

## VoIP



## Innovative IP Voice & Video

# Inteligy Tampabay

813-769-4694



IP PBX Appliances  
Multimedia IP Phones  
Enterprise/SMB IP Phones

- Analog Telephone Adapters
- Analog VoIP Gateways
- DECT IP Phones

# Multimedia IP 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™

## GXP2200EXT

- 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 & 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	Yes (English, German, French, Spanish, Chinese, Japanese, Korean, Portuguese, Italian 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 & 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 Codec	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 or BLF keys	24 BLF/fast-dial keys	Supports GXP2200EXT modules	8 BLF/fast-dial keys
Multi-language	English, Arabic, Chinese, Croatian, Czech, Dutch, German, French, Hebrew, Hungarian, Italian, Japanese, Korean, Polish, Portuguese, Russian, Slovenia, Spanish, Turkish		
Universal Power Supply	12V,1.0A, A/C Adapter included		12V, 0.5A A/C Adapter included

# Small Business / Home Office IP Phones

## GXP1450

### Small Business HD IP Phone

- Linux based
- 2 lines, 3 XML programmable soft keys and 3-way conference
- 180x60 graphic LCD with backlight and 4-level gray scale
- HD audio (available on handset and speakerphone)
- Dual network ports with integrated PoE and full duplex speakerphone
- Power adapter included, XML screen and personalized Web apps
- 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 conferencing
- 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
- Power adapter included, personalized Web apps and customizable screen content and formatting using XML
- 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 conferencing
- 132x48 graphic LCD display
- Dual network ports (10/100Mbps)
- Electronic Hook Switch (EHS) with Plantronics headsets
- Power adapter and personalized Web apps
- TLS/SRTP/HTTPS, TR-069, auto-provisioning and multi-language

Specifications	GXP1450	GXP1620/1625	GXP1610
Line Keys & SIP Accounts	2 Line Keys 2 SIP Accounts	2 Line Keys 2 SIP Accounts	2 Line Keys 1 SIP Account
Network Ports	2		
PoE	Yes	GXP1625 (yes) GXP1620 (no)	No
Headset	Yes		
Telephony Features	Hold, transfer, forward, 3-way conferencing, call park/pickup, shared-call appearance (SCA) / bridged-line appearance (BLA) (GXP1450, GXP162x), 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, iLBC (pending), Opus (pending - except on GXP1450), 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, Simplified Chinese, Traditional Chinese, Arabic, Czech, German, Spanish, French, Hebrew, Croatian Hungarian, Italian, Japanese, Korean, Dutch, Polish, Portuguese, Russian, Slovenian and Turkish		
Universal Power Supply	+5V, 600mA or higher (GXP16xx), +5V, 800mA or higher (GXP1450) A/C Adapter included		

# IP PBX Appliances

## UCM6510

### T1/E1/J1 IP PBX Appliance



- 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 32 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 and no licensing fees

## UCM6102/6104

### IP PBX Appliance For Small Businesses



- 2/4 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 concurrent calls with up to 25 conference attendees
- Dual Gigabit network ports with PoE; USB and SD ports
- 128x32 backlit graphic LCD
- Zero Configuration endpoint provisioning and no licensing fees

## UCM6108/6116

### IP PBX Appliance For Medium Businesses



- 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 60 concurrent calls with up to 32 conference attendees
- Gigabit network port 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	10M/100M/1000M	2	1	2
Ports	Integrated Router	Yes	N/A	Yes (UCM6102 only)
PSTN Failover Life-line Port		Yes		
PoE		Yes (802.3 at PoE+)		
Voice, Video & 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 Endpoint Provisioning for Grandstream Phones		
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		

# Analog VoIP Gateways

## GXW4004/4008

### 4/8 port FXS Gateway



- 4/8 ports with PSTN failover life-line
- Integrated router with QoS capability
- Multiple SIP profiles
- Advanced security protection with TLS/SRTP/HTTPS
- Auto-provisioning

