







PHONTECH TELEPHONES AND INTERCOM UNITS



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Amendment Record

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SERVICE AND WARRANTY

Warranty time: 2 years from factory.

All goods sold by Jotron AS are warranted to be free from defect in workmanship and material for the period of twenty-four (24) months from the date of delivery (unless stated otherwise and confirmed in writing) PROVIDED:

- (a) Jotron AS is given full particulars in writing of any claim prior to the expiration of such a period and within fourteen days of the discovery of the alleged defect.
- (b) The product/s have been stored, maintained or used according to manual/s and specification/s
- (c) Liability shall be limited at Jotron AS' options to replacement or repair or to a sum not exceeding the net invoice value of the defective goods.
- (d) Upon request the alleged faulty goods are returned to Jotron AS at the Buyer's expense.
- (e) Unless expressly stipulated in the acceptance of the order Jotron AS gives no warranty or guarantee of the fitness or suitability of the goods for any purpose whether disclosed or otherwise.
- (f) All other warranties or conditions expressed or implied are hereby excluded and Jotron AS shall in no circumstances be liable for consequential damages.

UTILIZATION:

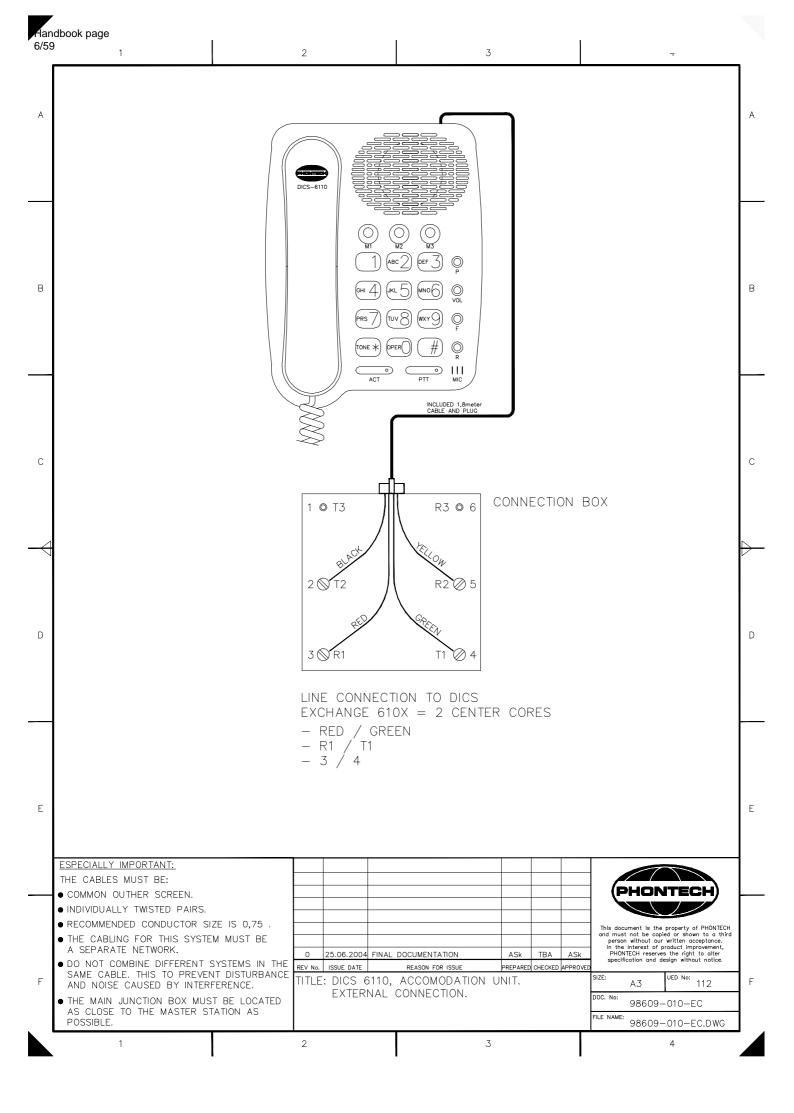
This equipment is not to be disposed in normal waste, but be handled in accordance with applicable waste disposal regulations in the country where the equipment is used

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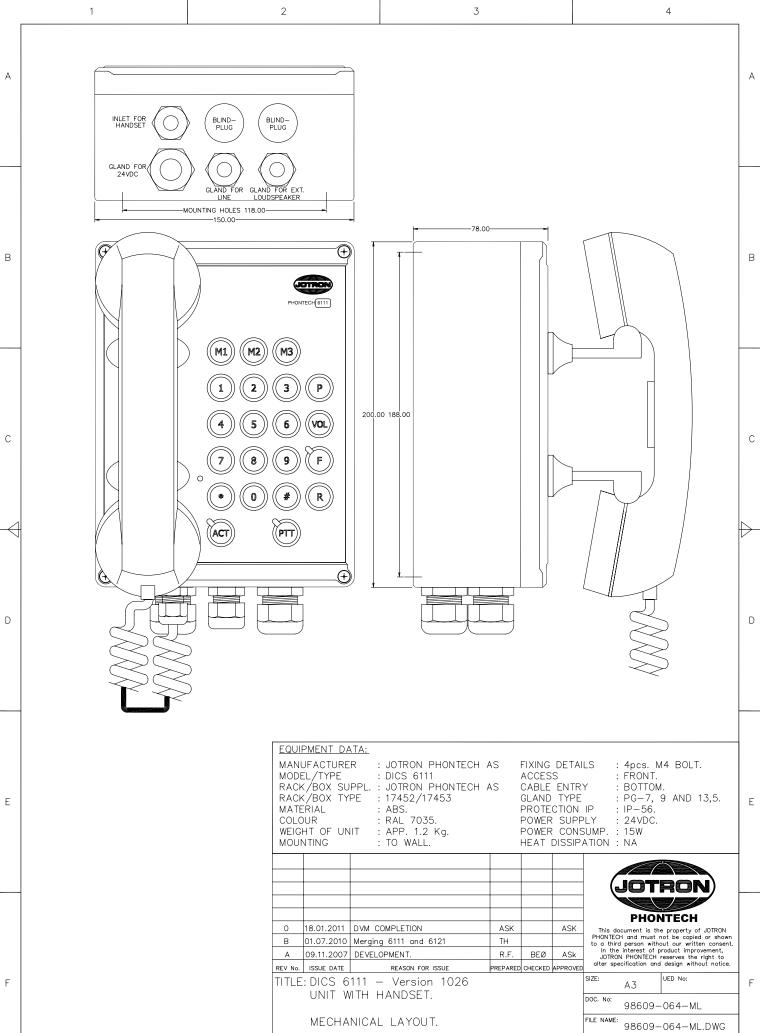
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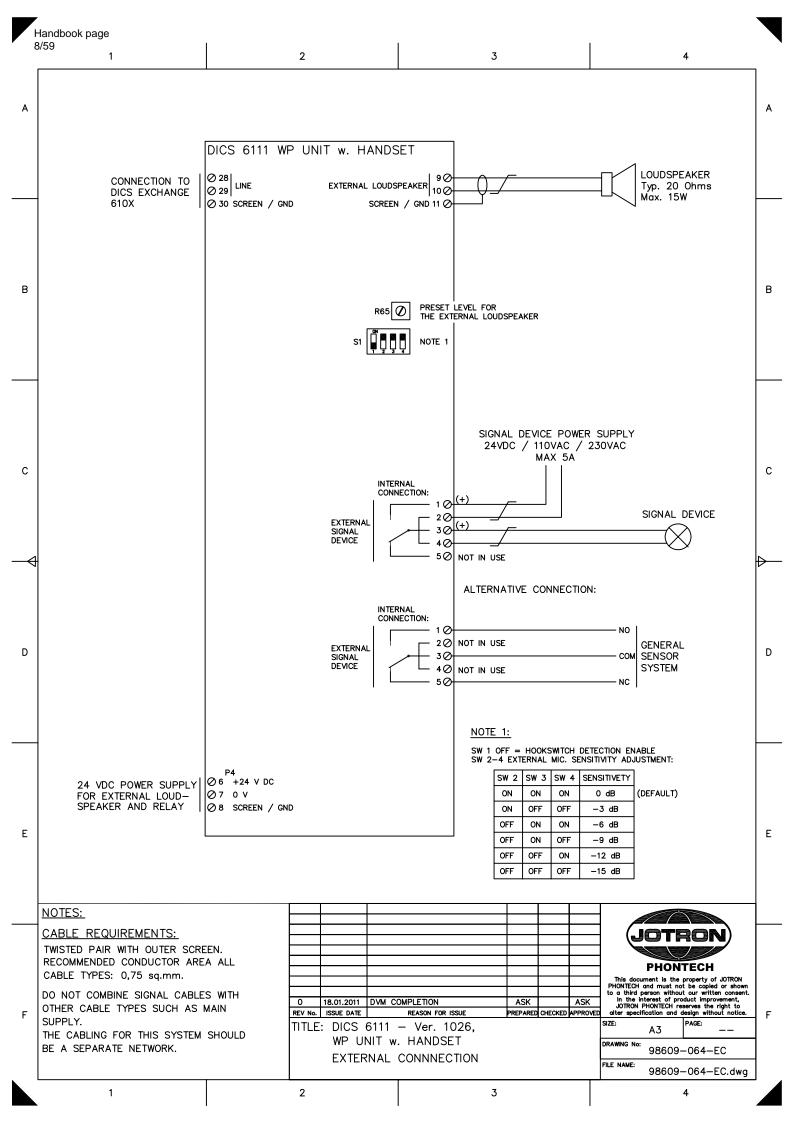
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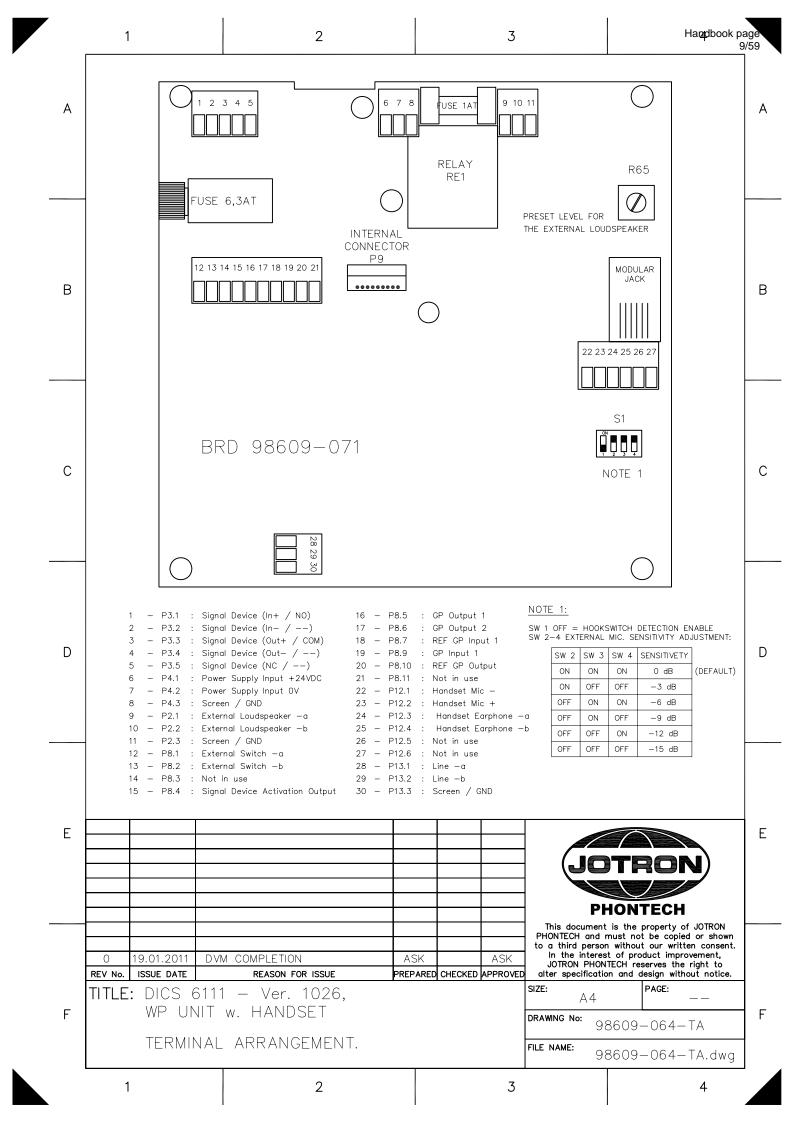
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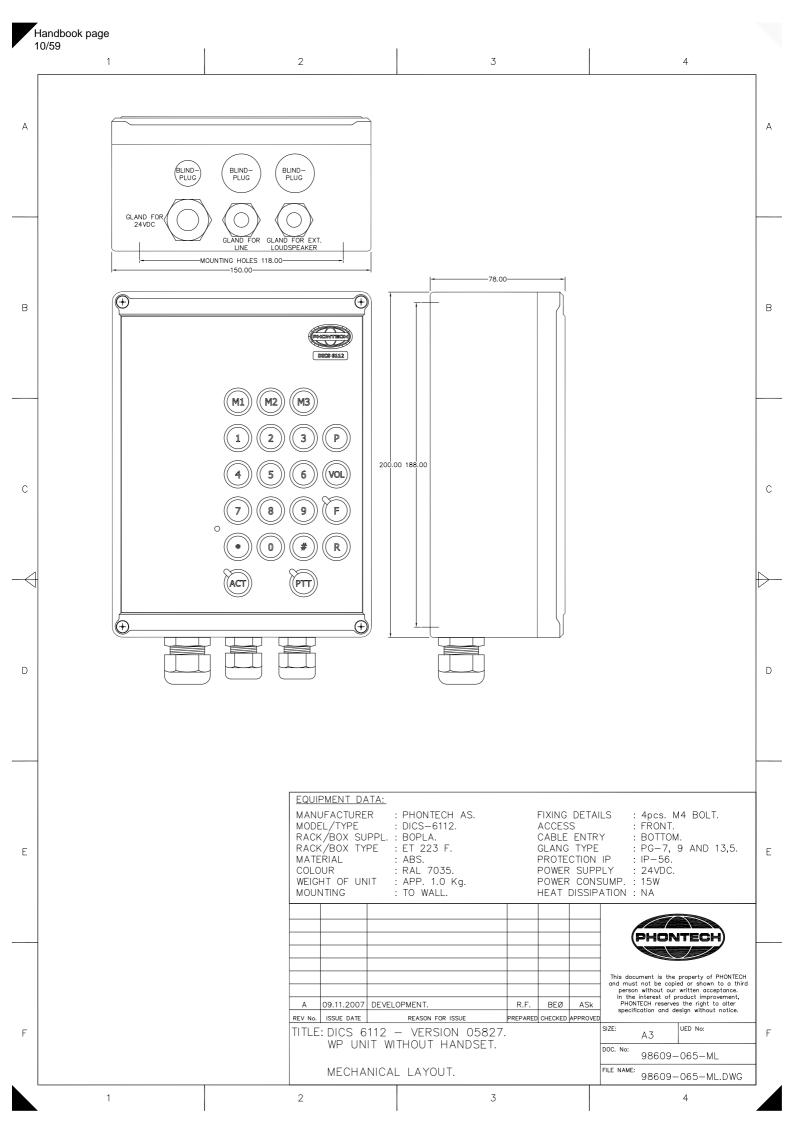


