### Version 1.4

# SIEMENS Gigaset C470IP/C475IP M1 Datasheet

### VoIP and Messaging with a pleasing touch

Interface: Operation: Standards: a/b (analogue), IP/Ethernet PSTN / PBX DECT, GAP, F-SMS, SIP, RTP, DHCP, STUN

#### Colour: Black and silver Configuration

- Connection of 1 external PSTN line + 2 external IP lines simultaneously
  - (Multiline: up to 6 SIP accounts)
- Up to 6 handsets can be registered

## **Main Features**

### General

- o Slim, modern design with Stand alone base station
- o Plug & Play
- Date and time functions are protected against power failure (date/time: with NTP synchronised)
- o Hybrid calls: PSTN and VoIP
- Comfortable menu with colored icons and text for easy use in 20 languages (partly available)

### ECO DECT

- Handset reduces the transmit power automatically depending on the distance to the base
- Reduction of the radio signal of the base settable by user
- Up to 60% less electricity consumption in comparison with conventional power supplies by switched mode power supply

### Handset-Display

 Illuminated 6 line graphical color handset display (128x128 pixel, 4096 colors)

### Standby

Display of digital clock or color picture as screensaver (4 selectable and settable by user)

### In Idle

- Date and time, state of charging and field intensity, handset name displayed
- o Text for softkeys shown

### **During Connection**

Display of call duration

### Handsfree Speaking / Headset / Earpiece

- Enhanced handsfree speaking with superior sound quality, separate speaker activated by illuminated Handsfree-key, volume adjustable (5 volume steps)
- Headset connection (mono) via 2,5mm jack connector (headset not included)
- o Adjustable handset earpiece volume (3 levels)

### Melodies

- o 5 polyphonic sounds and 10 melodies
- 5 loudness levels + off + crescendo
- Selectable internal call melody



### Dialing

- Bigger dialing font
- Default setting of outgoing line (VoIP/PSTN), over writable with long press of off-hook key
- Dialing plan: up to 30 dialing rules (call number, beginning of call number, connection type, PSTN, number blocking)
- Suffix dialing: For cost reductions an account Suffix can be attached to the dial string.
- Dial preparation with correction possibility
- o Redialing list for up to 20 different numbers
- Speed-Dialing using keys 2 9, 0 (shortcut)
- o Automatic prefix dialing of network provider

### **Receiving Calls**

- CLIP and CNIP displayed <sup>1) 2)</sup>
- o Display and handsfree key lighted
- o Ringer animation in display

### Missed calls

- Illuminated MWI-key signaling new messages with comfortable access to message list
- Call log list or Missed call list saving calls with CLIP / CNIP (approx. 30 entries, base required, factory setting) 1)
- Missed call notification via SMS to external number <sup>1) 2</sup>)

### Directory

- Directory with up to 150 numbers and names;
- VIP-entries (a specific melody can be assigned to each directory entry)

### SMS & Email <sup>1) 2)</sup>

- Automatic SMS registration
- Sending/receiving of SMS with up to 612 digits
- Storage of up to 30 SMS with 90 digits
- Up to 4 personalized SMS-Inboxes<sup>2</sup>
- Comfortable access to SMS-Services (info-channels)<sup>2)</sup>
- Transmission of e-mails via SMS <sup>1) 2)</sup>
- Copy of telephone numbers from SMS to directory
- V-Card support: name and number can be taken



# SIEMENS

### **Additional features**

- Alarm clock functionality with snooze function
- Keypad lock; protection from unintentional usage 0
- Baby-Alarm / room monitoring 0
- Editable internal handset name 0
- Support of Netprovider features via menu 1) 2) 0

### Features with more than 1 handset (Intercom)

- 2 VoIP and 1 PSTN Call simultaneously 0
- (+ 1 internal call or AM access simultaneously) 2 VoIP calls simultaneously 0
- (+ 1 internal call or AM access simultaneously)
- Collective internal call 0
- Transfer of external calls to another handset with prior 0 consultation call or recall
- VoIP resp. PSTN 3 party conference 0 (1 external/ 2 internal participants)
- Two 3-party-conferences simultaneously (1 external VoIP 0 or PSTN/2 internal participants each + AM access)
- Directory transfer between handsets 0 (base required)

### **IP Net Features**

- Multiline with up to 6 SIP accounts:
  - Configure up to 6 SIP accounts from different providers - Assign accounts individually to handsets (send & receive direction specifically)
- Ethernet (IEEE 802.3) 0
- VoIP support (embedded SIP VoIP, G711, G722 in base 0 station, G726, G729AB)
- Easy selection of provider profiles from HS 0
- Configuration of system via WEB Server 0
- Configuration of provider profiles via WEB Server 0
- Firmware Update directly from handset, without file 0 handling
- Quality of service\*: ToS, Diffserv 0
- (dependent on country, network and provider) DHCP client 0
- Operation behind router providing LAN connection 0
- VoIP supplementary services (device controlled): 0
- conference, hold/toggle, call forwarding Public Phone Book Search 0
- (availability is country specific / on request) Gigaset.net VoIP service: 0
  - free of charge calls between Gigaset VoIP phones
  - zero configuration
  - Gigaset.net phonebook
- Autoprovisioning via MAC Address for autoconfiguration including SIP username and password (availability on request)

