways & Analog Telephone Adapters

HT502/503

2 FXS / 1 FXS & 1 FXO Analog Telephone Adapter

- 2 FXS ports (HT502), 1 FXS and 1 FXO port (HT503)
- Up to 2 SIP profiles
- Integrated router and QoS capability
- Advanced security protection with TLS/SRTP/HTTPS
- Auto-provisioning

HT701/HT702/HT704

1/2/4 FXS port Analog Telephone Adapter

- 1 FXS port (HT701), 2/4 FXS ports (HT702/704)
- 1 SIP profile (HT701), Up to 2 SIP profiles (HT702/704)
- Advanced security protection with TLS/SRTP/HTTPS
- Auto-provisioning

Specifications	HT502	HT503	HT701	HT702/704
Telephone Interfaces	2 FXS	1 FXS, 1 FXO	1 FXS	2 FXS (HT702), 4 FXS (HT704)
PSTN Failover Life-line Port	N/A	Yes	N/A	N/A
Network (10/100Mbps) Ports	2 (with integrated router)		1	1
Short/Long Haul Ring Load	REN 3, Up to 3,000 ft (1km)		REN 5, Up to 3,00 ft (1km)	REN 3, Up to 3,00 ft (1km)
Voice and Fax Codecs	G.711, G.723.1, G.729A/B, G.729E (HT502 and HT503 only), G.726, iLBC, T.38			
Provisioning	Encrypted XML, TFTP/HTTP, TR-069, HTTPS			
Telephony Features	Hold, transfer, forward, 3-way conference, call waiting, DTMF via SIP INFO/RFC2833/In-audio, dial plan			
Universal Power Supply	12V, 500mA (1A for HT704); A/C adapter included			

GXW4004/4008 and GXW4104/4108

4/8 port FXS Gateway and 4/8 port FXO Gateway

- GXW4004/4008: 4/8 FXS ports with PSTN failover life-line
- GXW4104/4108: 4/8 FXO ports
- Integrated router with QoS capability
- Multiple SIP profiles
- Advanced security protection with SRTP

GXW4216/4224/4232/4248

15/24/32/48 port FXS Gateway

- 16/24/32/48 port models
- Multiple SIP profiles
- Advanced security protection with TLS/SRTP/HTTPS
- Gigabit network port
- 128x32 backlit graphic LCD and multi-language

Specifications		GXW4004/4008	GXW4104/4108	GXW4216/4224/4232/4248
Telephone Interfaces	FXS	4/8	N/A	16/24/32/48
	Telco 50-pins	N/A	N/A	1 (GXW4216/4224) 2 (GXW4232/4248)
Network Ports	FXO	N/A	4/8	N/A
	10/100/1000Mbps	2 (10/100Mbps)	2 (10/100Mbps)	1 (10/100/1000Mbps)
PSTN Failover Life-line Port	Integrated Router	Yes	N/A	N/A
Short and Long Haul	REN	2	N/A	2
	With 24 AWG lines	Up to 3,000 ft (1km)	N/A	Up to 6,000 ft (2km)
Voice and Fax Codecs		G.711, G.723.1, G.729A/B, G.726, iLBC, T.38		
Provisioning		Encrypted XML, TFTP/HTTP, Browser		
Telephony Features		Hold, transfer, forward, 3-way conference, call waiting (GXW40xx and GXW42xx only) DTMF via SIP INFO/RFC2833/In-audio, dial plan, hunt group (GXW40xx and GXW42xx only)		
Universal Power Supply		12V, 1.5A	12V, 1 A	12V, 5 A (GXW4248: 24 VDC, 6.25A)

IP PBXs & DECT Cordless IP Phones

UCM6510

T1/E1/J1 IP PBX

- 2 PSTN trunk FXO ports, 2 analog telephone FXS ports with life-line capability
- Up to 2000 SIP endpoint and up to 50 SIP trunk accounts
- Up to 200 concurrent calls with up to 64 conference attendees
- Dual Gigabit network ports with integrated PoE; USB and SD ports; integrated NAT router with advanced QoS support
- 128x32 backlit graphic LCD
- Zero Configuration endpoint provisioning & no licensing fees

UCM6102/6104/6108/6116

IP PBX for Small to Medium Businesses

- 2/4/8/16 FXO trunk ports and 2 FXS ports with life-line capability
- Up to 500 SIP endpoints and up to 50 SIP trunk accounts
- Up to 30/45/60/60 concurrent calls, up to 25/32 conference attendees
- Gigabit network ports with PoE; USB and SD ports
- 128x32 backlit graphic LCD
- Zero Configuration endpoint provisioning and no licensing fees