## GXW4104/4108

### 4/8 port FXO Gateway



- 4/8 ports
- Integrated router with QoS capability
- Multiple SIP profiles
- Advanced security protection with SRTP
- Auto-provisioning

## GXW4216/4224/4232/4248

### 16/24/32/48 port FXS Gateway



- 16/24/32/48 port models
- Multiple SIP profiles
- Advanced security protection with TLS/SRTP/HTTPS
- Auto-provisioning
- 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/GXW4224) 2 (GXW4232/GXW4248)
	FXO	N/A	4 / 8	N/A
Network	10M/100M/1000M	2 (10M/100M)	2 (10M/100M)	1 (10M/100M/1000M)
Ports	Integrated Router	Yes	N/A	N/A
PSTN Failover Life-line Port		1	N/A	N/A
Short &	REN	2	N/A	2
Long Haul	With 24 AWG Lines	Up to 3,000 ft (1km)	N/A	Up to 6,000 ft (2km)
Voice & 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)		
Caller ID		Bellcore Type 1 & 2, ETSI, BT, NTT and DTMF Based CID		
Polarity Reversal/Wink		Yes		
Universal Power Supply		12V, 1.5A	12V, 1A	12V, 5A (GXW4248: 24VDC, 6.25A)

# Analog Telephone Adapters & DECT IP Phone System



## HT502/HT503

### 2 FXS port 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

## 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 (10M/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 3,000 ft (1km)	REN 3, Up to 3,000 ft (1km)
Voice & 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			
Caller ID	Bellcore Type 1 & 2, ETSI, BT, NTT and DTMF Based CID			
Disconnect Methods	Polarity Reversal/Wink, Busy Tone, Loop Current			
Universal Power Supply	12V, 500mA (1A for HT704); A/C Adapter included			