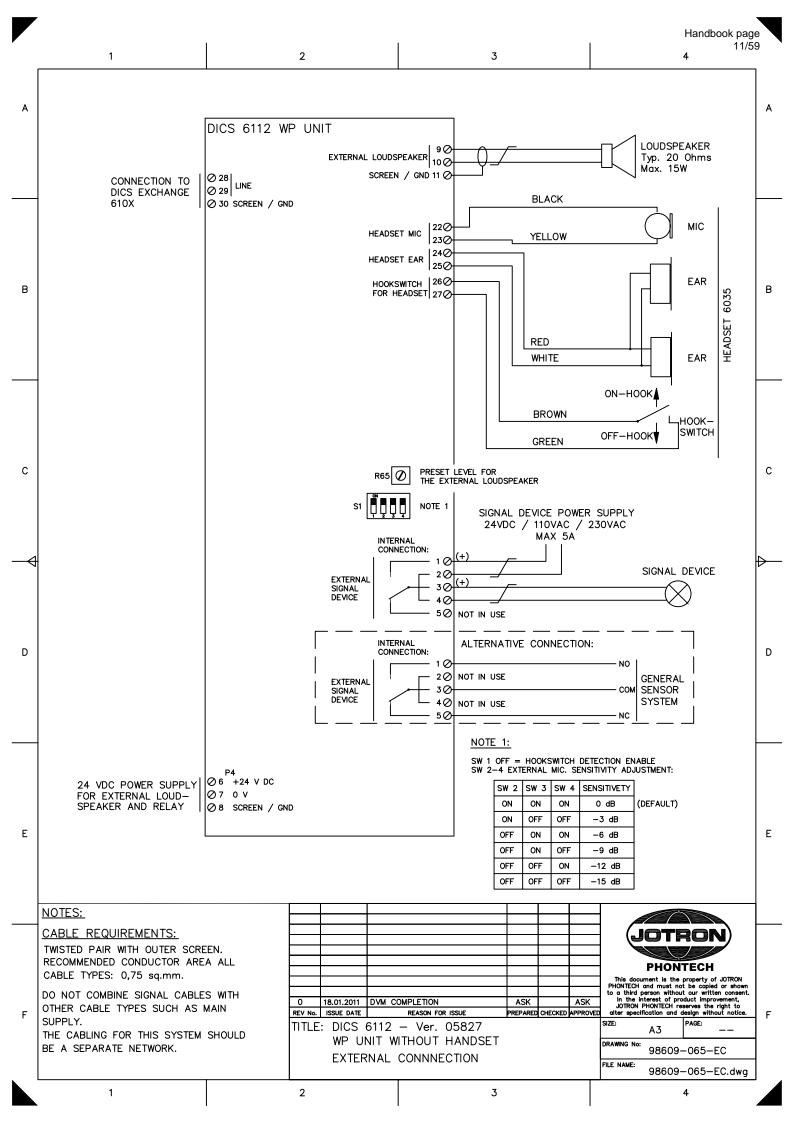


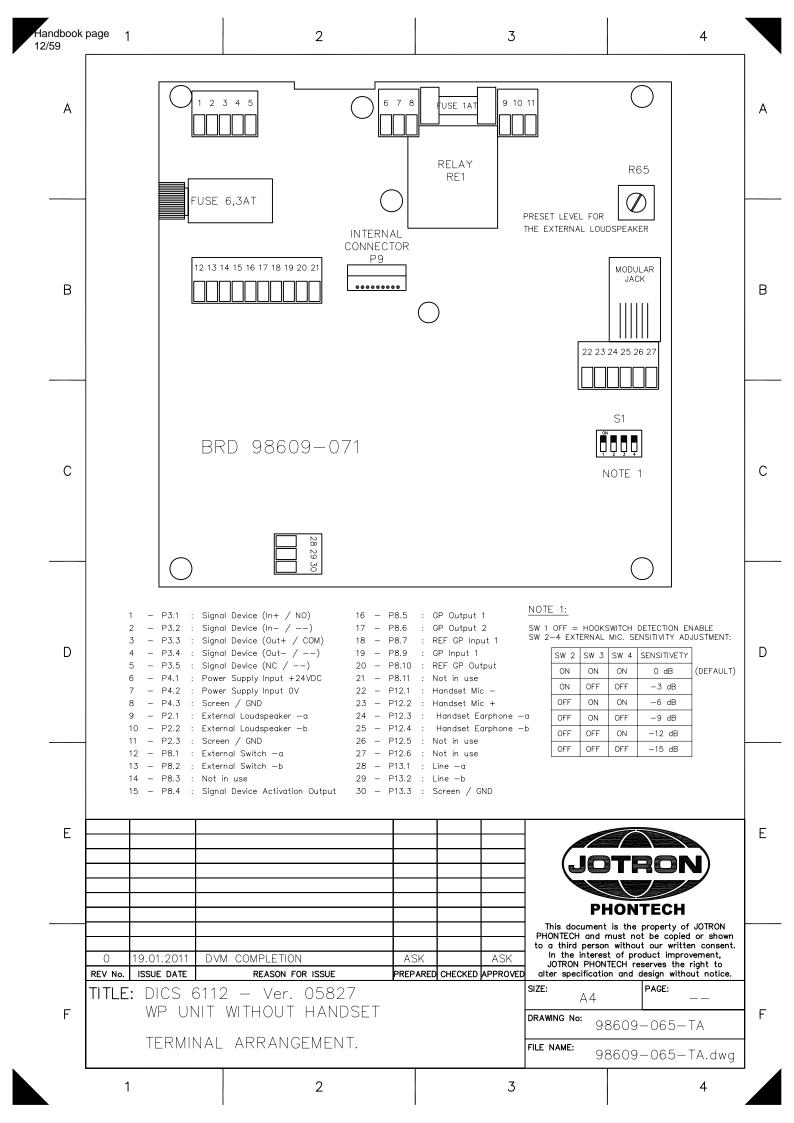


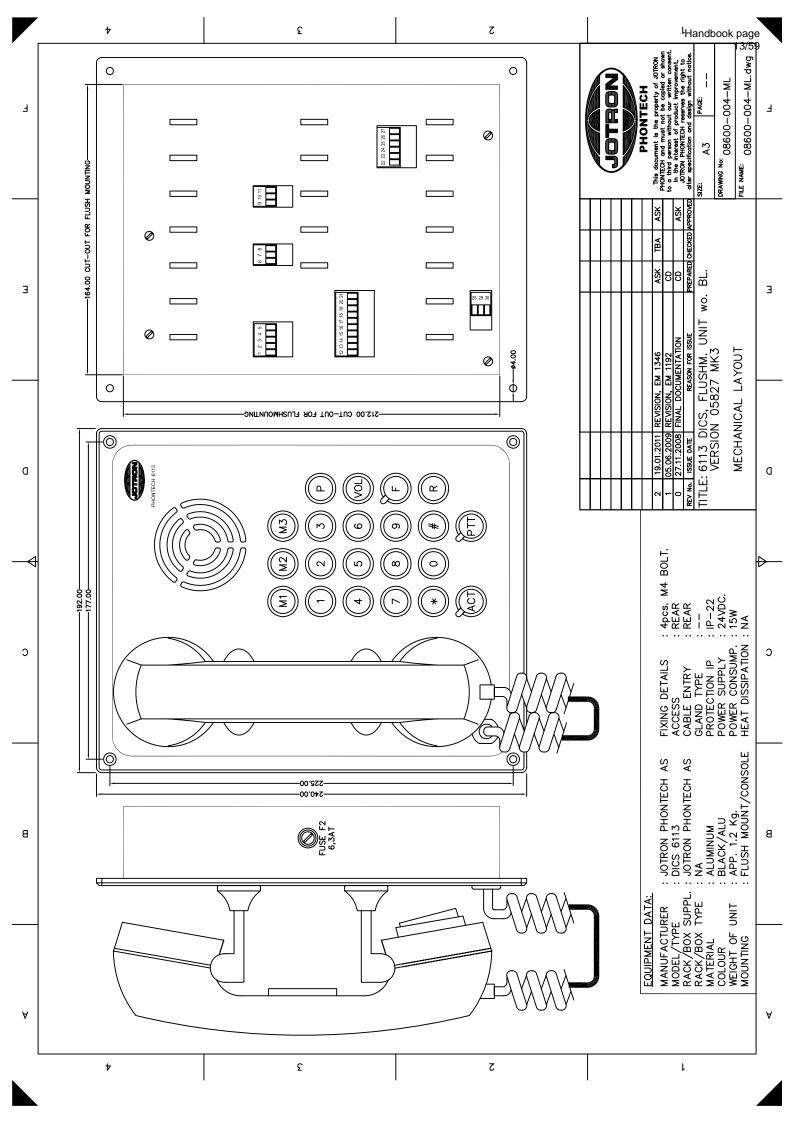


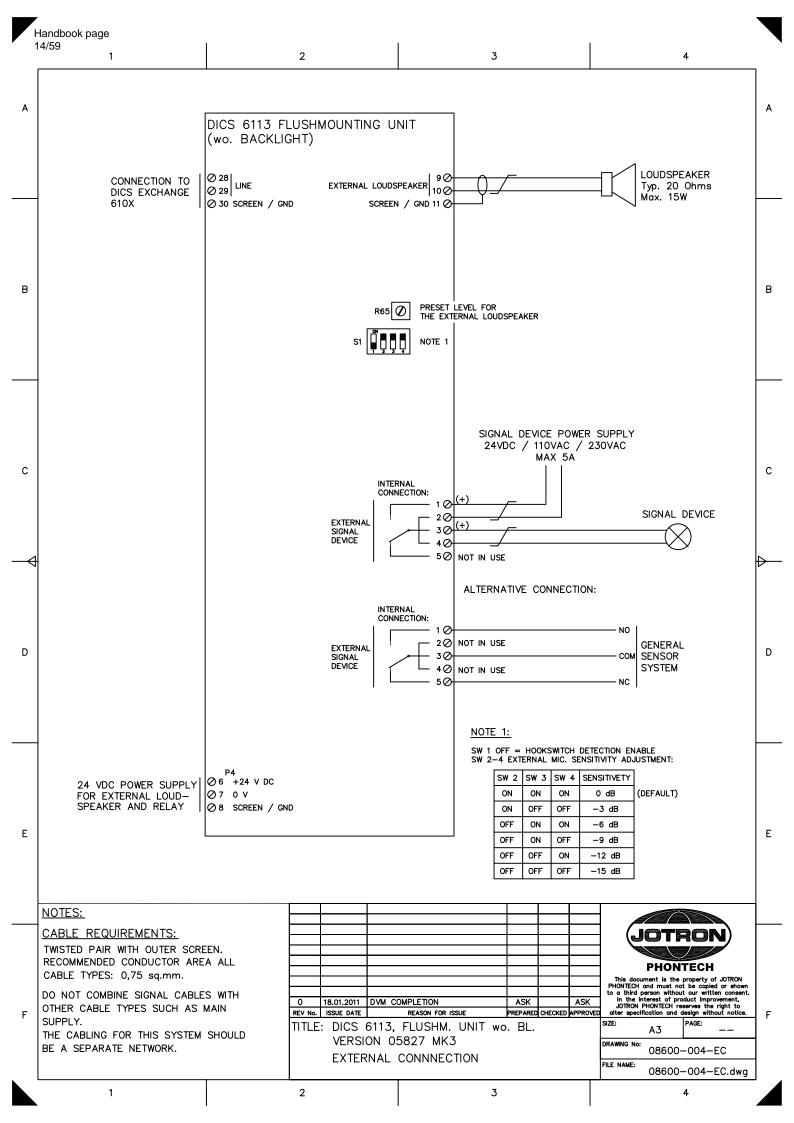


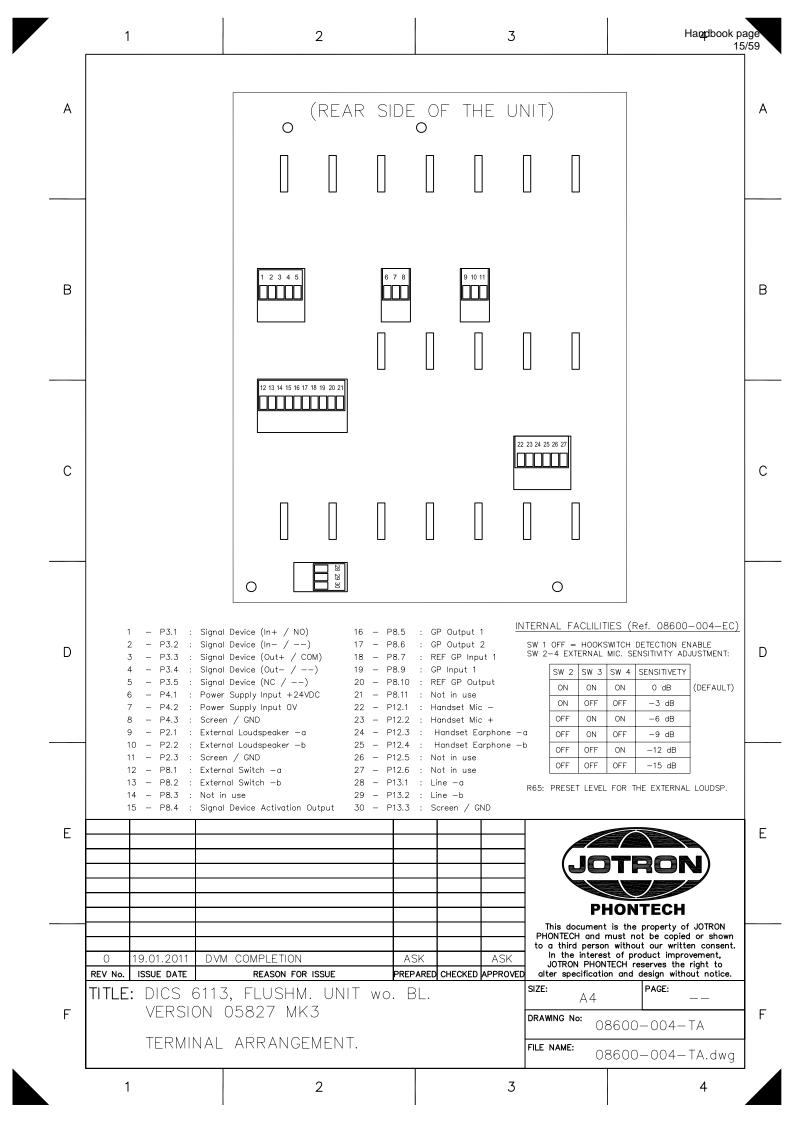


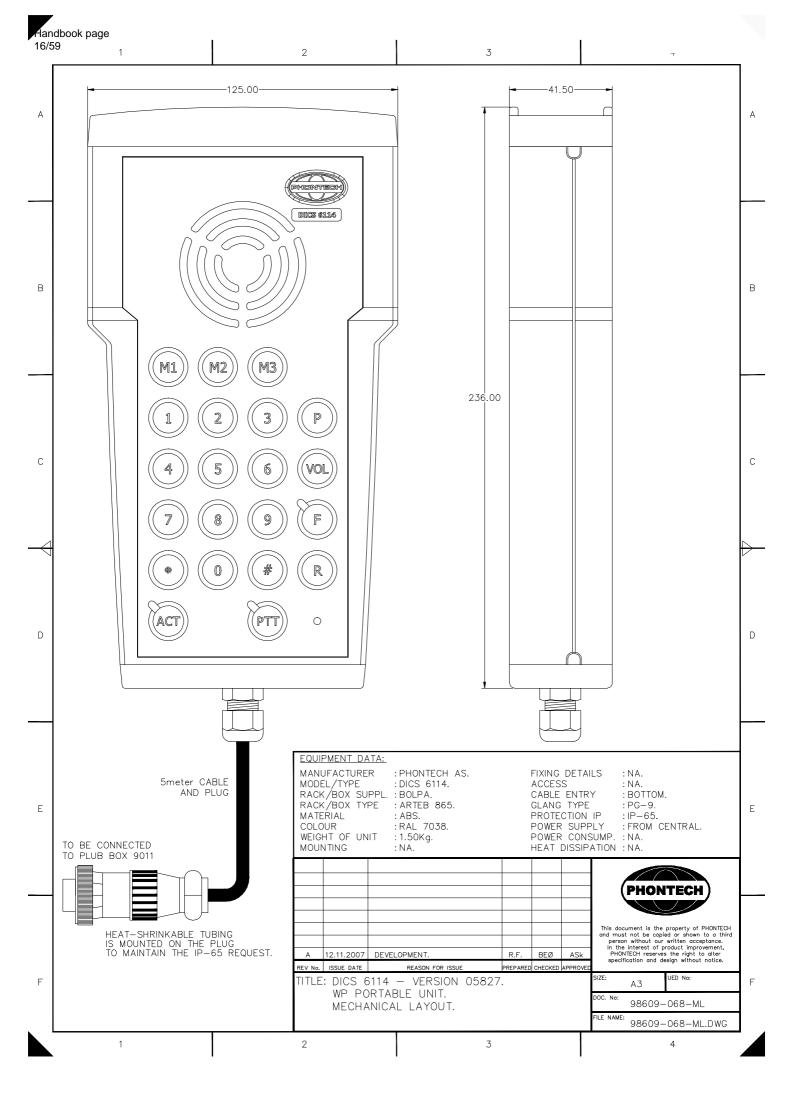


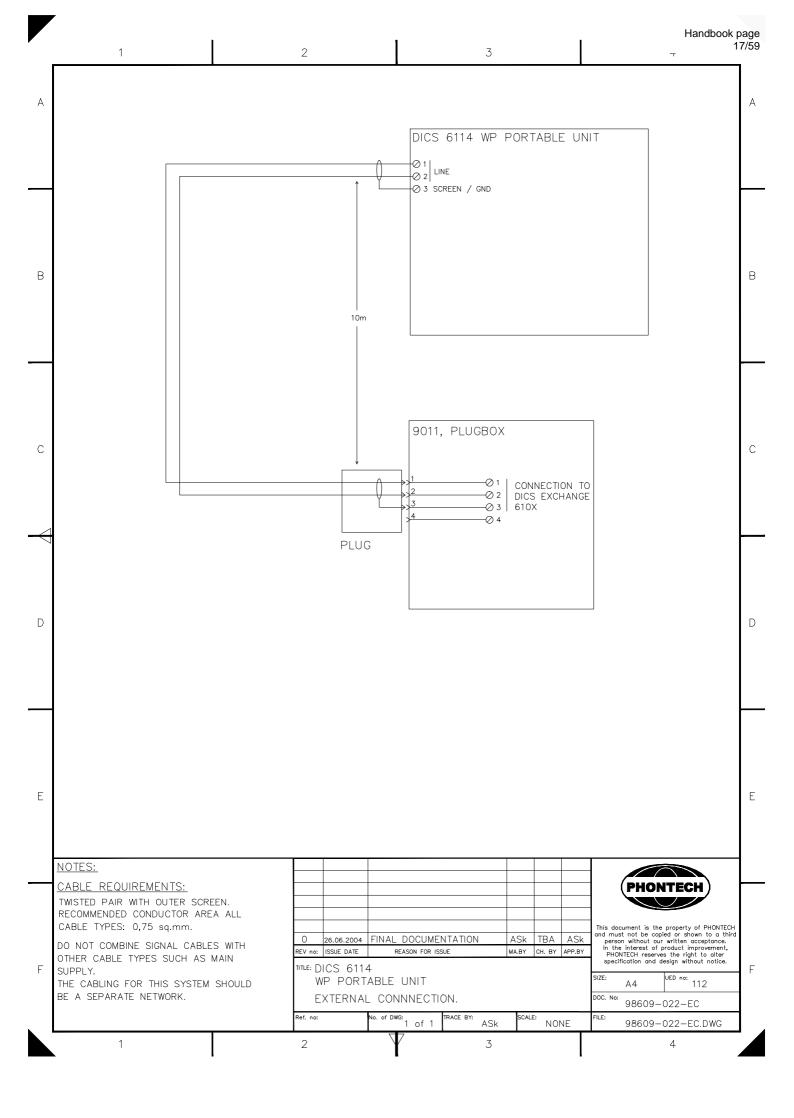


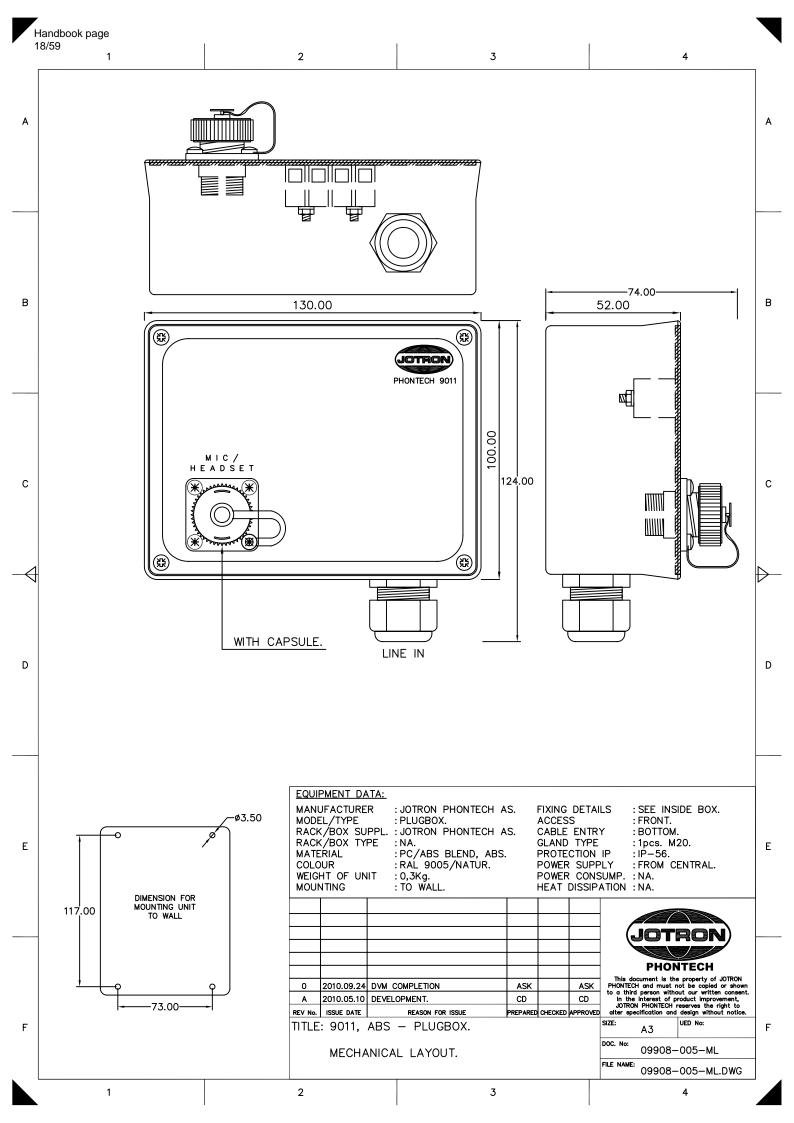


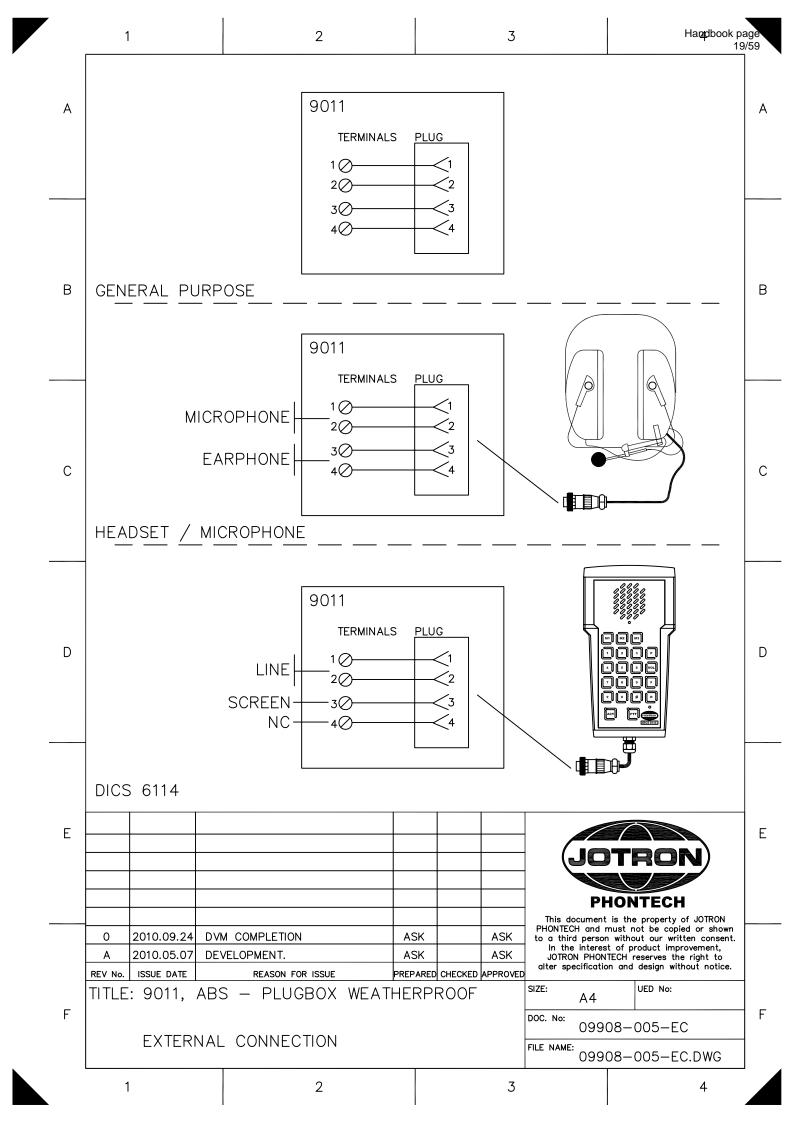


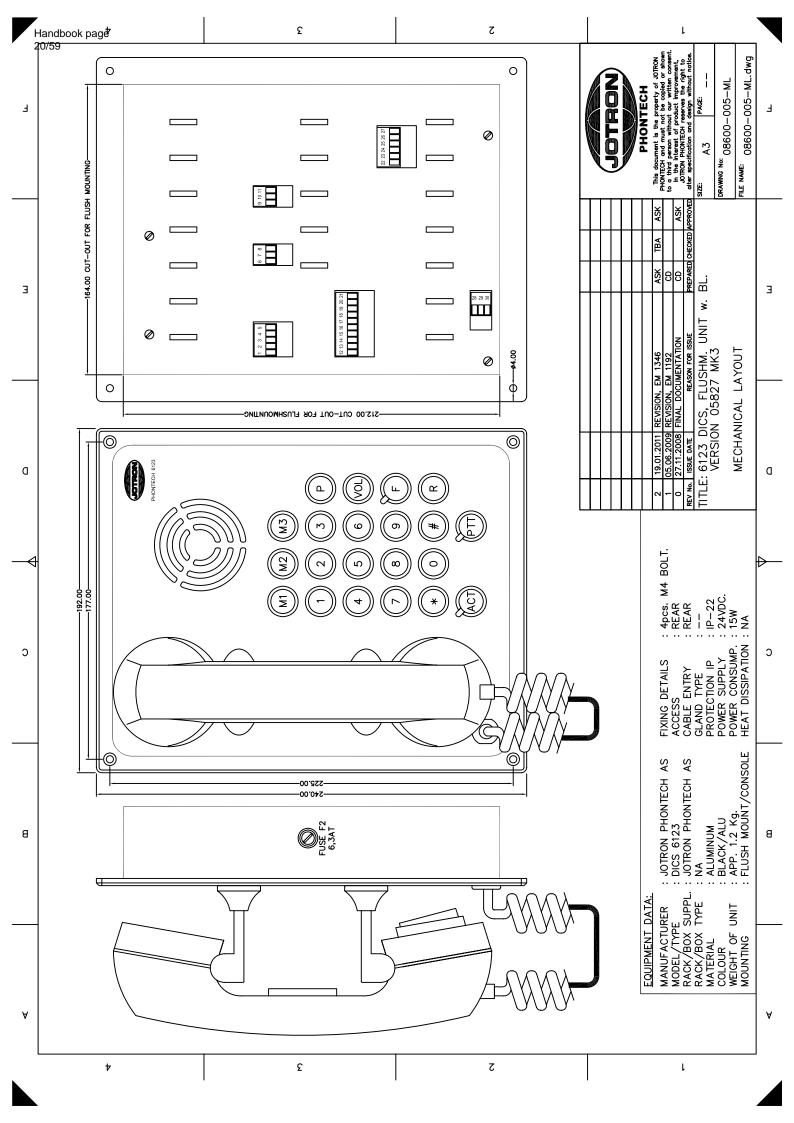


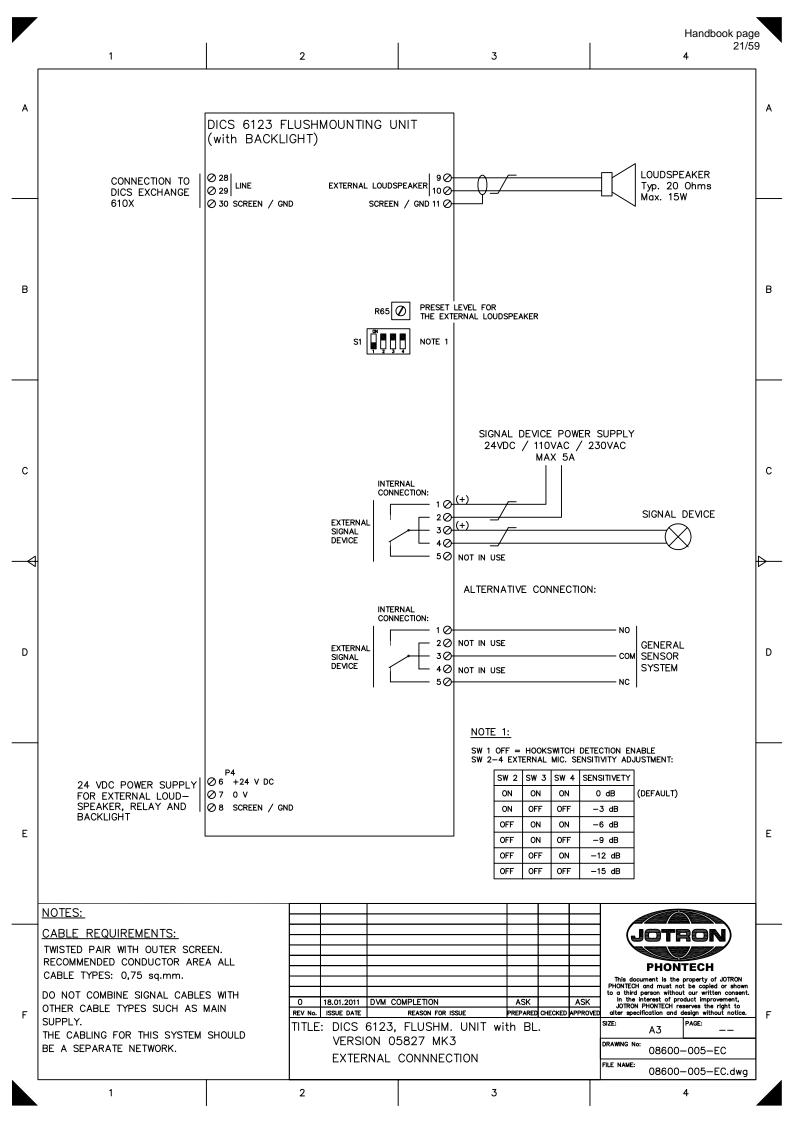


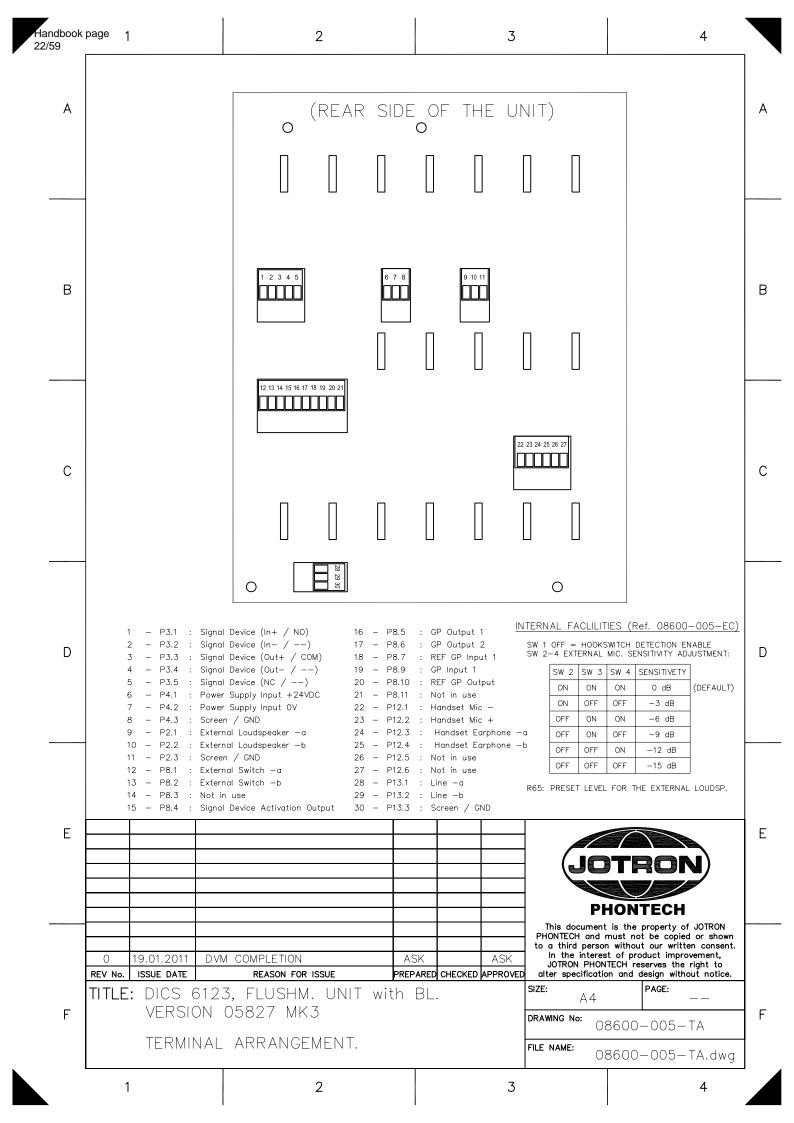


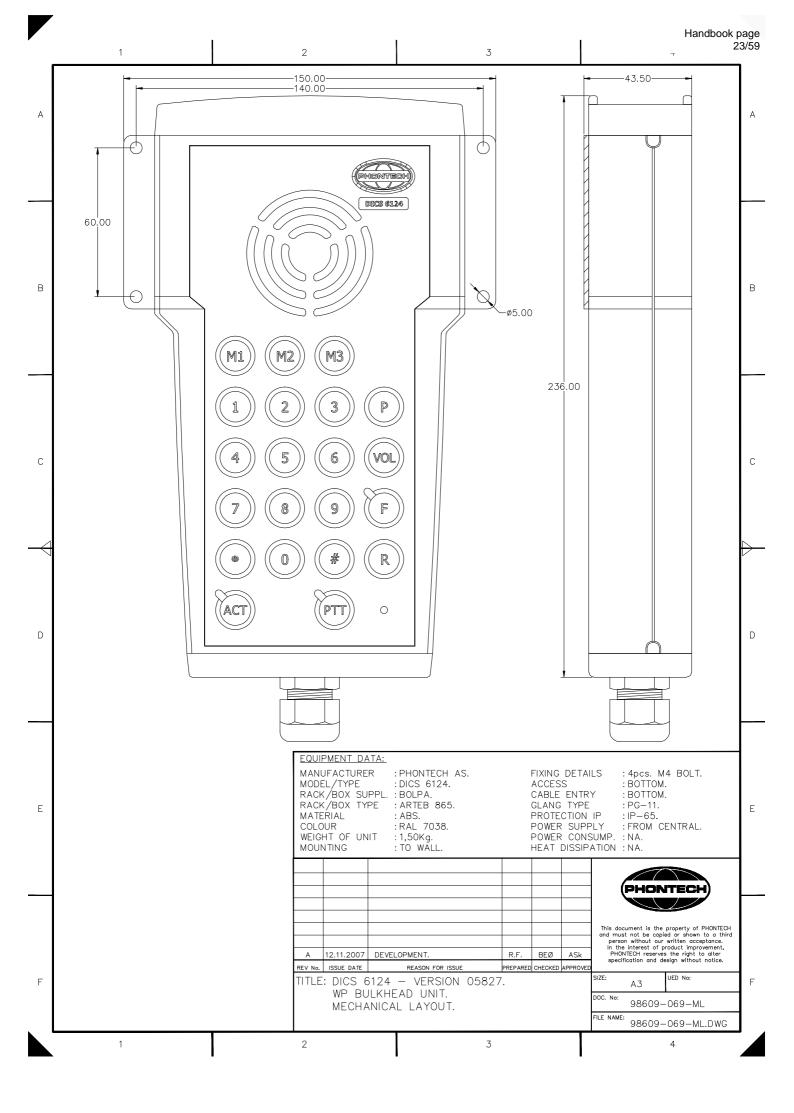


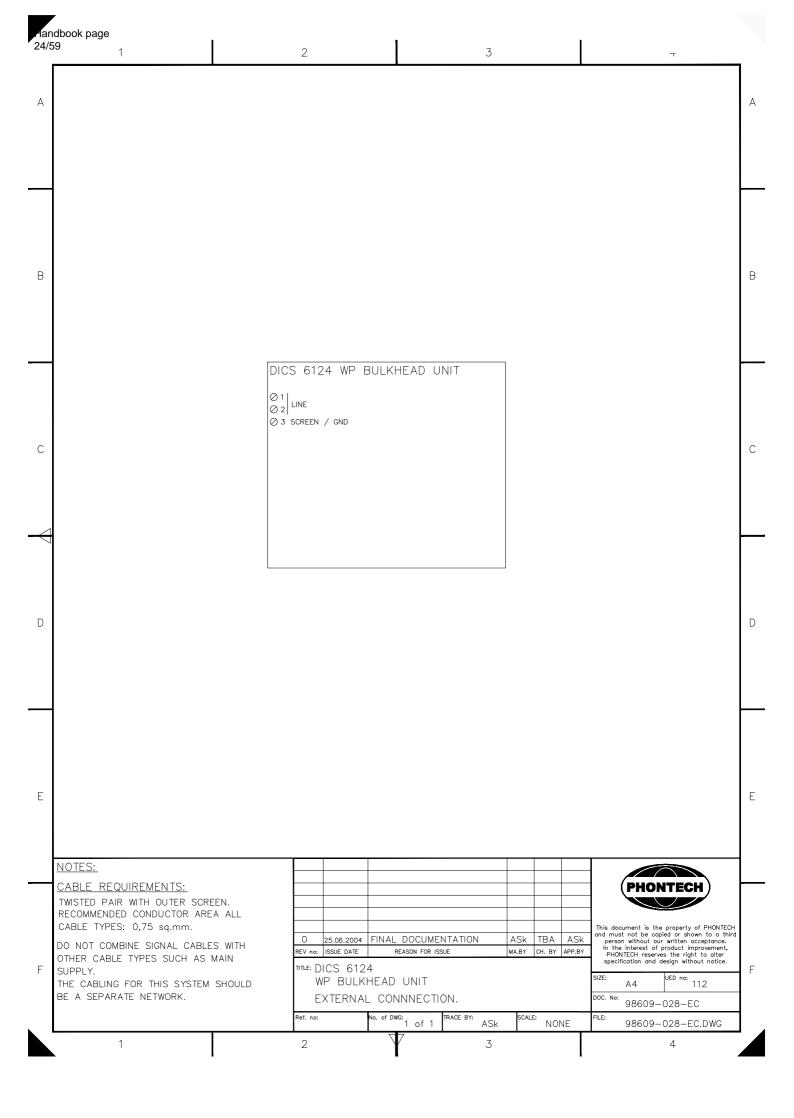














snom 300

SIP based VoIP Phone





- Backlit, two-line display
 (2 x 16 characters)
- 4 identities (lines)
- 27 keys, 7 LEDs
- 6 programmable function keys
- Speakerphone
- Power over Ethernet (PoE)
- Ethernet switch
- XML Mini-Browser

The IP telephone with the latest VoIP technology for the discerning caller! Ideal for ITSPs, offices, and private users.

As the basic model of the snom business telephone family, the SIP telephone snom 300 fulfils the requirements of VoIP telephony and additionally offers numerous functions that are indispensable for everyday working life. Due to its excellent costperformance ratio, it is ideal both for ITSP packages/products and for small and medium-sized enterprises, home offices, and private users.

The **snom 300** sets new standards when it comes to user-friendliness: A two-line LCD display shows call infor-

mation, and the menu-driven user interface provides simple feature management.

The **snom 300** is synonymous with effectiveness, time saving and flexibility: It provides all necessary office functionalities, for example choice of outgoing identity, "busy" indication, or picking up calls. Four different identities are possible.

Six user- or administrator-configurable (or carrier-preconfigurable) function keys can easily be allocated to arbitrary menu functions like DND, redirection, deny, etc., or assigned to multiple lines. The **snom 300** comes factory-equipped with two of its six

programmable keys configured as line appearances. You can configure up to all six function keys this way – flexible enough to suit the needs of every user.

Are you afraid of unsolicited eavesdroppers, data theft, or spam calls? Thanks to the **snom 300** you have no need to worry. Due to the security standards SRTP and SIPS, the **snom 300** is just as secure as the business model snom 320. This makes it perfectly suited for use in sensitive environments such as in banks and medical or military establishments.



Technical Data snom 300



GENERAL FEATURES

- **Dimensions:** approx. 200 x 185 x 120 mm
- Weight: approx. 730 g
- Certification FCC Class B, CE Marking
- Safety: IEC 60950-1:2001
- 1x LAN, 1x PC: RJ45 (Ethernet)
- Power over Ethernet: IEEE 802.3af, Class 1
- Power: PoE or 5V DC power supply listed in the user guide (1)
- **Ethernet:** 2x IEEE 802.3 10/100 Mbps
- Handset: 4P4C connector
- Headset: (wired) 4P4C connector; wireless via snom EHS Advanced (2)

USER INTERFACE

- 2x16 character backlit display
- 27 keys, 6 programmable function keys with LEDs
- Menu-driven user interface
- Localization
- Selection of ring tones, VIP ring tones, integration of customized ring tones
- Call indication with LED
- LED indication for missed calls, waiting messages, and calls on hold
- Speakerphone
- Multiple audio device suppor
- Clock with automatic daylight-saving time
- Call timer

CALL FEATURES

- 4 identities (lines)
- Directory with 100 contact entries
- Import/Export of Directory
- Speed dialing
- **URL** dialing