Specifications		UCM6510	UCM6108/6116	UCM6102/6104
Telephone Interfaces		2 FXS, 2 FXO, 1 T1/E1/J1	2 FXS, 8/16 FXO	2 FXS, 2/4 FXO
Network Ports	10/100/1000Mbps	2	1	2
	Integrated Router	Yes	N/A	UCM6102 only
PSTN Failover Life-line Port		Yes		
PoE		Yes (802.3 at PoE+)		
Voice, Video and Fax Codecs		G.711, G.722, G.723, G.726, G.729A/B, iLBC, GSM, H.264, H.263, H.263+, T.38		
Provisioning		Zero Configuration provisioning for Grandstream endpoints		
Telephony Features		Call park, call forward, call transfer, DND, ring/hunt group, conference bridge, paging/intercom		
Caller ID		Bellcore/Telcordia, ETSI-FSK, ETSI-DTMF, SIN 227-BT, NTT Japan		
Polarity Reversal/Wink		Yes		
Universal Power Supply		12V, 1.5A; AC Adapter Included		

DP715/DP710

DECT Cordless IP Phone System

- Integrated base station and handset (DP715)
- Charging unit and handset (DP710)
- 1.7" 102x80 backlit graphic LCD
- Each base station supports up to 5 registered DECT handsets, and up to 4 concurrent calls
- Up to 10-hour talk time, up to 80-hour standby time and multi-language support

Specifications	DP715/DP710
Frequency Range	1880 ~ 1900 MHz (Europe), 1920 ~ 1920 MHz (USA), 1910 ~ 1920 (Brazil)
Phonebook & Call History (by handset)	200 records, 10 dialed call records, 30 retrieved call records
Network Interfaces	Single 10/100Mbps ethernet port
Voice Codecs	G.711, G.723.1, G.726-32, AAL2, G.729A/B, iLBC
SIP Accounts	Up to 5 distinct SIP accounts per system, one or multiple handsets per SIP account
Security and Provisioning	TLS/SRTP/HTTPS, TFTP/HTTP, TR-069
Universal Power Supply	6V, 500mA for Base Station (DP715); 7V, 420mA for Charger Unit (DP710)



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Video Conferencing



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Analog VoIP Gateways
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Video Conferencing



GXV3275

Android Video IP Phone with 7" Touchscreen LCD

- Android OS, Version 4.2
- Up to 6 SIP accounts
- 7" Capacitive touchscreen LCD & advanced megapixel CMOS camera
- Built-in WiFi (802.11b/g/n), Bluetooth, mini-HDMI, SD & dual USB
- HD audio (available on handset and speakerphone)
- HD video (720p available via mini-HDMI output)
- Dual Gigabit network ports with integrated PoE
- Hundreds of thousands of Google Play™ apps available, including: Skype®, Microsoft Lync® & Google Hangouts™



GXV3240

Android Video IP Phone with 4.3" Touchscreen LCD

- Android OS, Version 4.2
- Up to 6 SIP accounts
- 4.3" Capacitive touchscreen LCD & advanced megapixel CMOS camera
- Built-in WiFi (802.11b/g/n), Bluetooth, mini-HDMI, SD & USB
- HD audio (available on handset and speakerphone)
- HD video (720p available via mini-HDMI output)
- Dual Gigabit network ports with integrated PoE
- Optional extension module (GXP2200EXT)
- Hundreds of thousands of Google Play™ apps available, including: Skype®, Microsoft Lync® & Google Hangouts™



GXP2200 EXT

- Extension module used with GXV3240 & GXP2140
- Features 20 extension keys and 2 arrow keys for page switching with bi-color LEDs alongside a large LCD with 320x160 resolution
- Device powered by the host phone
- Up to 4 modules can be cascaded to provide display of up to 160 extensions



GXP2160

Enterprise HD IP Phone with Multi-Keys

- Linux based
- 6 lines, 6 SIP accounts, 5 soft keys, 5-way conference
- 4.3" 480x272 TFT color LCD
- HD audio (handset, speakerphone & headset)
- Dual Gigabit network ports with integrated PoE
- Built-in Bluetooth, USB, EHS (Electronic Hook Switch)
- Multi-language
- 24 BLF/fast-dial keys