DP715



DP710

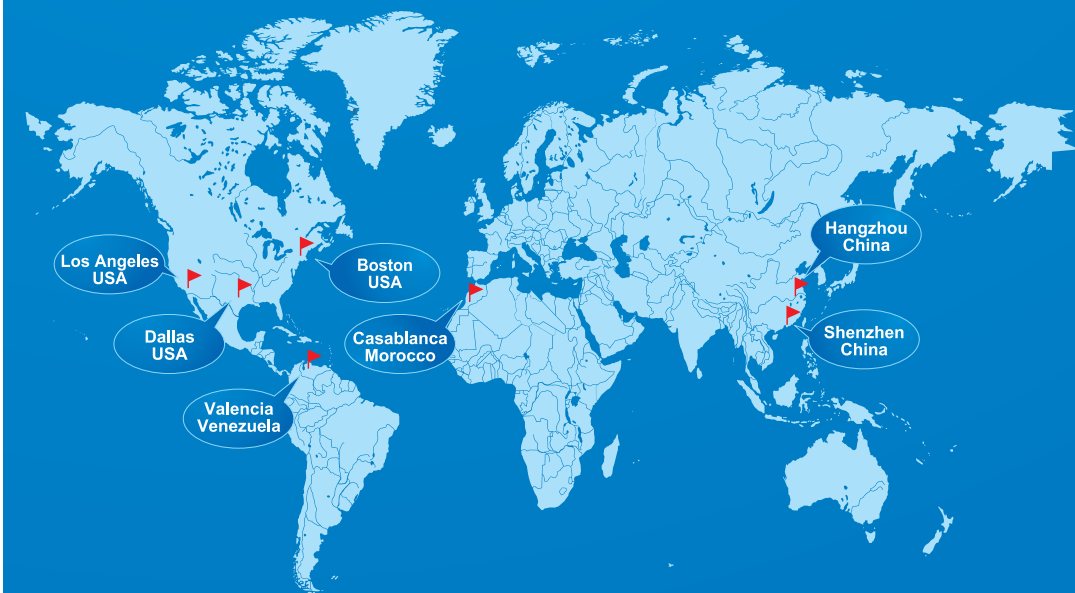
## DP715/DP710

### 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 ~ 1930 MHz (U.S.), 1910 ~ 1920 MHz (Brazil)
Phonebook & Call History(per handset)	200 Records, 10 Dialed Call Records, 30 Received Call Records
Network Interface	Single 10/100Mbps Ethernet Port
Voice Codecs	G.711, G.723.1, G.726-32 AAL2, G.729A/B, iLBC
SIP Accounts	Up to Five Distinct SIP Accounts per System; One or Multiple Handsets per SIP Account
Security & Provisioning	TLS/SRTP/HTTPS, TFTP/HTTP, TR-069 (pending)
Universal Power Supply	6V, 500mA for Base Station (DP715); 7V, 420mA for Charger Unit (DP710)





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## Analog VoIP Gateways



### GXW4004/4008

#### 4/8 port FXS Gateway

- 4/8 ports with PSTN failover life-line
- Integrated router with QoS capability
- Multiple SIP profiles
- Advanced security protection with TLS/SRTP/HTTPS
- Auto-provisioning



### GXW4104/4108

#### 4/8 port FXO Gateway

- 4/8 ports
- Integrated router with QoS capability
- Multiple SIP profiles
- Advanced security protection with SRTP
- Auto-provisioning



### GXW4216/4224/4232/4248

#### 16/24/32/48 port FXS Gateway

- 16/24/32/48 port models
- Multiple SIP profiles
- Advanced security protection with TLS/SRTP/HTTPS
- Auto-provisioning
- 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/GXW4224) 2 (GXW4232/GXW4248)
	FXO	N/A	4 / 8	N/A
Network	10M/100M/1000M	2 (10M/100M)	2 (10M/100M)	1 (10M/100M/1000M)
Ports	Integrated Router	Yes	N/A	N/A
PSTN Failover Life-line Port		1	N/A	N/A
Short & 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 & 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)		
Caller ID		Bellcore Type 1 & 2, ETSI, BT, NTT and DTMF Based CID		
Polarity Reversal/Wink		Yes		
Universal Power Supply		12V, 1.5A	12V, 1A	12V, 5A (GXW4248: 24VDC, 6.25A)

## Analog Telephone Adapters & DECT IP Phone System



### HT502/HT503

#### 2 FXS port 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 (10M/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 3,000 ft (1km)	REN 3, Up to 3,000 ft (1km)
Voice & 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			
Caller ID	Bellcore Type 1 & 2, ETSI, BT, NTT and DTMF Based CID			
Disconnect Methods	Polarity Reversal/Wink, Busy Tone, Loop Current			
Universal Power Supply	12V, 500mA (1A for HT704); A/C Adapter included			



DP715

DP710

### DP715/DP710

#### DECT Cordless IP Phone System

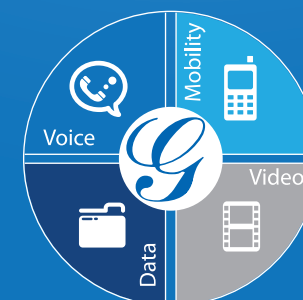
- Integrated base station and handset (DP715)
- Charging unit and handset (DP710)
- 1.7" 102x80 backlit graphic LCD
- Each base station supports up to 5 registered DECT handsets, and up to 4 concurrent calls
- Up to 10-hour talk time, up to 80-hour standby time and multi-language support

Specifications	DP715/DP710
Frequency Range	1880 ~ 1900 MHz (Europe), 1920 ~ 1930 MHz (U.S.), 1910 ~ 1920 MHz (Brazil)
Phonebook & Call History(per handset)	200 Records, 10 Dialed Call Records, 30 Received Call Records
Network Interface	Single 10/100Mbps Ethernet Port
Voice Codecs	G.711, G.723.1, G.726-32 AAL2, G.729A/B, iLBC
SIP Accounts	Up to Five Distinct SIP Accounts per System; One or Multiple Handsets per SIP Account
Security & Provisioning	TLS/SRTP/HTTPS, TFTP/HTTP, TR-069 (pending)
Universal Power Supply	6V, 500mA for Base Station (DP715); 7V, 420mA for Charger Unit (DP710)