- Local dial plan
- Automatic redial on busy

- Call completion (Busy/Unreachable) (3)
- Caller identification (caller ID)
- Call waiting
- Auto answer
- Call blocking (deny list)
- Blocking of anonymous calls
- Lists of missed, received, and dialed calls
- Hold
- Music on hold (3)
- Handling of up to 4 calls simultaneously
- Blind and attended transfer
- Call forwarding
- 3-way conference on the phone
- Extension monitoring, call pickup (3)
- Call park, call unpark (3)
- Call completion (3)
- Multicast paging
- DND mode (do not disturb)
- Mute microphone
- Keyboard lock
- Client Matter Code (CMC) (3)
- Unified Communications ready

WEB SERVER

- Embedded web server HTTP/HTTPS
- Remote configuration/provisioning
- Dial from web interface
- Password protection
- Diagnostics (tracing, logging, syslog)

SECURITY, QUALITY OF SERVICE

- HTTPS (server/client)
- Transport Layer Security (TLS)
- SRTP (RFC3711), SIPS
- RTCP, S-RTCP
- VLAN (IEEE 802.1X)
- LLDP-MED

CODECS, AUDIO

- G.711 A-law, µ-law
- G.722, G.726, G.729A, G.723.1, G.727

- GSM 6.10 (full rate)
- Comfort noise, voice activity detection

SIP

- RFC3261 compliance
- UDP, TCP and TLS
- Digest/basic authentication
- PRACK (RFC3262)
- Reliability of provisional responses (RFC3262)
- Early media support
- DNS SRV (RFC3263), redundant server support
- Offer/answer (RFC3264)
- Message Waiting Indication (RFC3842), subscription for MWI events (RFC3265)
- Dialog-state monitoring (RFC 4235)
- In-band DTMF/out-of-band DTMF/ SIP INFO DTMF
- STUN client, ICE (NAT traversal)
- ENUM (RFC3261), NAPTR (RFC2915), rport (RFC3581), REFER (RFC3515)
- Bridged line appearance (BLA) (3)
- Auto provisioning with PnP (3)
- Busy lamp field support (BLF) (3)

INSTALLATION

- Automatic software update
- Automatic settings retrieval via HTTP/HTTPS/TFTP with authentication
- Installation via web interface
- Remote management via TR-069/TR-111
- Static IP provisioning, DHCP
- NTP

(1) Available separately: Power supply PSAC10R-050 (snom PN 00002730)

(2) Available separately: snom EHS Advanced V2.0 (PN 00002362)

(3) If supported by PBX

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TELEKOMMUNIKATION

SICHERHEITSTECHNIK

GEBÄUDEMANAGEMENT

DATENTECHNIK

FEP-IP 4

Flush-mounting Phone for VoIP

The flush-mounting phone FEP-IP4 is designed for using in a Voice-over-IP (VoIP) environment directly. Everywhere a highly available phone is required in a professional environment, FEP-IP4 is the solution:

- Integrated in control panels
- Used in naval architecture
- As emergency phone on the wall
- Integrated in sales booth with self-service for customer information
- In IT racks

FEP-IP4 fits to LAN directly. It uses the standard SIP protocol for the interconnection to a Soft-PBX.

For professional applications with highly available requirements there is an additional interface at the phone: A standard POTS-interface (analog tip/ring). So, if VoIP doesn't work, a connection to the public network remains on-line.

The dimensions of FEP-IP are identical with the long-time available FEP-2010, our flush-mounting phone for the POTS-line (analog tip/ring).

Version options of FEP-IP4:

- Handset with magnetic clamp for vertical installation.
- 4 Screw holes for wall or desk mounting.
 (Opening for assembly: L=260 x W=90 mm)
- Illuminated keypad
- One key only to start calling to a predefined destination (Without keypad).
- Handset with Push-to-Talk (PTT) or Push-to-Mute (PTM) key.
- Custom-specific dimensions and surface of the front.
- Helix cable for handset pluggable (RJ 10)

General features of the FEP-IP4:

- LAN-interface: Ethernet 100/10 Base-T(x) with IP-Protocol , RJ45 jack
- VoIP-signalling SIP (RFC 3261)
- Speech coding G.711
- Fallback: POTS interface (analog tip/ring)
- DTMF post-dialling according to CCITT Q.23 (RFC2833 or inband signalling)
- WEB-based configuration
- Optical call indication
- Switch off option for ringer (optical call indication only)
- 10 short-code dialling destinations
- Redialling
- Callback of unanswered calls (optical indication)
- Remote dialling using TCP/IP connection
- Option: Automatic call to a predefined destination after hook-off
- Option: Ringing only on tolerated callers
- Front surface / handset colour options :
 - grey (RAL 7032) / grey
 - beige (RAL 1015) / beige
 - black (RAL 9005) / black
 - Aluminium anodized natural (C1) / black
- Handset with helix cable, 4 wire, 30 cm length (other length on demand), hard-wired to FEP-IP4
- Dimension FEP-IP4 standard:
 - Outer diameter: L=280 x W=110 x H=50 mm
 - Opening for assembly: L=270 x W=100 mm
- PoE according to IEEE 802.3af (Kl.0) or external 12V-/24V-DC power supply (3 W)

Add-ons for FEP-IP4:

- Plastic chassis for wall mount
- PoE injector



V 4.02 Subject to technical changes

Fig. left: FEP-IP4 flush mounting phone

Fig. right.: FEP-IP4 phone with magnetic clamp for handset and wall-mount chassis



FON E-MAIL

info@tas.de

INTERNET



FEP-IP 4 with Display

Flush-mounting phone for VoIP with additional features

Special features for FEP-IP4 with display:

- LCD-Display with backlight, 5 lines
- User menu with navigation over 4 keys
- Phone directory (100 destinations)
- Up to 15 keys for direct calls (user labeled)
- 2 status LED for direct calls (called, calling, waiting, lost call)
- Interface for headset
- Open listening
- Automatic call accept (Headset or open listening)
- Surface black RAL 9005
- Handset black
- Handset with helix cable, 4 wire, 30 cm length (other length optional), pluggable (RJ 10)
- Desk mount using 4 screw wholes
- Dimension FEP-IP with display:
 - Outer diameter: L=280 x W=190 x H=60 mm
 - Opening for assembly: L=260 x W=170 mm
- PoE according to IEEE 802.3af (KI.0) or external 12V-/24V-DC power supply (6 W)



Fig.: FEP-IP with display, keys for direct call and handset with magnetic clamp

Version options of FEP-IP4 with display:

- Handset with magnetic clamp for vertical installation.
- Handset with Push-to-Talk (PTT) or Push-to-Mute (PTM) key.
- Custom-specific dimensions and surface of the front.

FEP - IP 19"

Flush-mounting phone for 19" Racks

The flush-mounting phone FEP-IP 19" is designed for implementation in a control cabinet. Directly connected to a LAN switch this VoIP-phone is a perfect solution for phone calls in computing rooms

Special features FEP – IP 19":

- Surface black RAL 9005
- Handset black
- Dimension FEP-IP 19": 2HE
- PoE according to IEEE 802.3af
- Custom specific options on demand



Fig.: FEP - IP 19" for a control cabinet with custom specific options

FON E-MAIL INTERNET

0 21 66 / 858-0

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Weatherproof VoIP-Telephone FernTel IP

VoIP-Telephone for indoor and outdoor use

- ► Protection degree IP 65 acc. to IEC60529
- ► Ambient temperature -20 °C to +60 °C
- ➤ Call tone ≥ 95 dB(A), 1 m
- ► Pixel-based, illuminated LCD Display
- Illuminated keypad
- ► Intelligent and user-friendly menu structure
- ► Standard H.323, SIP, TSIP, SIPS protocols
- Power supply Power over Ethernet
- ➤ Connection to 10/100-BASE-T Ethernet LAN, RJ-45

Application

With approved technology of FHF, the FernTel IP is been suitable for all outside installations.

