

Grandstream DP710 Additional Handset



Product Name: Grandstream DP710 Additional Handset

Manufacturer: Grandstream

Model Number: DP710

Grandstream DP710 DECT Additional Handset

The Grandstream DP710 is an Additional Handset for the DP715 DECT Base Station. The DP710 offers users an affordable, powerful, high quality and simple to configure VoIP DECT phone, and is designed specifically for both residential users and small business.

Grandstream DP710 Key Features

- Secure and automated provisioning using HTTP/HTTPS/Telnet/TFTP, multiple SIP accounts, SIP over TCP/TLS, SRTP
- Advanced telephony features including Caller ID, Call Waiting, 3-Way Conference, Transfer, Forward, Do Not Disturb, Message Waiting Indication, auto answer, multi-language voice prompt, flexible dial plan
- When multiple handsets share the same SIP account, Hunting Group option may flexibly support:- Linear Mode, all phones ring sequentially in the predestinated order - Parallel Mode, all phones ring concurrently and after one phone answers the rest phones may place new calls - Shared Line Mode, all phones ring concurrently and always share the same line similar to an analog phone
- Support comprehensive voice codecs including G.711, G.723.1, G.729A/B, G.726 and iLBC
- Multi-Languages - English, German, French, Spanish, Dutch, Italian, Czech, Danish, Greek, Norwegian, Polish, Portuguese, Russian, Swedish, Turkish
- Currently Pending - TR069, IPV6
- The DP710 is NOT compatible with Skype

Grandstream DP710 Additional Handset Technical Specification

Air Interfaces

- Telephony standards: DECT / GAP Frequency range: 1880 - 1900 MHz (Europe), 1920 - 1930 MHz (US) Number of channels: 120 (Europe), 60 duplex (US) Modulation: GFSK Speech coding: 32 kbit/s Emission power: 10 mW (average power per channel) Range: up to 300 m outdoors, maximum of 50 m in buildings

Handset Display

- 1.7" 102x80 FSTN LCD with color backlight

Audio Interface

- Handsfree speaker

Telephony Features

- Caller ID display or block, call waiting, Flash, blind or attended transfer, forward, hold, do not disturb, 3-way conference

QoS

- Layer 2 (802.1Q VLAN/802.1p), Layer 3 (ToS, DiffServ, MPLS)

IP Transport

- RTP/RTCP

DTMF Method

- In-audio, RFC2833 and/or SIP Info

IP Signaling

- SIP (RFC 3261)

Multiple SIP accounts per base station

- Up to five (5) distinct SIP accounts per system; Independent SIP account per handset; Multiple handsets per SIP account

Hunting Group

- Linear mode; Parallel mode; Shared Line mode

Provisioning

- HTTP, HTTPS, TELNET, TFTP, TR-069 (pending), secure and automated provisioning

Security

- Security protection: SIP over TLS and SRTP.

Device Management

- Web interface or secure (AES encrypted) central configuration file for mass deployment

Phonebook(Per Handset)

- 200 numbers (up to 24 digits) with an associated name (up to 16 characters); 10 outgoing call entries; 30 incoming calls entries

Multi-language Input

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- English; Latin; Greek; Russian

Polyphonic Ringtones

- 18 different ringer melodies are available to indicate an incoming call (internal intercom or external VoIP)

Price: £28.95

VoIPVN <http://voip.com.vn>