## VoIP



IP PBX Appliances  
Multimedia IP Phones  
Enterprise/SMB IP Phones



Analog Telephone Adapters  
Analog VoIP Gateways  
DECT IP Phones



Multimedia IP Phones

Enterprise IP Phones

Small Business / Home Office IP Phones

IP PBX Appliances

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™

GXP2200EXT

- 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

GXP1450

Small Business HD IP Phone

- Linux based
- 2 lines, 3 XML programmable soft keys and 3-way conference
- 180x60 graphic LCD with backlight and 4-level gray scale
- HD audio (available on handset and speakerphone)
- Dual network ports with integrated PoE and full duplex speakerphone
- Power adapter included, XML screen and personalized Web apps
- 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 conferencing
- 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
- Power adapter included, personalized Web apps and customizable screen content and formatting using XML
- 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 conferencing
- 132x48 graphic LCD display
- Dual network ports (10/100Mbps)
- Electronic Hook Switch (EHS) with Plantronics headsets
- Power adapter and personalized Web apps
- TLS/SRTP/HTTPS, TR-069, auto-provisioning and multi-language

UCM6510

T1/E1/J1 IP PBX Appliance

- 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 32 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 and no licensing fees

UCM6102/6104

IP PBX Appliance For Small Businesses

- 2/4 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 concurrent calls with up to 25 conference attendees
- Dual Gigabit network ports with PoE; USB and SD ports
- 128x32 backlit graphic LCD
- Zero Configuration endpoint provisioning and no licensing fees

UCM6108/6116

IP PBX Appliance For Medium Businesses

- 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 60 concurrent calls with up to 32 conference attendees
- Gigabit network port with PoE; USB and SD ports
- 128x32 backlit graphic LCD
- Zero Configuration endpoint provisioning and no licensing fees

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 & 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	Yes (English, German, French, Spanish, Chinese, Japanese, Korean, Portuguese, Italian and more)	
Universal Power Supply	12V, 1.5A, A/C Adapter included	

Specifications	GXP2160	GXP2140	GXP2130
Line Keys & 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 Codec	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 or BLF keys	24 BLF/fast-dial keys	Supports GXP2200EXT modules	8 BLF/fast-dial keys
Multi-language	English, Arabic, Chinese, Croatian, Czech, Dutch, German, French, Hebrew, Hungarian, Italian, Japanese, Korean, Polish, Portuguese, Russian, Slovenia, Spanish, Turkish		
Universal Power Supply	12V, 1.0A, A/C Adapter included		12V, 0.5A A/C Adapter included

Specifications	GXP1450	GXP1620/1625	GXP1610
Line Keys & SIP Accounts	2 Line Keys 2 SIP Accounts	2 Line Keys 2 SIP Accounts	2 Line Keys 1 SIP Account
Network Ports	2		
PoE	Yes	GXP1625 (yes) GXP1620 (no)	No
Headset	Yes		
Telephony Features	Hold, transfer, forward, 3-way conferencing, call park/pickup, shared-call appearance (SCA) / bridged-line appearance (BLA) (GXP1450, GXP162x), downloadable phonebook (XML, LDAP, up to 500 items), call waiting, call histroy (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, iLBC (pending), Opus (pending - except on GXP1450), 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, Simplified Chinese, Traditional Chinese, Arabic, Czech, German, Spanish, French, Hebrew, Croatian Hungarian, Italian, Japanese, Korean, Dutch, Polish, Portuguese, Russian, Slovenian and Turkish		
Universal Power Supply	+5V, 600mA or higher (GXP16xx), +5V, 800mA or higher (GXP1450) A/C Adapter included		

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	10M/100M/1000M	2	1	2
Ports	Integrated Router	Yes	N/A	Yes (UCM6102 only)
PSTN Failover Life-line Port		Yes		
PoE		Yes (802.3 at PoE+)		
Voice, Video & 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 Endpoint Provisioning for Grandstream Phones		
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		