The new FernTel IP is the ideal telephone for many differential work areas – seawater, high air humidity, dust and not exactly gentle mechanical stress are no problem for it

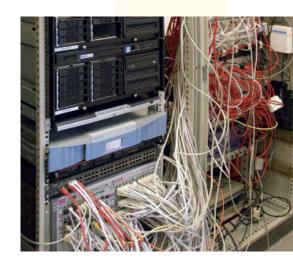
The stylish housing is made of impactresistant and shockproof plastic moulding. Even acids, alkalis or lubricants cannot damage the high quality components of our versatile desk / wall telephone FernTel IP. Its modern design is the perfect "packing" for the finest high-tech.

Its striking signal colours ensure the FernTel IP cannot be missed whenever a telephone is urgently needed, e.g. in emergencies in poor weather and light conditions.

The FernTel IP is suitable for almost universal use thanks to its amazing transformability. A deft hand movement and the desk telephone for indoor use is converted into a wall telephone for outside use.

The FernTel IP makes possible effective working with high comfort. The illuminated keypad and display finish the comfort. The standardized features according to H.450 are supported. The Ethernet connectivity supports with the internal 2 port switch connecting a laptop in an outdoor area. The laptop gets a network access through the telephone. The FernTel IP offers qualitative high-grade features according to industrial standards instead of proprietary solutions.

The FernTel IP is the ideal telephone for many differential work areas.



Technical specifications

Housing Polycarbonate

Housing Dimensions height x width x depth 293 x 191 x 128 mm

Weight approx. 2.3 kg

Protection Degree Acc. to IEC60529 IP 65

Power Supply Power over LAN (IEEE 802.3af)

Connection 2x RJ-45, Ports, 10/100-BASE-T Ethernet LAN

Ringing Volume approx. 95 dB(A) at 1 m distance

Display 128 x 64 Pixel

Temperature Range -20 °C to +60 °C

Protocol H.323, SIP, TSIP, SIPS

11.020, 011, 1011, 011 0

total H.323 version 4 incl. H.225, H.235, H.245

and RAS Gatekeeper routed Signalling, H.450 Session Initiation Protocol (SIP) RTP, SRTP Real Time Protocol

RTCP Real Time Control Protocol – first level of Quality of Service

RAS Protocol Support for External Gatekeeper

DTMF H.245 Alphanumeric or Signal Type

Additional VoIP-Features H.245 Fast Connect En-block dialling Overlapped Sending

Security Password Protected Administration

Encoded Password Authorization acc. to H.235

Quality of Service Priority of IP-Packages acc. to TOS and DiffServ,

VLAN Priority acc. to IEEE 802.1p / 802.1q

Voice Encoding G.711 A-law / μ -law (64 kbps), G.723.1 (5.3 kbps),

G.729A (16kbps)

Echo Compensation G.168

Access via HTML Web-Browser

Password protected authentication

Troubleshooting Log- and Trace-Files, State Display of Interfaces and

Connections, Ping Connection Test sending of SNMP Traps

over Internet Protocol

Update Configuration recording/reading,

Boot code and firmware update via HTML upload,

Automatic update via Update-Server

DSL-Access PPPoE Protocol

VPN Tunnelling with PPTP Encoding via MPPE

NAT Network Address Translation – for Transformation of official

IP Addresses into private IP Addresses and vice versa

DHCP Dynamic Host Configuration Protocol –

IP interfaces settings

ICMP Internet Control Message Protocol – for Ping tests

Dial Tone Generation Automatic Dial tone Generation European

and US Standard

Call Transfer Call Transfer with/without consultation call
Call Diversion Call Diversion Unconditional, Busy, No Reply

Call Hold / Retrieve Call Hold / Retrieve

Message Waiting Message Waiting Indication

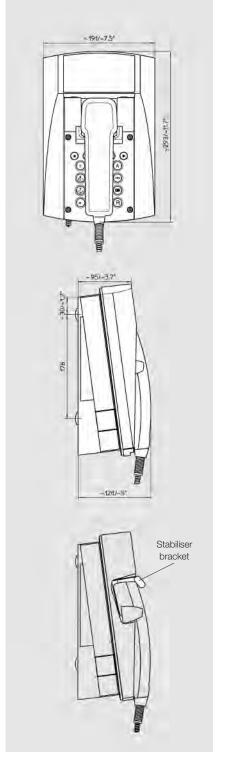
Calling Name Identification Name Display

3 Party Conference 3 Party Conference of internal and/or external Subscriber

Calling Number Identification Display of Calling Number
Multiple Registrations Up to 6 Registrations

Telephone Book Local, Integration of an External Database

Time/Date Exact Time and Date Information via Time Server







ExResistTel IP2 Explosion-proof VoIP Telephone

IP Telephone for indoor and outdoor use in zone 1 and 2

- ▶ IP 66 protection class as per IEC60529
- ➤ Ambient temperature range -40°C to +60°C (heated display)
- ▶ Ring signal \geq 95 dB(A) at a distance of 1 m
- ► Pixel-based illuminated LCD display
- V4A alphanumerical keypad
- Web based monitoring (operating, handset, hands free, ringing)
- Intelligent, user friendly menu structure
- ► Standard protocols H.323, SIP, TSIP, SIPS
- ► Power supply: PoE or external supply
- ► Connection to 10/100-BASE-T Ethernet
- Hands free communication



Application

Proven technology from FHF makes the ExResistTel IP2 suitable for all indoor and outdoor applications in hazardous areas.

The new ExResistTel IP2 is the ideal unit for all kinds of weather conditions at a wide variety of very diverse facilities – whether sea water, high humidity or extreme mechanical demands.

The housing is made of impact and shock resistant fiberglass-reinforced polyester. Even acids, alkalis or lubricants have no effect on the housing. Its robust design is the perfect "packaging" to meet the latest requirements demanded of VoIP telephones for use in hazardous areas. It is always available when a telephone is urgently needed, such as in emergency situations.

The ExResistTel IP2 makes work more effective by providing especially convenient telephone services.

An illuminated, heated display rounds out the convenience features of the ExResistTel IP2.

It also supports all features of the H.450 standard.

The ExResistTel IP2 offers high-quality features based on industry standards and our decades of experience.

A headset, available as accessory equipment, can be easily connected to the telephone. A handsfree function is also integrated into the unit.

Ex-Telephone for outdoor facilities

Proven technology from FHF makes the ExResistTel IP2 suitable for all zone 1+2 applications.



Features

Display 182 x 64 pixels

Protocols H.323, SIP TSIP SIPS

General H.323 version 4 including H.225, H.235, H.245 and RAS

Gatekeeper routed signalling, H.450, Session Initiation Protocol (SIP) RTP,

SRTP Real Time Protocol – for voice data transmission

RTCP Real Time Control Protocol – first level of "Quality of Service"

RAS protocol Support for an external gatekeeper

DTMF H.245 "Alphanumeric" or "Signal Type"

Additional VoIP features H.245 fast connect en-bloc dialing overlapped sending

Security Encrypted password authentication as per H.235

Quality of Service IP packet prioritization via TOS and DiffServ

VLAN priority as per IEEE 802.1p / 802.1q

Audio codecs G.711 A-law / μ-law (64 kbps), G.729A (16 kbps)

Echo compensation G.168

Access HTML via web browser

Password protected with secure authentication

Troubleshooting Log and trace files and status display of interfaces and connections

Ping connection test for Internet Protocol, sending of SNMP traps

Updates Configuration save and restore,

Boot code and firmware updates via HTML upload

Automatic updating via update server

DSL access PPPoE protocol

VPN Tunneling with PPTP encryption with MPPE

NAT Network Address Translation – translates public IP addresses into private local

address space addresses and vice versa

DHCP Dynamic Host Configuration Protocol – sets up the IP interfaces

ICMP Internet Control Message Protocol – for ping tests

Call signal generation Automatic call signal generation as per European and US standards

Call Transfer in all common variants: with/without asking, before/after answering, etc.

Call diversion Call Diversion / Redirection

Call hold Call Hold / Retrieve

Call Waiting With corresponding signaling to calling party

Message Telephone displays that a message is waiting
Pickup Telephone displays that a call can be picked up

Pickup list Telephone displays a list of calls that can be picked up

Name display For signaling which name should be displayed

Call back Call Completion with all common variants such as call back when busy and

call back when free

3-way conference With 3 parties, also external parties

Caller ID For special signaling of individual phone numbers or phone number groups

Multiple registration Maximum of 6 registrations

Telephone book All registrations available automatically from central telephone book,

External databases integrated via LDAP

Time Precisely accurate time data via time server access

Technical specifications

Connection Data

Powered via Power over Ethernet as per IEEE 802.3af using the spare data lines or via external power supply

Voltage of external power supply 19.2 V - 52.8 V DC

Power consumption PoE (class 0) 12.95 W

Connection Screw terminals (10/100 Mbit/s)

Ring signal volume approx. 95 dB(A) maximum

at a distance of 1 m

Housing (height x width x depth) 293 x 227 x 135 mm Weight (standard model) approx. 5.000 g

Display 182 x 64 pixels

Mounting position Vertical wall mounting

Switching capacity of relay 250 V AC, 5 A 30 V DC, 5 A 50 V DC, 1 A

230 V DC, 0.5 A

Handset

Voice capsule Electret microphone
Earpiece capsule Dynamic capsule

with magnetic field generator

Handset securing mechanism in cradle Standard equipment

Environmental Conditions

Ambient operating temperature -40 °C to +60 °C

Conformity

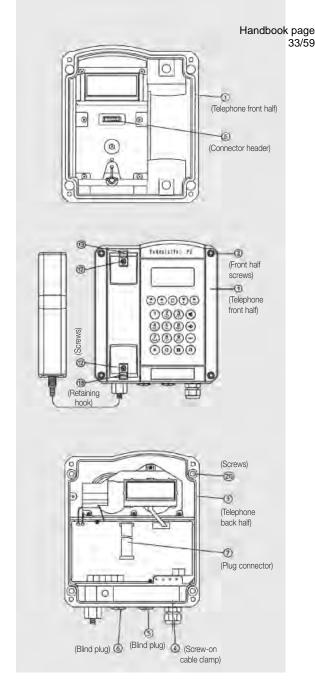
Types of protection

Protection class IP 66 as per IEC 60529

Impact resistance IK 09 as per EN IEC 62262:2002

II 2G Ex e ib [ib] mb IIC T4 Gb

III 2D Ex ib [ib] tb IIIC T 135 °C Db

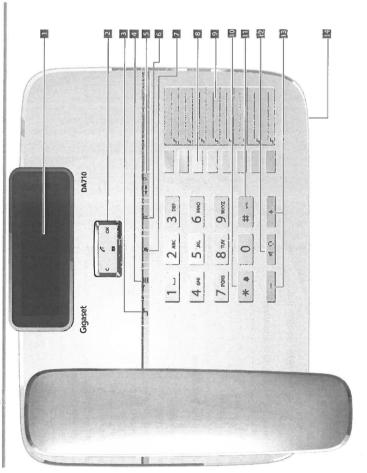


Order information

Туре	Designation	Housing Color		Art. No.
ExResistTel IP2	VoIP Telephone	Black	with relay contact	112 861 80

Subject to change without notice \cdot Printout 06/14



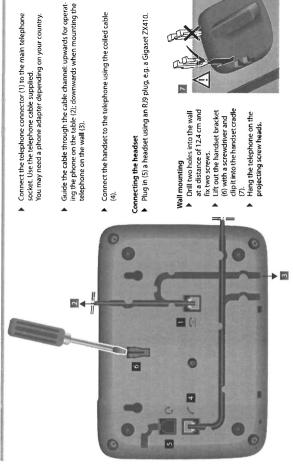


13 Keys for setting the volume of the handset, headset, speaker and ring tone 14 Microphone for hands-free 1 Adjustable display (Changing the language → page 11)
2 Control key
3 Shift key
4 Menu key
5 Redalfpause key
6 Recall key
7 Mute key

Display and keys

ΓED

Connecting the telephone



Notes

First use

Lift the handset for 5 seconds and place it back in the cradle.
 Lift the handset again. You will hear the free tone and the device is now ready to be used.

The phone is connected to the power source via the phone line. In case of a power cut (e.g. if the PBX is switched off at night), the above mentioned steps need to be repeated. Phonebook entries and speed dialling destinations are stored indefinitely.

The device is designed to be used as a single-line system (on the main extension or a telephone system). It cannot be operated as a second phone on a line splitter.

Operation on a PBX or router

The private branch exchange (PBX) or the router must supply the phone with continuous DC voltage, even when ringing. If this is not the case, the phone may switch off briefly when ringing. As a result, saved information may be lost. Refer to the operating manual for your PBX or router for information in this regard or contact the manufacturer.

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8 Direct dial key
9 Insert strip for writing the assignment of direct dial keys
10 Star key, selecting/deselecting the ring tone (press and hold)
11 Hash key, selecting/deselecting the key lock (press and hold)
12 Hands-free/headset key

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Safety precautions

When installing, connecting and operating the telephone, always observe the following precautions:

- Only use the jacks and leads supplied.

 Only connect the connecting lead to the intended socket.
 Only connect the approved accessories, such as the Gigaset ZX410 headset for example.

 Your Gigaset DA710 has been tested and approved with the Gigaset
- ZX410 headset.

 Other headsets may only function to a limited extent or not at all. Be sure to reduce the volume of the Gigaset DA710 when connecting sure to reduce the volume of the Cigaset DA710 man and aman and a man and a another headset. As a headset from another manufacturer may emit a harmful and loud accounts ragnal due to alignment differences. Lay the connecting lead where it will not cause accidents. Position the telephone on a non-slip surface.
- For your safety and protection, the telephone may not be used in the bath or shower rooms (wet locations). The telephone is not splash
 - other electrical devices.

 Protect your telephone from moisture, dust, corrosive liquids and proof. Never expose the telephone to sources of heat, direct sunlight or •

77

Index

- Never open up the telephone yourself. vapours.
- Do not touch the plug contact with pointed or metallic objects. Do not carry the telephone by the leads.
- If you give your Gigaset DA710 to someone else, make sure you also give them the operating manual. Your Gigaset DA710 has a permanent memory, so you may want to delete any stored numbers before

Setting up the telephone for use

Recommended installation of telephone:

- Do not expose the telephone to direct sunlight or other sources of
- Operate at temperatures of between $+5^\circ$ C and $+40^\circ$ C. Mantain a distance of at least one metre between the telephone and radio equipment, e.g. radio telephones, radio equipment, e.g. radio telephones, radio paging equipment or TV sets. Otherwise telephone communication could be
- Do not install the telephone in dusty rooms as this can limit the service life of the telephone.
 - Furniture lacquer and polish can be adversely affected by contact with parts of the unit (e.g. feet).

Operating the telephone

The keypad and the display are available to operate your Gigaset DA710. In addition, the display ment together with the control key provides access to the configuration and use of the many functions of the telephone. All usage possibilities are described in the handbook.

Note
Some of your telephone's functions may only be available without restriction if they are supported by your and the caller's network provider, e.g. functions for which information on the caller's telephone number is required.

This applies for the following functions, for example:

◆ Displaying the number of an incoming call (↑ page 8)

◆ Inflat(↑ page 1)

◆ Call forwarding and other network services (↑ page 13)

Display

In stand-by mode, the time, date and calendar week are displayed. You can set the date and time and the format in which it is displayed (+ page 11). Symbols on the display also provide information regarding the status of your telephone.

— Toolbar	— Time (24h format)		Date / Calendar week
₩ di∧ C	14:00	23.08.2011 CW34	

Display symbols

Flashes for a new call in the call list if the caller's number transmission (CLIP) is active (→ page 13)

- Key lock activated Phonebook open Shift key pressed ੂ ਪ ੇ ।
- Flashes on incoming VIP call Is indicated if at least one entry with activated VIP mode is present in the phonebook (→ page 9). Ring tone switched off
 - ₩ C
 - Mute mode activated Headset connected

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Control key

The telephone's functions are activated and the menu is navigated using the control key. The functions offered depend on the operating

		In standby:	_)		rds/downwards.	Hold down: Scrolle the list fact unwards/downs
ï.	•	E [L			in menus and lists:	Scrolls one entry upwards/downwards.	Hold down. Crolle the
Situation.		-	-5				

list oook

Scrolls one entry upwards/downwards. Hold down: Scrolls the list fast upwards/downwards.

