

DP750 & DP720 DECT IP PHONE

Product Overview



- Founded in 2002
- Over 500 employees
- Product Portfolio contains over 40 products: Video Conferencing Systems, 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





Awards & Recognition

Global Presence



15 time winner





9 time winner

2 time winner









United States - Venezuela - Morocco - China - Malaysia

Over 10 Years of Growth and Innovation...





Product Interoperability

IP-PBX and Softswitch

































































ITSPs

































































Product Interoperability

Door Intercom









VoIP Applications, Services, and Hardware









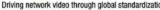






Video Management Systems





































VoIP Product Lines

IP Multimedia Phones

Enterprise IP Phones

Small Business / Home Office IP Phones

Analog Gateways & ATAs

IP PBX Appliances

Video & Audio Conference Systems

Surveillance Product Lines

Dome IP Cameras

IP Video Encoder

Outdoor IP Cameras

Video Management Software

Network Video Recorder



























DP750 & DP720

DECT Cordless IP Phone

- Up to 10 SIP accounts per system, up to 5 DP720 handsets per base station
- Full HD audio quality
- 300m range outdoors, 50m indoors
- Up to 20-hour talk time, 250-hour standby time
- DP750 Base, DP720 Handset













DP750

Long-range DECT VolP Base Station

- Up to 10 SIP accounts per system, up to 5 DP720 handsets per base station
- Supports 3-way conferencing, full HD audio
- 300m range outdoors, 50m indoors
- Fully complaint with the SIP/DECT standard
- Use with Grandstream's UCM for Zero Config Provisioning
- Integrated PoE









DP720

DECT Cordless HD Handset

- Up to 20-hour talk time, 250-hour standby time
- 1.8 inch TFT LCD color screen
- Integrated 3.5mm headset jack, speakerphone
- Multi-line access for up to 10 lines per handset
- Up to 5 DP720 handsets per DP750 base station















Multiple SIP Accounts

Up to 10 distinct SIP accounts per system

- Each handset may map to any SIP account(s)
- Each SIP account may map to any handset(s)

















Enhanced Telephony Features

Easy access to common telephony features, directly from the phone menu

- Hold
- Transfer/Forward
- 3-way conference
- Call pick-up

Other features include:

- Downloadable phone book
- Call waiting
- Auto Answer
- Click-to-dial
- Flexible dial plan
- Music on hold





Quick Access Telephony Features

Call Forward



- 1. Select the phone **Menu**
- 2. Access Call Settings
- Select Call Forward



3-Way Conference

- 1. Press the Dial Key
- 2. Press the left softkey for **Options**
- 3. Select Conference
- Enter the second party and press the Dial Key

Call Transfer



- 1. Select the Options softkey
- Select Transfer for blind transfer, or New Call for attended transfer
- 3. Press the **Dial Key**



Voicemail

- 1. Press the Menu key
- 2. Select Voice Mail
- 3. Select Play Message
- 4. Choose the account to retrieve the voice message





Air Interface

Frequency Bands:

- 1800 1900 MHz (Europe)
- 1920 1930 MHz (US)
- 1910 1920 MHz (Brazil)
- 1786 1792 MHz (Korea)
- 1893 1906 MHz (Japan)
- 1880 1895 MHz (Taiwan)

Number of Channels:

- 10 (Europe)
- 5 (US, Brazil, Japan)
- 3 (Korea)
- 8 (Taiwan)









Color Screen

New vibrant color screen

- 1.8 inch (128x160) color TFT LCD
- Access a comprehensive phone menu and common telephony features directly through the screen (hold, transfer, forward etc.)