GXP2140

Enterprise HD IP Phone

- Linux based
- 4 lines, 4 SIP accounts, 5 soft keys, 5-way conference
- 4.3" 480x272 TFT color LCD
- HD audio (handset, speakerphone & headset)
- Dual Gigabit network ports with integrated PoE
- Built-in Bluetooth, USB, EHS (Electronic Hook Switch)
- Multi-language
- Optional extension module (GXP2200EXT)



GXP2130

Enterprise HD IP Phone with Multi-Keys

- Linux based
- 3 lines, 3 SIP accounts, 4 soft keys, 4-way conference
- 2.8" 320x240 TFT color LCD
- HD audio (handset, speakerphone & headset)
- Dual Gigabit network ports with integrated PoE
- Built-in Bluetooth
- Multi-language
- 8 BLF/fast-dial keys



GXP1630

Small Business Gigabit HD IP Phone

- Linux based
- 3 lines, 3 SIP Accounts, 3 soft keys and up to 4-way conference
- 8 BLF/speed dial keys
- 132x64 graphic backlit LCD display
- HD audio (available on handset and speakerphone)
- Dual Gigabit network ports with integrated PoE
- Electronic Hook Switch (EHS) with Plantronics headsets
- TLS/SRTP/HTTPS, TR-069, auto-provisioning and multi-language



GXP1628

Small Business Gigabit HD IP Phone

- Linux based
- 2 lines, 2 SIP Accounts, 3 soft keys and up to 3-way conference
- 8 BLF/speed dial keys
- 132x48 graphic backlit LCD display
- HD audio (available on handset and speakerphone)
- Dual Gigabit network ports with integrated PoE
- Electronic Hook Switch (EHS) with Plantronics headsets
- TLS/SRTP/HTTPS, TR-069, auto-provisioning and multi-language



GXP1620/1625

Small Business HD IP Phone

- Linux based
- 2 lines, 2 SIP accounts, 3 soft keys and up to 3-way conference
- 132x48 backlit graphic LCD display
- HD audio capability (available on handset and speakerphone)
- Dual network ports (10/100Mbps) with integrated PoE (GXP1625 only)
- Electronic Hook Switch (EHS) with Plantronics headsets
- TLS/SRTP/HTTPS, TR-069, auto-provisioning and multi-language



GXP1610

Small Business HD IP Phone

- Linux based
- 2 line keys, 1 SIP account, 3 soft keys, 3-way conference
- 132x48 graphic LCD display
- Dual network ports (10/100Mbps)
- Electronic Hook Switch (EHS) with Plantronics headsets
- TLS/SRTP/HTTPS, TR-069, auto-provisioning and multi-language



GVC3200

FHD Video Conferencing System

- Patent-pending built-in MCU supports up to 9-way video conferencing
- Remote controllable PTZ lens with 12x optical zoom
- Based on Android 4.4 and offers full access to video conferencing apps in the Google Play Store (such as Skype, Google Hangouts, etc.)
- Supports third-party SIP based video conferencing platforms
- Cross-protocol local conference mixing between SIP and Android apps
- Gigabit port, integrated dual-band WiFi, integrated Bluetooth
- 3x HDMI outputs for 3 separate HD monitors
- Share a computer screen via HDMI or VGA at 1080p for collaborative viewing of a presentation, document mark up and more
- Bluetooth multi-touch remote control included
- Includes everything needed for installation and use in the box, including external speaker/microphone, wall mount, HDMI cables, network cable, remote control and more



Specifications	GXV3275	GXV3240
Camera	Megapixel	
Network Ports	2 x 10M/100M/1000M Network Ports, PoE	
WiFi and Bluetooth	Internal (802.11b/g/n) and Bluetooth	
Peripheral Ports	2 x USB, SD, mini-HDMI, Stereo Headset with Microphone, RJ9 headset jack	USB, SD, mini-HDMI, RJ9 headset jack
Voice, Video and Audio Codecs	Support for G.711µ/a, G.722 (wide-band), G.726-32, iLBC, Opus (pending), in-band and out-of-band DTMF (In audio, RFC2833, SIP INFO)	
Security and Provisioning	256-bit AES Encrypted Configuration File, TLS, SRTP, HTTPS, 802.1x Media Access Control	
Multi-language	English, German, Italian, French, Spanish, Portuguese, Russian, Chinese, Korean, Japanese, and more	
Universal Power Supply	12V, 1.5A, A/C Adapter included	