Hold down: Moves the cursor fast to the left/right. While editing names and numbers: While pressing the direct dial key: Moves the cursor to the left/right.

To exit the menu without saving any previous entries simply lift and then hang up the handset. After a short time the display will then

selection.
Press (in order to return to the previous menu level.
Press the () key to return to the start menu.

The steps to select a menu function are set out in these operating

▶ ■ Phone setup

This is accessed as follows: instructions as follows:

Description of the menu selection steps

once again show the standby mode.

Displays long phone numbers completely. In menus, lists and with direct dial keys: Exits the menu, list or direct dial key.

Terminates the action without changing the setting. Deletes the character to the left of the cursor. While editing names and numbers: Hold down: Deletes the line. While changing settings:

Does the ment using the ■ key.
 Using the control key □ select the entry Phone setup and press the right-hand side of the control key □ (OK).
 Using the control key □ (SE) select the entry Language and press the right-hand side of the control key □ (OK).

Phone number entered or direct dial key pressed: Dials the phone number. If the entry is empty: Exits the editing mode.

In menus and lists:

Saves the edits or activates the selected settings. While editing entries/changing settings: Opens the submenu/context menu.

Writing and editing text

Press the [II] key to open the menu.
 Select Call list (CLIP) or Phonebook or Redial in order to open the desired list.

Using the menu

keys.

Open the phonebook using the control key 📺 Open the call list using the control key 🗂

Using keys

Phonebook, call list and redial list can be opened using the menu or

Phonebook, call list and redial list

Enter text using the keypad.

Multiple letters and numbers are assigned to each key between L1—J and [9]... and [1]... The entry of a specific character is carried out by pressing the corresponding key multiple times. You will find a table of applicable characters in the Appendix → page 16. To switch between capilel letters (ARC), lower case flood and numbers (123), press the shift key [—] until the desired figure is briefly displayed on the bottom right. This mode is not available in lines for telephone number entry.

Scroll with the control key (2) to the desired entry and press the right-hand side of the control key (3) (OK) to open the context menu.

The first list entry will be displayed in each case.

Open the redial list.

1

Moving the cursor

Press the control key up/downwards in order to move the cursor to the left/right in the text. 8

Deleting/correcting characters

Press the left-hand side Dof the control key. The character to the left of the cursor is deleted. Press and hold down to delete the whole line.

de **en** fr nl

Menu tree

The functions of your telephone are offered to you via a menu that con-

sists of several levels.

▶ Press the (■) key to open the menu.

The menu functions and submenu functions where necessary are displayed in list form.

Scroll with the control key () to the required function and press the right-hand side of the control key () (OK) to confirm your

The complete menu tree of the display menu is depicted below.

→ page 13

→ page 14

→ page 13

→ page 14 → page 14

Call list (CLIP)	Entry 1-50	Call back	→ bage 7	CLASS	Call Diver
		Save number	→ page 10		CW Signa
		Delete entry	→ page 10		Withhold
		Delete all	→ page 10		Hotline
Phonebook	New entry		♦ page 9		Ringback
	Entry 1-100	Dial entry	→ page 7		Auto reca
		Edit entry	+ page 9		Phonema
		New entry	♦ page 9	Redial	Entry 1-5
		Delete entry	4 page 9		
		Delete all	→ page 9		
Security	Change PIN	→ page 12			
	Key lock	→ page 12			
	Emergency	→ page 13			
	Direct call	→ page 12			
	Call block	→ page 13			
Phone setup	Language	→ page 11			
	Date/time	→ page 11			
	Dial mode	→ page 14			
	Flash time	→ page 14			
	PABX code	→ page 14			
	Date/Time format	→ page 11			
	Area code	→ page 14			
	Speaker LED	→ page 11			
Audio setup	Ring tones	→ page 12			
	VIP Melody	→ page 12			
	Handset Volume	→ page 11			
	Speaker Volume	→ page 11			
	Headset Volume	→ page 11			
	Mute mode	→ page 11			
		•			

→ page 10 → page 10 → page 10

Delete entry Save number

Delete all

→ bage 7

Dial entry

→ page 14

→ page 14

2

9

Felephoning

n addition to the headset, you can also call using the speaker (handsree function) or using a headset connected to the telephone • page 8). In the following instructions, the symbol

Making a call

Entering the telephone number using the keypad Dial the number, lift the handset. \ !!!

Lift the handset, dial the number. Ţ

Press the hands-free/headset key to make a call via the speaker or the connected headset. Instead of lifting the handset: ৰ ় Press the hands-fr

You can switch over at any time during the conversation.

Making a correction when entering the telephone number (only possible if the handset is in its cradle):

Oress the control key Todelete a digit entered in e

Press the control key 🗂 to delete a digit entered in error to the left of the cursor. Dialling from the phonebook

If you have already saved numbers in your phonebook ($\, \star \,$ page 9), you can make a call directly from the phonebook. >
 Phonebook

Open the phonebook using the control key Searching an entry

Select an entry. Hold down the top or bottom of the key to scroll fast in the list.

Enter letters ($\stackrel{\bullet}{\to}$ page 5). The first entry starting with this letter is displayed. Lift the handset. Dialling numbers 5 🍱

Select Dial entry and initiate the call using the speaker with OK. Lift the handset to make the call using the hand-Open context menu. ъ ŏ 🗗

set.

Note You can also first lift the handset before opening the phonebook. Note

Dialling from the call list

The call list contains the numbers of the last 50 incoming and outgoing calls. **Prerequisite**: Telephone number display is possible for incoming calls (+ page 13).

Calls (+ page 13).

Calls from one number are only displayed once (with the information of the last call). If the number is saved to the phonebook or to the

direct dial keys, the accompanying name is also displayed if present.

► → Call list (CLIP)

Open the call list using the control key Dialling numbers

Select the number or name.

8

Lift the handset.

88

Select Call back and initiate the call using the speaker with OK. Lift the handset to make the call using the handset. Select the entry and open context menu with **OK**.

You can also first lift the handset before opening the call list.

Last number redial

The last five numbers dialled are automatically saved (each a max. of

Dialling the last saved number:

Press the redial key, lift the handset. **ا**

Lift the handset, press the redial key.

Dialling one of the last five numbers saved: ► Bedial

Press the redial key. Dialling numbers

Select the number. Lift the handset. 8

Select **Dial entry** and initiate the call using the speaker with **OK**. Lift the handset to make the call using the hand-Select the entry and open context menu with **OK**. 88

Note

You can also first lift the handset before opening the redial list.

Dialling a number using the direct dial key

32 digits). Use the shift key ✓ to access the key's second storage area. The display symbol f indicates that the shift function is active. Saving and editing the direct dial keys is possible using the context menu of the direct dial key (→ page 10), from the phonebook, the call list and the redial list. You can save 16 numbers to 8 direct dial keys and dial directly (two numbers can be allocated to each direct dial key, each with a max. of

Press the hands-free key to switch open listening on or off.

When the speaker is switched on and the handset is picked up, open listening is switched on. In this case, the hands-free microphone is

The persons present in the room can listen to the call over the speaker.

Jsing open listening / hands-free / headset

Switching open listening on/off During the call using the handset: When the speaker is switched on and the handset is in the cradle, hands-free via the hands-free microphone is switched on.

switched off.

Replace the handset whilst pressing the hands-free key.

Switching from open listening to hands-free:

·(

Lift the handset, press the direct dial key.

Press the direct dial key and open the context menu with Press the direct dial key, lift the handset.

Initiate the call over the speaker using the control key. Lift the handset to make the call using the handset.

š

You can also make a call via the microphone with the handset in the crade. The optimum distance from the microphone is approx. 50 cm.

Switching the hands-free on/off

Replace the handset whilst pressing the hands-free key.

Switching on hands-free during a call

·(

Lift the handset during the call. The call is transferred to

the handset.

Ending a call

Note nected.

Hands-free key, wait for dial tone.

Switching on hands-free prior to dialling الا الا Hands-free key, wait for dial t

Switching off hands-free

Ţ

Press the hands-free key during a call over the speaker.

Hands-free and open listening are not possible if a headset is con-

You can extend the selected number by entering additional digits.

Incoming calls

Incoming calls are indicated by the ring tone and in the display. For number fishlays (* * page 13), the number is visible in the display and the * symbol flashes. This symbol disappears when you answer the call or (if you don't answer) by calling up the call list. The name is also displayed for callers who are stored in the phonebook or on the direct dial keys with names.

If the VIP mode is activated (→ page 12) the VIP symbol flashes in the display and the VIP ring tone signals the call.

Accepting a call

Lift the handset.

Press the hands-free key to accept the call via the speaker or the connected headset.

Call waiting

The display shows the caller number if the number can be displayed, and the name if it is saved in the phonebook (\rightarrow page 9). If the Call waiting service is activated (→ page 13), an incoming call is indicated during the conversation. You will hear the call waiting tone.

Press the callback key again to end the second call and to return to the first call. Press the callback key to accept the call. The first caller Œ

Once the headset is connected this is indicated in the display with the Ω above). symbol.

key in the same way as for the hands-free function (see

neadset whilst the handset is in the cradle. (Please note the safety prerautions → page 4.)
The telephone is operated using the hands-free/headset

Connect the headset (→ page 2). You can then make a call over the

Using a headset

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de **en** fr nl

∞

Settings during a call The settings described below can also be changed in the menu page 11).	↓ Pho ching a	Either: save in phonebook OK Confirm the saving of the entry in the Phonebook. (**) Select VIP mode (On / Off) and confirm with OK.	or '= → Redial Transferring a number from the redial list to the phonebook/
Setting the handset/headset volume There are three adjustable levels.	Select an entry. Hold down the the top or botton of the key to scroll fast in the list. or	Or: save to a direct dial key (direct dial key
Setting the speaker There are seven adjustable levels. There are seven adjustable levels. $\frac{1}{N} \left[- \right] \text{ Set the volume using the volume up/volume down keys.}$ The set volume is chown in the dischaw.	Enter letters (+ page 5). The first entry starting with this letter is displayed. Note	Call list (CLIP) Incoming calls are saved if the number is transmitted (+ page 13). Up to 50 calls are saved. If the number is the same, only the last incoming call	Select. Save furniser and confirm with OK. Find the editing number of confirm with OK. While editing numbers and names, move the cursor with (**)
Mute You can switch off the microphone or handset and microphone during a		is saved. The olders call is automatically detected when there are more than 50 calls. If the caller is saved to the phonebook or to a direct dial key, the name is also displayed.	Either: save in phonebook OK Confirm the saving of the entry in the Phonebook. Fall calert VID made (for 1 and 1 confirm with OK)
call according to the setting of the mute function (* page 11): ## Press the mute key to switch the mute function on or off, buring muting of the microchone a malchdy can be alwaed.	Saving funitibers Saving funitibers The first list entry is displayed.	You can use the call first for the following functions: • Calling a number from the list (*) page 7). • Saving in the phonebook or to direct dial keys.	ave to a
or any invarige of the materials as increased can be payed. Once the telephone has been muted this is indicated in the display with the # symbol.	Select New entry and confirm with OK. Enter number and confirm with OK.	Opening the call list Open the call list in standby mode:	Press the desired direct dial key. The notice Saved appears in the display.
Ringer and volume While the talenhone is finding the volume and ringer can be changed	Enter name and confirm with OK. The notice Save to PBK? appears in the display.	Press the control key	Delete entry / Delete all
Set the ringer volume with the volume up/volume down	er: save i	T → Call list (CLIP)	
keys (5 levels, 0=silent). Select ringer with the numeric keys (10 different).	OK Confirm the saving of the entry in the Phonebook. Select VIP mode (On / Off) and confirm with OK.	Transferring a number from the call list to the phonebook/direct dial key	 Select Delete entry or Delete all and confirm with OK. OK Press the control key to confirm the action.
Using the phonebook, lists and direct	Or: save to a direct dial key \(\subseteq \text{		Direct dial keys You can save up to 16 numbers on direct dial keys. Use the shift key
		Select Save number and confirm with OK. Change number (if required) and confirm with OK.	to access the keys' second storage area. The display symbol J indicates that the shift function is active. You can also seein direct dial love from the antries in the phonehook.
The following options are available on your telephone for storing your contacts and the incoming and outgoing calls: • Phonebonk with may 100 entries	Note You can save an entry first in the phonebook and then to a direct dial key.	Enter name and confirm with OK . While editing numbers and names, move the cursor with []. Either: save in phonebook	You can also assign utfect darkeys from the entities in the phonebook, in the call list or the redial list. They can also be used as function keys e. g. for call forwarding. Your network provider will notify you of the services and functions offered and
Call list with the last 50 incoming calls Redial list with the last 5 outgoing calls	Delete entry / Delete all ☐ ➡ → Phonebook	OK Confirm the saving of the entry in the Phonebook. Select VIP mode (On / Off) and confirm with OK.	the corresponding key sequences. The menu offers you additional optional settings for network services $ + $ page 13).
 Driect dail keys for storing 10 important numbers Phonebook Phone digaset DATA Contains a phonebook in which you can save up to 100 entries, each with a maximum of 32 digits for numbers and 16 charaction. 	The first list entrity is displayed. Select the entry and open context menu with OK. Select Delete entry or Delete all and confirm with OK.	Or: save to a direct dial key \(\tau \) \(\tau \) Press the desired direct dial key. Then notice 5 saved annears in the disclaw.	Saving numbers / Changing the allocation _ _ Press the direct dial key that you want to assign with a number or the entry of which is to be changed.
acters for names. You can make a call (\star page 7), add new entries and manage and modify entries (\star page 9) using the Phonebook.	OK Press the control key to confirm the action. Edit entry	Delete entry/Delete all	Empty is displayed when the storage is empty, If the keys is already assigned you will see the saved number. For long further than 16 digits) switch between the display of digits 1-16 numbers (more than 16 digits) switch between the display of digits 1-16
you can a manually get runn bers and names or transfer them from the call list (★ page 10). You can also allocate VIP mode to a number.	The first lite parts it is direct und		and 17-32 with the control key 🐑 New entro
The III symbol is indicated in the display if the phonebook has been opened. In addition, the phonebook entry number is also indicated in	121 121	Select Delete entry or Delete all and confirm with OK. OK Press the control key to confirm the action.	OK Switch to the entry mode with OK.
the display (01 99, 00 is displayed for 100). Opening phonebook In standtus	Select Edit entry and confirm with OK. Change number and confirm with OK. Change name and confirm with OK.	Last number redial list The last five numbers called are automatically saved (each a max. of 32 digits).	Changing the allocation OK Open the context menu with OK. Select Edit entry and confirm with OK.
Press the control key 🗐	The notice Save to PBK? appears in the display.	Opening the last number redial list P Press the redial key.	Enter/change number and name Time Enter/change number and confirm with OK. Time Enter/change name and confirm with OK.
	6	10	de en fr nl

ĕ	Press the control key to save the entry.	Set LED on hands-free/headset key
Delete ent	Delete entry / Delete all	You can specify when the telephone LED should flash (key 💌 ा).
\ \ 		→ Phone setup → Speaker LED
	Press the direct dial key the entry of which is to be deleted and open the context menu with OK .	Select the required setting and confirm with OK .
8	Select Delete entry or Delete all and confirm with OK.	Incoming Call: for a new call
ŏ	Press the control key to confirm the action.	Messages: if a network provider signals the receipt of a new message
. T. V		
Adjustii	Adjusting the telephone	OK Press the control key to save the setting.
		Setting the volume
Basic settings	iings	Handset, speaker and headset volumes can be set using the menu or
All basic sett The current s	All basic settings are accessed via the menu Phone setup. The current setting is marked at the right-hand side of the display with *.	changed during the telephone conversation directly using the keys (→ page 9).
Changing	Changing the Language	Setting the handset volume
You can change the lang guages to choose from.	You can change the language for display messages. There are four languages to choose from.	I here are three adjustable levels.
Ph	→ Phone setup → Language	Select the required volume and confirm with OK .
8	Select the desired language and confirm with OK.	The current value will be displayed.
ŏ	Press the control key to save the setting.	Setting the speaker volume
Setting the	Setting the date and time	
You can adju	You can adjust the current date manually and change the format of the	🍱 → Audio setup → Speaker Volume
display. The displays. You	display. The time is updated by incoming calls with telephone number displays. You can adjust this setting if necessary (→ page 15).	Select the required volume and confirm with OK . The current value will be displayed.
Ph	Phone setup → Date/time	Setting the headset volume
E	Enter 6-digit date (DDMMYY).	There are three adjustable levels.
	Enter 4-digit time (HHMM).	► → Audio setup → Headset Volume
OK The entry is	OK Confirm entry. The entry is always performed as described here, regardless of the set	Select the required volume and confirm with OK .
date and time format.	ne format.	ine current value will be displayed.
Setting the	Setting the Date/Time format	Mute
†	Phone setup → Date/Time format	
B	Select the date format and confirm with OK . For example:	4 And
	31. December 2011 YYYY MM. DD: 2011 12 31	Select the required function and confirm with OK. Micro off: You are still able to hear the person on the end of the
	DD.MM.YYYY: 31.12.2011	line, however they cannot hear you.
	DD/MM/YYYY: 31/12/2011 MM/DD/YYYY: 12/31/2011	Micro&Spk. Off: The telephone is completely muted. Mute melody: Whilst in mute mode, the person on the
ŏ	Press the control key to save the setting.	other end of the line hears a melody.
3	Select the required time format and confirm with OK . 24 h: 24 hour format 17h: 12 hour format	ne set function can also be activated using the mute key during a call (→ page 9).
ŏ	Press the control key to save the setting.	