Voice Codecs

- G.722
- G.726
- G.711
- G.723.1
- G.729A/B
- iLBC
- OPUS
- ACE, AGC
- Ambient noise reduction





Accessibility







- Integrated 3.5mm headset jack
- Speakerphone for hands free mode
- Belt clip for easy mobility
- Micro-USB port for alternative charging and nonbattery operation
- Hearing Aid Compatibility (HAC) compliant





Ring Groups

- Circular Mode: all phones ring sequentially from the phone next to the phone that answered last
- Linear Mode: all phones ring sequentially in a predesignated order
- Parallel Mode: all phones ring concurrently and after one phone answers, the remaining available phones can make new calls
- Shared Mode: all phone ring concurrently and always share the same line similar to analog phones







Create a Total IP Solution With the UCM Series IP PBX







- Auto-Discovery, ZERO Configuration
- 1-touch call recording
- Call Detail Records (CDR)
- Music-on-hold
- Voicemail
- Easily connect to LDAP Phonebooks
- Conference Rooms
- Remote-Extension BLF





Multiple Device Registration

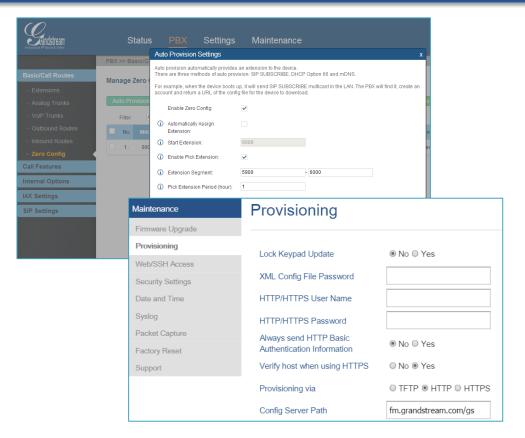
- The UCM supports multiple registrations per extension.
 Users can register multiple devices to the UCM using the same extension number.
- Access any call, anywhere, on any SIP device (mobile, home, office etc.)









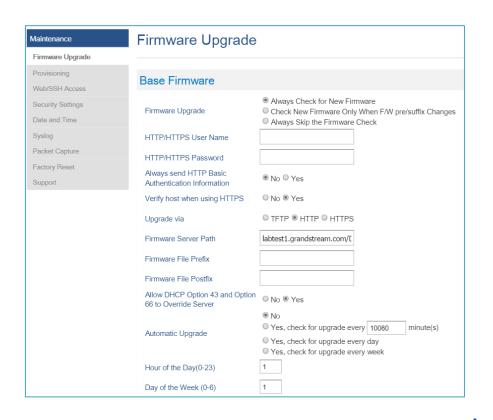


Configuration & Provisioning

- Auto-Discovery, ZERO Configuration supported through the UCM6100 and UCM6510 IP PBX
- Configuration via DP720 handset or via Web GUI
- Handset provisioning over-the-air
- Mass provisioning using TR-069 or AES encrypted XML configuration file







Firmware Upgrades

- Firmware upgrade via TFTP/HTTP/HTTPS
- Software upgrades over-the-air (SUOTA)
- Push the firmware upgrade through the UCM series IP PBX
- UCM supports Model Update feature to support new DP750/720 features via firmware upgrades without needing an additional UCM firmware upgrade



DECT IP Phone Comparison

Hunting Group support

Voice Codecs

LDAP Support

Upgrade/Provisioning

LED indicators (Base)

PoE support (Base)

Headset support

Keypad buttons

HD Audio

Battery Life

MWI LED (handset)

Hearing Aid Compatibility (HAC)

Charging via Micro-USB

Phonebook

Call logs

Security

Languages

Range

LCD

Grandstream DP750/720



Yealink **W56P**



Handsets **SIP Accounts Concurrent Calls**

Up to 5 Up to 10

Up to 4

Up to 5 Up to 5 Up to 4 Yes

Yes (Circular, Linear, Parallel, Shared line) G.711µ/a-law, VAD, CNG, PLC, AJB, G.723.1, G.729A/B, G.726-32,iLBC, G.722, OPUS, G.722.2/AMR-WB (special order) 3000 entries