Specifications	GXP2160	GXP2140	GXP2130
Line Keys and SIP Accounts	6 Line keys, 6 SIP accounts	4 line keys, 4 SIP accounts	3 line keys, 3 SIP accounts
Network & Headset Ports	2 x 10M/100M/1000M (Gigabit) Network Ports, PoE		
Peripheral Interfaces	Integrated Bluetooth, USB, RJ9, Headset Jack		RJ9, Integrated Bluetooth
Voice Codecs	G.711, G.722 (wide-band), G.723 (pending) G.729A/B, G.726, iLBC, Opus (pending), in-band and out-of-band		
Security and Provisioning	TLS/SRTP, 802.1x, TR-069, Encrypted XML config file, TFTP/HTTP/HTTPS		
Personalized Web Apps	Weather, Currency SDK/API for custom application development		
Expansion Module/BLF Keys	24 BLF/speed dial keys	Supports 4 GXP2200EXT modules	8 BLF/speed-dial keys
Multi-language	English, German, Italian, French, Spanish, Portuguese, Russian, Turkish, Polish, Chinese, Korean, Japanese, Czech, Arabic, Hebrew, Croatian, Hungarian, Dutch, Slovenian and more		
Universal Power Supply	12V, 1.0A A/C Adapter included		12V, 0.5A A/C Adapter included

Specifications	GXP1630	GXP1628	GXP1620/1625	GXP1610
Line Keys & SIP Accounts	3 Line Keys 3 SIP Accounts	2 Line Keys 2 SIP Accounts	2 Line Keys 2 SIP Accounts	2 Line Keys 1 SIP Account
Network Ports	2 Gigabit ports		2 10/100Mbps ports	
PoE	Yes		GXP1625 (yes) GXP1620 (no)	No
Headset	Yes			
Telephony Features	Hold, transfer, forward, 3-way conferencing (GXP1630: 4-way conferencing), call park/pickup, shared-call appearance (SCA) / bridged-line appearance (BLA), downloadable phonebook (XML, LDAP, up to 500 items), call waiting, call history (up to 200 records), flexible dial plan, hot desking, personalized ringtones, server redundancy & fail-over			
Voice Codecs	G.711, G.722, G.723 (pending), G.726-32, G.729 A/B, in-band and out-of-band DTMF (In audio, RFC2833, SIP INFO)			
Security & Provisioning	TLS/HTTPS/TR069, 802.1X, SRTP, Encrypted XML, TFTP/HTTP, Browser			
Multi-language	English, German, Italian, French, Spanish, Portuguese, Russian, Turkish, Polish, Chinese, Korean, Japanese, Czech, Arabic, Hebrew, Croatian, Hungarian, Dutch, Slovenian and more			
Universal Power Supply	5V, 600mA, A/C adapter included			

Specifications	GVC3200
Camera	1/3" 2 Megapixel CMOS 1920x1080 @30fps
Lens	12x optical zoom, +/- 23 degrees tilt, +/- 90 degrees pan
Network Interfaces	Gigabit port, integrated dual-band WiFi (b/g/n)
Bluetooth	Integrated Bluetooth 4.0+ EDR, includes multi-touch Bluetooth remote
Video Inputs / Outputs	Output: 3x HDMI with resolution up to 1080p / Input: 1x HDMI and 1x VGA with resolution up to 1080p
Voice Codecs	G.711µ/a, G.722 (wide-band), G.726-32, iLBC (pending), Opus, G.722.1, G.722.1c (pending) in-band and out-of-band DTMF (In audio, RFC2833, SIP INFO)
Video Codecs	H.264 BP/MP/HP, video resolution up to 1080p, frame rate up to 30fps, bitrate up to 4Mbps
Embedded MCU	Up to 4-way 1080p conference, 5-way 720p, 9-way at VGA resolution
Security	User and administrator level passwords, MD5 and MD5-sess based authentication, 256-bit AES encrypted configuration file, TLS
Multi-Language	English, German, Italian, French, Spanish, Portuguese, Russian, Turkish, Polish, Chinese, Korean, Japanese and more
Package Content	GVC3200 video conference system, external USB speaker/MIC, remote control, lens cover, universal power supply, network cable (1.5 meters), USB extension cable (5 meters), mounting kit, 4 HDMI cables (one 1.5 meter cable, two 3 meter cables and one 5 meter cable), 2 AAA batteries, quick installation guide, brochure, GPL license, 128/256-bit SRTP, HTTPS, 802.1x media access control