Setting the ringer

The following optional settings are available:

Melody and volume

- Switching the ringer off

Setting the melody and volume

You can secure your telephone against unauthorised access by using the Key lock and PIN entry, in order to lift the Key lock.

-ocking the telephone

Security settings

Note

If you have changed the telephone PIN you must always enter the PIN after selecting the Security menu entry.

You can set the ringer melody and volume or switch the ringer off. There are 16 different ringers available, the volume can be adjusted in 5 levels

When the Key lock is activated, all keys are locked, with the exception of the menu key, the hands-free/headset key and the stored emergency numbers (\star page 12).

You can also change both settings directly using the keys while the tele-

Note

The melodies are played over the speaker if you lift the handset before commencing with the settings.

Setting up the Key lock

the lock key.

or ►■ → Security → Key lock

Hold down the lock key. Deactivating the Key lock

Note
The ring melody can only sound when the ring pulse arrives from the telephone network. The length of the pulses may differ from network to network it is therefore possible that some inig tones for selection may seem clipped. If this is the case please select another ring tone.

/ + Set the volume for the ringer currently selected.
 OK Press the control key to save the setting.

Press the control key to save the setting.

© ...[9:sve] Ringer 1 - 10

Select the required ringer.

Audio setup → Ring tones
Select the required ringer

phone is ringing (→ page 9).

Press the menu key. O ...(9....) If the preser PIN (0000) is changed, enter a 4-digit PIN and confirm with **OK**.

symbol in the display. Changing the PIN

The default setting for the **Key lock** PIN is 0000. You should change this PIN in the interests of security.

number from the phonebook VIP mode status (\star page 9), an incoming call from this number will be indicated with the VIP Melody ring tone.

You can set one of the ringers as VIP Melody. When you have assigned a

Setting the VIP Melody

► Security → Change PIN

0 ... (9 **** Enter a new 4-digit PIN and confirm with **OK**. Repeat the entry and re-confirm with OK.

Emergency calls

There are two emergency numbers stored in the telephone as a default setting 110, 112. You can an additional emergency number with a max of 28 digits. These emergency numbers can also be selected when the Key Jock is switched on.

Selecting an emergency number

If you do not wish to be disturbed, you can switch off the telephone's ringer. The three following optional settings are available: All tones off, VIP only (see above), All tones on (default setting).

(=) /(+) Set the volume for the ringer currently selected.OK Press the control key to save the setting.

Setting the ring mode

(a) ... (b) ... (c) ... (c) ... (d) ... (d) ... (e) ..

Select the required ringer.

Audio setup → VIP Melody

Select the required ringer

Select the required setting and confirm with OK.

→ ■ → Audio setup → Ringer mode

Press the control key to save the setting.

Enter the emergency number. Emergency is indicated in the display. Saving the emergency number

► Security → Emergency

10 ... [9....] Enter the emergency number and confirm with **OK**. Switching the ring tone off using the key $\frac{1}{2}$ — Hold the key down to switch the ring tone off and back on. Once the ring tone has been switched off, this is indicated with the β symbol in the display. de **en** fr nl

12

1

Direct calling (baby call)

ed by pressing	- and
☴	Š.
number	the key
activated, the stored number is dia	(except for the keys
s activated, t	the handset (excer
t calling is a	/ after lifting t
n direct o	key after

► Security → Direct call Activating direct calling

Direct calling is activated and indicated in the display.

Deactivating direct calling

Press the menu key.

Confirm the question Disable direct? with OK. Enter PIN (→ page 12) and confirm with OK

Blocking numbers

You can block phone calls to specific dialling codes (e.g. chargeable codes), max. 3 numbers with 5 digits each.

Select Call block 1 (or 2 or 3) and confirm with OK. Select Call block 1 (or 2 or 3) and confirm with OK

O --[9---] Enter the call block number and confirm with OK. 🏣 🕁 Security 🕹 Call block

Network services

range of helpful, additional services (e. g. call forwarding, callback when busy, withheld number, etc.). These services can be selected using speculate key combinations, of which your network provider will notify you. These services can be called up and activated in the telephone menu and the corresponding key combinations can be set or adjusted if nec-Public telephone systems offer you – in some cases only on request – a

A list with predefined key combinations can be found in the Appendix (→ page 15).

Caller display (CLIP)

If the caller display service is activated, an incoming call is shown on the display with the number and saved in the call list. **Prerequisite**: The network provider supports the following service features and the number transfer is not withheld by the caller:

- CLI (Calling Line Identification): The caller's number is transferred. CLIP (Calling Line Identification Presentation): The caller's number is
- You can transfer this number to the phonebook and edit it (\rightarrow page 9). If you have saved the local dialling code (\rightarrow page 14), a call with the same dialling code is automatically only displayed as the number with-

out the dialling code.

- Call forwarding can be set up for three different conditions:
 - All Calls (CFU, Call Forwarding Unconditional) No Answer (CFNR, Call Forwarding No Reply) When Busy (CFB, Call Forwarding Busy)

de **en** fr nl

13

Activating call forwarding

Select Activate and confirm with **OK**. The conditions for call forwarding are displayed. ► CLASS → Call Divert

Select the desired condition for call forwarding and confirm with OK.

The key combination for initiating call forwarding is dis-

Enter target number and confirm with **OK**.

The call forwarding for the selected condition is activated.

Deactivating call forwarding

LESS → CLASS → Call Divert

The conditions for call forwarding are displayed. Select Deactivate and confirm with OK.

Select the desired condition for call forwarding and con-

The call forwarding for the selected condition is deactivated. Other network services

Setting/changing key combinations for network services

If the preset key combinations for the network services do not match the information from your network provider, or if no key combination is preset, these can be changed or set in the corresponding network services

Your network provider will notify you of the key combinations - if

► CLASS

Select required network service.

Press recall key. 6

Enter key combination for the network service and confirm with **OK**.

Use the shift key 🖅 in the position, in which you are required to enter

For example: Call forwarding when busy is activated with *67* cnumber> #. You are required to enter data for cnumber>. Use the following key combination for the configuration:

CW Signal

If the call waiting function is available from your network provider and you activate it, an incoming call is signalled during the conversation (→ page 8).

► CLASS → CW Signal

Select Activate / Deactivate and confirm with **OK**. The signalling of an incoming call by call waiting is activated or deactivated.

Operation on a PABX

f you activate the withheld number function, your number will not be

displayed for the person you are calling.

•■ → CLASS → Withhold number

Withheld number (anonymous call)

Select Activate / Deactivate and confirm with OK.

Select Activate / Deactivate and confirm with The withheld number function is activated or deactivated.

Special functions/Recall key

During an external call, you can make an enquiry or forward the call. To do this, press the recall key [R]. The subsequent procedure depends on your PABX. To set the recall key, the telephone's flash time must be set consistently with your PABX. Please refer to the operating instructions for your PABX.

Changing the dialling mode/flash time

The network service Hotline allows a special number to be selected simply by lifting the handset, e.g. in case of emergency. This function is not available if it is not supported by your network provider.

The telephone supports the following dialling modes: Tone dialling

Pulse dialling

Enter target number and confirm with OK.

Select Activate and confirm with OK.

► CLASS → Hotline

Select Deactivate and confirm with OK.
The network service Hotline is activated or deactivated.

Call back when busy

Depending on your PABX, you may need to change your phone's dialling mode or the flash time. (Default setting: Tone dialling)

► → Phone setup → Dial mode Changing the dialling mode

Select Tone or Pulse and confirm with OK.

If you make a call but the person you are calling is already on the line, you can arrange an automatic callback as soon as the connection is free.

Select Activate / Deactivate and confirm with OK.

Select Activate / Deactiva

The callback is activated or deactivated.

Automatic callback

► CLASS → Ringback

You can change the flash time if the dialling mode is set to tone dialling (see above) (default setting 90 ms). Changing the flash time

► → Phone setup → Flash time

Select flash time, the following values are available: 90, 120, 270, 375, 600 ms and confirm with **OK**.

Temporary switch to tone dialling in "Pulse" position

this function, the number of the caller who called you last while your line was busy will be dialled automatically. fyour network provider supports automatic callback and you activate

Select Activate / Deactivate and confirm with OK.

Select Activate / Deactivate and con The automatic callback is activated or deactivated.

Network mailbox

► CLASS → Auto recall

answering machine), you can set the telephone to tone dialling for the To use functions that require tone dialling (e.g. remote control of duration of the call.

After the connection is established:

* Press the star key.
After the connection is terminated, the setting reverts to pulse dialling.

If you use a network mailbox, the telephone display shows new incoming messages. The request for the mailbox can be started via the menu. You can also set the LED so that it lights up to indicate incoming mes-

If your phone is connected to a PABX, you may have to use a dialling pre-fix to make external calls. You can save up to three prefixes in your tele-Setting dialling prefixes

If a dialling prefix is recognised during dialling, a dialling pause will automatically be applied. You can adjust the time of the dialling pause if necessary (→ page 15).

▶ → Phone setup

in public telephone systems the recall key (flash) is required for the use of various additional services; e.g. for "callback when busy". If necessary, the flash time on your telephone must be adjusted to the

The network service number is selected.

Recall key

TE → CLASS → Phonemail

sages (+ page 11).

The current setting is displayed.

0 [9] Enter the dialling prefix (1 to 3 digits) and confirm with **OK**. Use the control key to switch between the dialling prefixes: 8

Press the control key up/downwards in order to jump left/

► ■ → Phone setup → Area code

then used to display only the number of the person calling without area code in the call lists, provided that they have the same area code.

fnecessary you can change the default setting for the area code for your connection which is saved in the telephone. The number saved here is

requirements of the telephone system (→ page 14).

Setting up of dialling code

Enter the new area code (max. 6 digits) and confirm with The default number is displayed. 09

Š

Appendix

Other settings

In addition to the options described in section Adjusti phone (*) page 11), you can perform further settings Fature codes (e.g. reset the factory settings). These as the following table.
Default settings are displayed in **bold**. If there is no bo ting, the presetting is country-dependent.

15: Press the direct dial

or the activation/deac

Key combination

*21°<number>#

Enter the key sequence for the Featur with **OK**. Press the menu key and 1.

Key		Value	Description
sedneuce	ance.		
02	0	18	Sets the length of the pause that
	_	3.8	can be inserted with the pause key
	2	6.5	
25	0	Restores all settings to the factory settings	e factory settings
	_	Deletes the contents of al	Deletes the contents of all programmable direct dial keys
	7	Deletes all dialling codes	
	m	Restores all network service codes	ce codes
	4	Deletes all stored numbers (not phonebook)	s (not phonebook)
20	0	Time display off	Sets the time format
	_	24 hour format	
	2	12 hour format	
57	0	Deactivate	Interval for the LED (hands-free/
	_	500 ms/500 ms	headset key) on incoming call
	7	30 ms/70 ms	
20	0	Deactivate	Eliminates click tone at the begin-
	_	Activate	ning or end of ringer.
			If network transmits only a very
			short ringer impuls, the click tone
			can be prevented by deactivating
			(setting 0) the setting.
83	0	Deactivate	Withholds the first ringer. With this
	_	Activate	setting, it is possible to determine,
	7	Automatic	if an incoming call comes from a VIP
			number, and to select a VIP ring
			tone instead of the normal ring

justing the tele-				default setti "AC".
ings with the help of ese are summarised in	9#	6~E	Seconds Settings 3-15 Default setting: 8	Time for terr ming call Settings 10-
o bold value for a set-				keys 1-5.
	Pred	lefine	Predefined key combinations	s
ure code and confirm	The fo tivatio	ollowing on of ne	The following key combinations are predefined for tivation of network services.	redefined fo
	Function	noit		
	Activ	ate call f	Activate call forwarding for all calls (CFU)	(
gth of the pause that	Deac	tivate ca	Deactivate call forwarding for all calls	
ted with the pause key	Activ	ate call f	Activate call forwarding when no reply (CFNR)	CFNR)
tings	Deac	tivate ca	Deactivate call forwarding when no reply	^
able direct dial keys	Activ	ate call f	Activate call forwarding when busy (CFB)	
	Deac	tivate ca	Deactivate call forwarding when busy	
sbook)	Activ	ate signa	Activate signalling for call waiting (CW)	
e format	Deac	tivate sig	Deactivate signalling for call waiting	
	Activ	Activate hotline	Je.	

This will be replaced by the phone number, which has been specified for the corresponding function in the menu CLASS (→ page 13). <number>

Deactivate hotline

Standard characters

Start signal for CLIP recognition. If CLIP does not function with the

Ringer AC Value

Key sequence #4 0 ۰-

Description

ing 0, select setting mination of an inco-

The following characters can be entered using the keypad:

14×										1		
13×										٠.		
12x												
11x 12x 13x										t		
30 X		U.				ō			°ra			
š		ē				ô			0	.7		
š		ď	ŵ	-		ó		û	88		%	
×	n	'n	'n	-		ó		æ	ý		٨	
ĕ	*	à	é	-		ř	8	ù	ÿ	+	П	
SX	<>	:ra	:0	:		ö	7	:D	6	0	٧	S
4×	3	2	3	4	2	9	s	8	z		^	ø
ă	æ	٥	+	-	-	0	_	>	У	-)	-
7X	-	q	a	۲	¥	u	ь	n	×		/	@
ž	٦]	в	v	6	j	Ε	۵	ţ	۸		*	#
Key	[-]	2 **	3 %	4 %	5	6 mm	7 198	8 m	2.xxs 6	0	*	#
				_	_							_

67<number>#

61<number>#

53<number>#

*) Space The mode for text entry is changed by repeatedly pressing the shift key The mode between capitals (ABC), lower case (abc) and digits (123). The mode is briefly indicated on the lower right of the display when switching. By selecting the (123) mode, you can enter digits directly by pressing the respective key once.

Wipe the unit with a damp cloth or an antistatic cloth. Do not use solvents or microfibre cloths. Never use a dry cloth; this can cause static.

Contact with liquid 🖺

If the device comes into contact with liquid:

Allow the liquid to drain from the device.

- When it has fully dried out, you will normally be able to use it again. In rare cases, contact with chemical substances can cause changes to the Pat all parts dry. Place the device (the keypad facing down) in a dry, warm place for at least 72 hours (not in a microwave, oven etc.).
 Do not switch on the device again until it is completely dry. telephone's exterior. Due to the wide variety of chemical products available on the market, it was not possible to test all substances.

Questions and answers

Lift the handset for 5 seconds, put it back in the cradle and lift it again. Is the connecting lead correctly plugged into the telephone and the You lift the handset but you hear no dialling tone: You lift the handset, but don't hear the free tone: telephone socket?

The dialling tone is audible but the telephone will not dial: The connection is OK. Is the dialling mode set correctly?

The other party cannot hear you:

Mute activated?

Updates time automatically according to CLIP information

Activate Deactivate

#2

de **en** fr nl

85/85 ms 85/110 ms 85/140 ms 110/110 ms 70/70 ms

Sets the pulse ratio for pulse dial-ling Sets signal time/signal pause for tone dialling

Sets the contrast of the display Setting 10: Direct dial key 1 Setting 11: Direct dial key 2

1 = minimum contrast
5 = default setting
11 = maximum contrast

Settings 1-11

6~[

Regular pulse noises can be heard during a call:
The connection is receiving metering pulses from the exchange, which
the telephone cannot interpret. Contact your network operator.