300 records

Yes

SIP/TLS, SRTP, HTTPS, AES-256, 802.1x

HTTP, HTTPS, TFTP, TR-069, XML

Yes - Arabic, Chinese (Simplified, Traditional), Croatian, Czech, Danish, Dutch, English, Estonian, Finnish, French, German, Greek,

Hebrew, Hungarian, Italian, Japanese, Korean, Norwegian, Polish, Portuguese, Romanian, Russian, Slovenian, Spanish, Swedish, Turkish

5 LED indicators: Power, Network, Register, Call, DECT

300 meters outdoor and 50 meters indoor

Yes (IEEE 802.3af class 1)

1.8 inch (128x160) color TFT LCD

Yes (3.5mm headset jack)

3-color MWI LED

23

Yes

Yes (both on handset and speakerphone)

25-hour talk time, 200-hour standby time

Yes

G.711µ/A, G.722 G.723, G.726, G.729, iLBC . VAD, CNG, AEC, PLC, AJB 500 entries Yes, amount not specified

> Yes (3.5mm headset jack) 1-color MWI LED 31 Yes Yes

Yes

SIP/TLS, SRTP, HTTPS, AES-256, 802.1x

HTTP, HTTPS, TFTP

Yes (languages not specified)

3 LED indicators: Power, Network, Call

300 meters outdoor and 50 meters indoor

Yes (IEEE 802.3af class 1)

2.4 inch (240x320) color TFT LCD

30-hour talk time, 400-hour standby time

No



DECT IP Phone Comparison

Grandstream DP750/720



Panasonic KX-TGP600/KX-TPA60



Up to 5

Up to 8 Up to 8

Handsets **SIP Accounts Concurrent Calls**

LED indicators (Base)

PoE support (Base)

Headset support

Keypad buttons

HD Audio

Battery Life

MWI LED (handset)

Hearing Aid Compatibility (HAC)

Charging via Micro-USB

Range

LCD

Up to 10

Up to 8 Yes

G.711µ/a-law, G.729, G.722, G.722.2 (AMR-WB)

Turkish, Romanian, Kazakh, Montenegrin

1 LED indicator: Status

240 meters outdoor and 90 meters indoor

Yes (IEEE 802.3af)

1.8 inch color TFT LCD

Yes (2.5mm headset jack)

1-color MWI LED

26

Yes

Yes

11-hour talk time, 200-hour standby time

Nο

Hunting Group support Voice Codecs Phonebook

Up to 4

Yes (Circular, Linear, Parallel, Shared line) G.711µ/a-law, VAD, CNG, PLC, AJB, G.723.1, G.729A/B, G.726-32,iLBC, G.722, OPUS, G.722.2/AMR-WB (special order) 500 entries 3000 entries Call logs 300 records 40 records Yes No SIP/TLS, SRTP, HTTPS, AES-256, 802.1x

5 LED indicators: Power, Network, Register, Call, DECT

300 meters outdoor and 50 meters indoor

Yes (IEEE 802.3af class 1)

1.8 inch (128x160) color TFT LCD

Yes (3.5mm headset jack)

3-color MWI LED

23

Yes

Yes (both on handset and speakerphone)

25-hour talk time, 200-hour standby time

Yes

HTTP, HTTPS, TFTP, TR-069, XML Arabic, Chinese (Simplified, Traditional), Croatian, Czech, Danish, Dutch, English, Estonian, Finnish, French, German, Greek,

SIP/TLS, SIPS, SRTP, HTTPS HTTP, HTTPS, TFTP English, Spanish, French, German, Italian, Dutch, Greek, Hungarian, Portuguese, Polish, Slovakian, Czech, Croatian, Russian, Ukranian,

LDAP Support Security Upgrade/Provisioning Languages Hebrew, Hungarian, Italian, Japanese, Korean, Norwegian, Polish, Portuguese, Romanian, Russian, Slovenian, Spanish, Swedish, Turkish



Thank you

Any Questions?





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