### Email notification/subject/header

- Direct access to POP3 accounts without PC 0
- Email content reading on PC 0
- New Emails signaled by illuminated MWI-key with 0 comfortable access to message list: Information about sender, time & date, subject

### Instant messaging

- Messenger client, based on XMPP-protocol 0 - links common Messaging services (MSN, Yahoo, ICQ, AOL...) by Jabber Server with corresponding transport service
- Buddy list with presence status: - up to 20 buddies - visual indication of presence states:
  - Online, Other, Offline
- Set own presence status
- Send and receive instant messages
- 0 Illuminated MWI-key signaling new instant 0 messages
- Call buddies 0

### Features of the integrated answering machine

- Digital recording of voice messages with up to ca. 30 minutes recording time
- Via PIN-protected remote playback (4 digits) from 0 extern
- User guidance via handset Prompts via 0 announcements
- Predefined announcement (plug & play) 0
  - 1 Individual announcement and 1 advisory announcement (170 Sec.) recordable by user
- Phrase prompts in country specific language (factory settings per country version)
- Time-controlled change between announcement 0 and advisory message
- Display of new messages at the handset 0
- Live recording of a telephone call 0
- Call screening at handset 0
- Date and Time are announced 0
- Recording time adjustable: unlimited, 60, 120, 180 0 seconds
- Call acceptance adjustable: 0 immediately, automatic, after 10, 18, 30 Sec.
- Recordings have unlimited protection against power 0 failure
- 1) Country / customer specific
- 2) Provider dependant



## Version 1.4

# SIEMENS Technical data

### Handset

0	Standby-time:	up to 300 hours
0	Talk time:	up to 12 hours
0	Battery:	2 x NiMH (AAA)
0	Indoor range:	up to 50 meters
0	Outdoor range:	up to 300 meters
0	Dimensions:	Length 28 mm
		Width 48 mm
		Height 146 mm

### **Base station**

- Power supply: 230 V plug-in power supply
- Paging key on front of base
- Connection of up to 6 handsets
- o Dimensions: 157x107x46mm

### Package contents

- o 1 Handset
- o 1 Handset charger
- o 1 Belt clip
- 1 Standalone base station(IP+PSTN)
- o 2 AAA batteries
- o 2 Power supply units (base and handset charger)
- o 1 Telephone cord
- o 1 Ethernet cable
- o 1 User manual

### Market entry

o Market entry first retail version October 2007



# 

Name	Org.	Tel.	E-Mail
Gisèle D'Hoore	SHC PM P IP	+49-89-722-38740	gisele.dhoore@siemens.com

### Copyright © SHC GmbH & Co KG 2006

Copying of this document and giving it to others and the use or communication of the contents thereof, are forbidden without express authority. Offenders are liable to the payment of damages. All rights reserved in the event of the grant of a patent or the registration of a utility model or design.

### History of Changes

Version	Date	Author(s)	Remarks	
1.0	26.06.2006	Gisèle D'Hoore	Document created	
1.1	27.06.2006	Gisèle D'Hoore	Feedback A. Kruk > Update	
1.2.	02.11.2006	Gisèle D'Hoore	Update Multibase: No Multicell!!!! Update Multiline: up to 6 SIP accounts Update redialing list: 20 entries instead of 10 Update OPCAP: same features like Oslo-US Update Standby: 150h instead of 170h Update Talktime: 10h instead of 13h Email/Instant messaging: Update according to S450IP datasheet Gigaset.net VoIP service added	
1.3.	24.01.2007	Giséle D'Hoore	Check against datasheet C470/C475 (Datasheet_C470_C475_V1.5_151206.doc) Added ECO DECT Connection of 1 external PSTN line + 2 external IP lines simultaneously Features with more than 1 handset: update 3 calls simultaneously Recording time adjustable: unlimited,60 ,120, 180 seconds instead of unlimited, 30 ,60 , 120 seconds	
1.4.	14.03.2007	Giséle D'Hoore	<ul> <li>Updated ECO DECT: reduction of the radio signal of the base settable by user</li> <li>Added: codec G722 (base station)</li> <li>4 Screensaver</li> <li>AM call acceptancer after 10, 18, 30</li> <li>talk time up to 12 hours (R-Team Venus ANLG 28.02.07)</li> <li>standby-time: up to 300 hours (R-Team Venus ANLG 28.02.07)</li> <li>added: Bigger dialing font (22pt)</li> </ul>	