Protecting our environment

bility. That is why we actively take steps to create a better world. In all areas of our business - from product planning and production to sales and waste of disposal - following our environmental conscience in eve-We at Gigaset Communications GmbH are aware of our social responsi rything we do is of utmost importance to us. Learn more about our earth-friendly products and processes online at www.gigaset.com. Our environmental statement

Environmental management system

Gigaset Communications GmbH is certified according to the international standards ISO 14001 and ISO 9001. ISO 14001 (Environment): Certified since September 2007 by TüV SÜD Management Service GmbH. ISO 9001 (Quality): certified since 17/02/1994 by TüV SÜD Management Service GmbH.

If you should encounter any problems when operating the telephone connected to a communication system with analogue network access, please contact the network operator responsible or your dealer. Contacts

Disposal

All electrical and electronic products should be disposed of sep-arately from the municipal waste stream via designated collec-ton facilities appointed by the government or the local authori-

This crossed-out wheeled bin symbol on the product means the product and separate collection of your old appliance will help prevent potential negative consequences for the environment and human health. It is a precondition for reuse and recycling of used electrical and electronic is covered by the European Directive 2002/96/E. The correct disposal

For more detailed information about disposal of your old appliance, please contact your local council refuse centre or the original supplier of the product. equipment.

Authorisation

This device is intended for analogue phone lines in the UK. Country-specific requirements have been taken into consideration. We, Gigaset Communications GmbH, declare that this device meets the essential requirements and other relevant regulations laid down in A copy of the 1999/5/EC Declaration of Conformity is available at this Directive 1999/5/EC.

www.gigaset.com/docs nternet address:

0682

de **en** fr nl

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*	VIP melody
Key combinations, predefined	W © Gig Withheld number14 Subject All right
1	MANAM
Last number redial 77	

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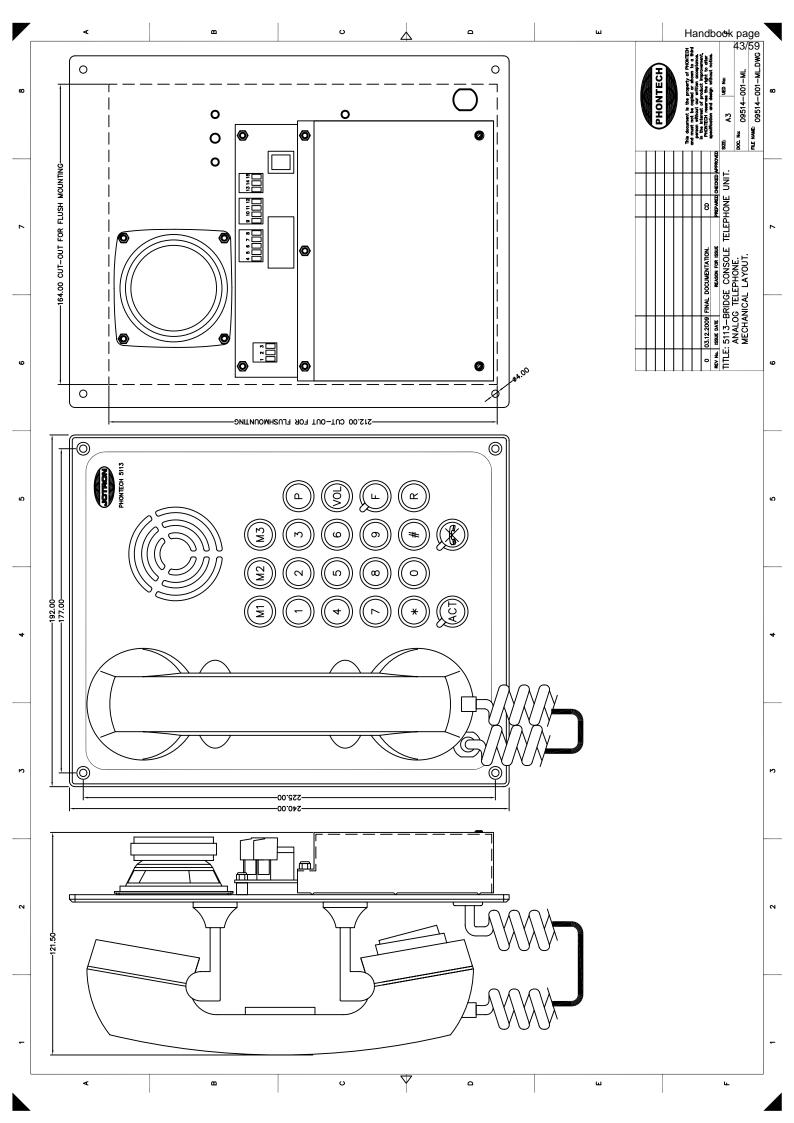
w.gigaset.com

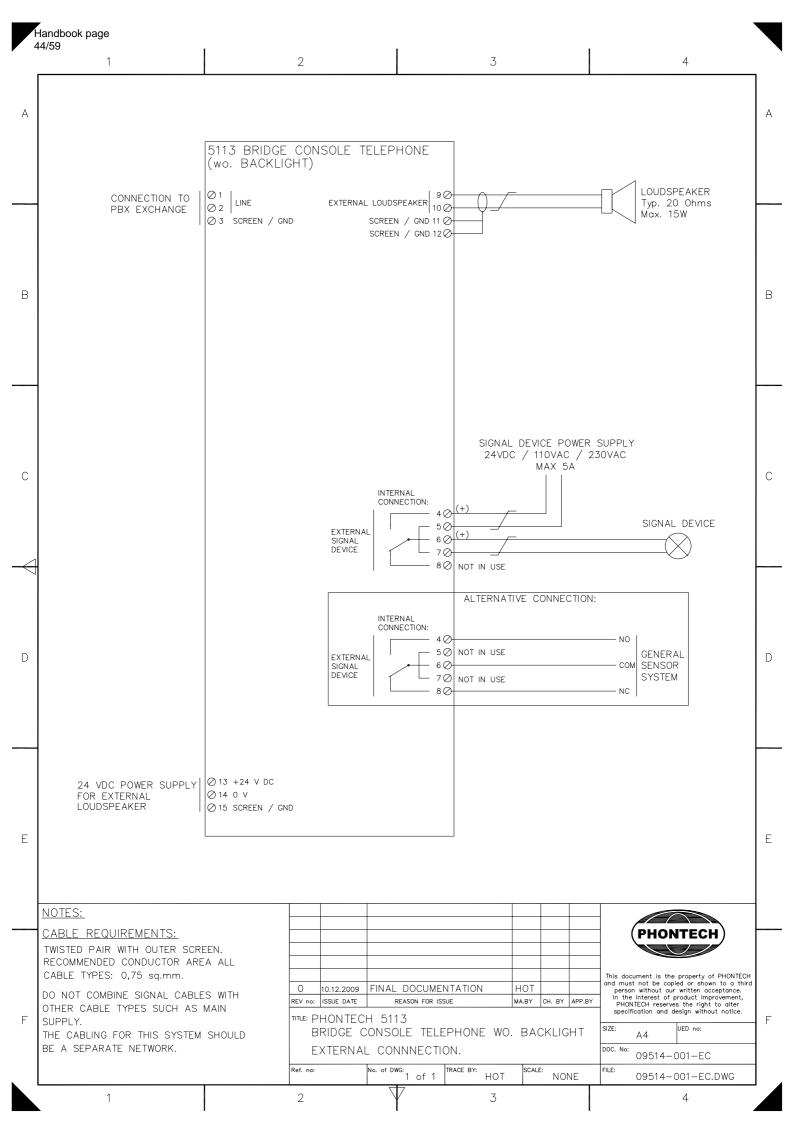
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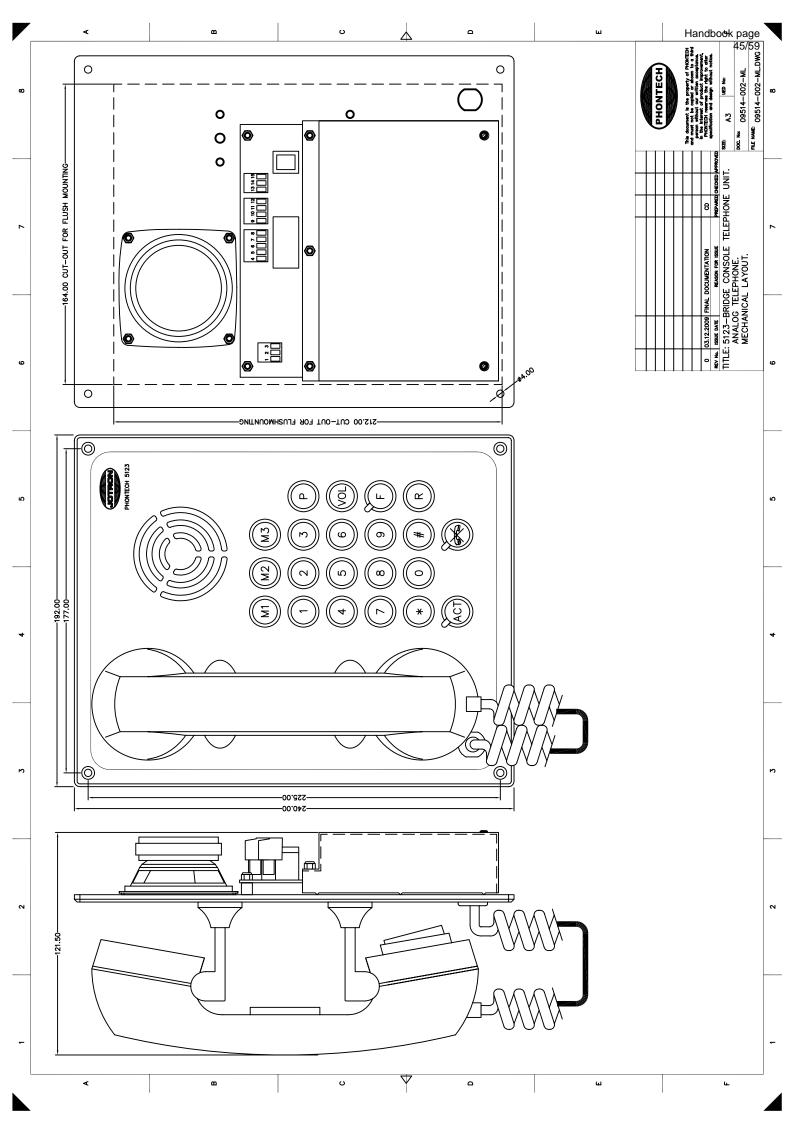
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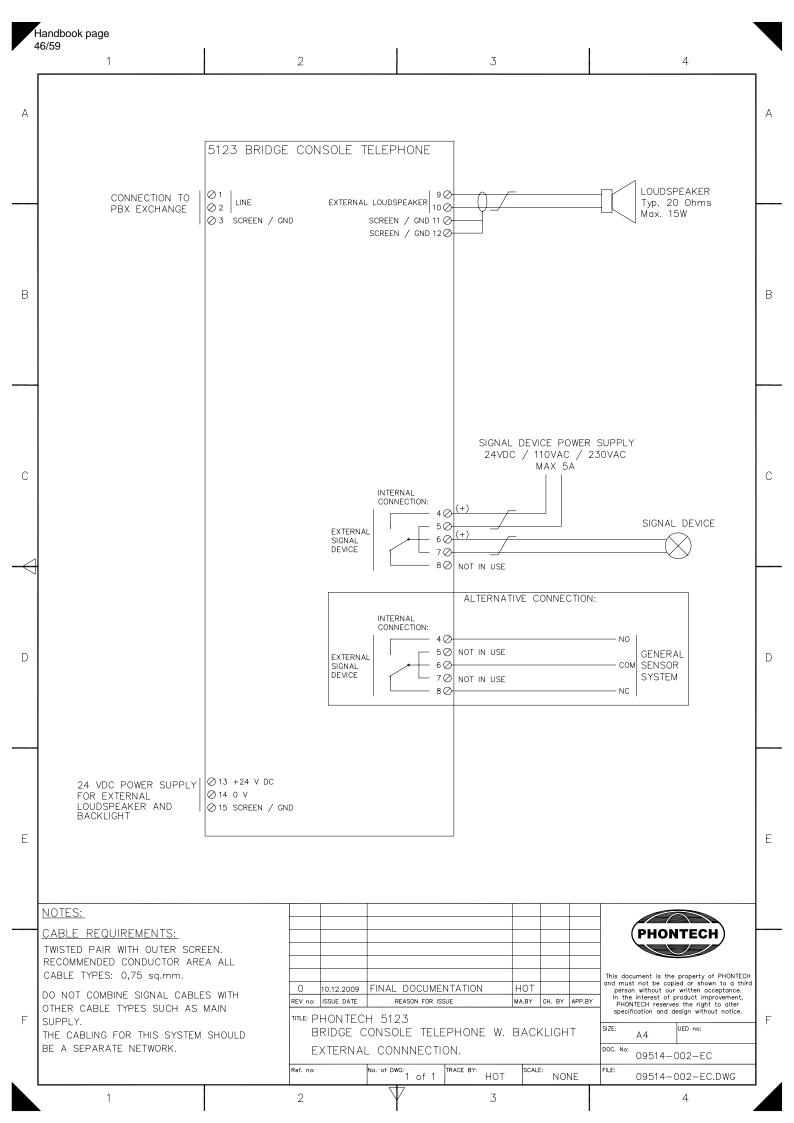
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Weatherproof Telephone FernTel 3

Analogue desk/wall telephone for indoor and outdoor use

- ► Shock-resistant housing (Polycarbonat)
- ▶ Protection degree IP 65 acc. to IEC60529
- Option Handset can be fixed (stabiliser bracket)
- ► Ambient temperature -25°C to +55°C
- ➤ Call tone ≥ 95 dB(A), 1 m
- Direct key at Diplayversion
- Receiver volume can be boosted
- ► Telefonbook at Displayversion
- ► PIN code
- ► 7B-Version with call tone unit
- Assembly-friendly
- Menu 4 languages

Use as workshop telephone

The FernTel 3 is the ideal telephone for many different work areas.

Application

The FernTel 3 telephone of FHF is as stylish for indoor use as it is resistant for outdoor use in areas without explosive atmospheres. The new FernTel 3 is the ideal telephone for many different work areas – seawater, high air humidity, dust and not exactly gentle mechanical stress are no problem for it.

Its modern design is the perfect "packing" for the finest high-tech.

Its striking signal colours ensure the FernTel 3 cannot be missed whenever a telephone is urgently needed, e.g. in emergencies in poor weather and light conditions.

The FernTel 3 is suitable for almost universal use thanks to its amazing transformability.

A deft hand movement and the desk telephone for indoor use is converted into a wall telephone for outside use. The FernTel 3 is offered in different variations. With 16 buttons without Display or with 21 buttons with Display. Both variations are offered with spiral cord or steel armoured cord. As addition, the FernTel 3 with Display has a direct key. Of course, there is the new FernTel 3 also as Hotline phone (ZB-Variant), with call tone unit.

In the casing-inside there is a keyboard for deposited that Hotline number. One becomes as accessories a stabiliser bracket for applications on movable machines, like for example on Ships.



Technical specification

Connection data

Supply voltage 24...66 Vdc
Supply current 19... 100 mA
Ringing alternating current 30...90 Vac
Ringing frequency 16.. 68 Hz

Enquiry key (flash) only for DTMF 80 ms, 120 ms, 600 ms

Dialling procedure

DTMF Frequencies according to ITU-T Q.23
Tone duration unlimited or 90 ms

PD Pulse/Pause ratio

1.5:1 (60/40 ms) or 2:1 (66.7/33.3 ms)
Ringing volume Approx. 95 dB(A) at 1 m distance

Housing

Housing material Polycarbonat
Height x Width x Depth 293 x 191 x 128 mm
Weight Approx. 2.3 kg

Display 2 lines, 16 positions per row, 7x 5 matrix

Pictograms

Menu 4 languages

Operating utilization position Table or vertical wall mounting

Receiver

Mouthpiece Electret-foil microphone
Receiver inset dynamic receiver inset with magnetic field generator

Connections Single- or multiwired up to 2.5 mm²

Labelling

Power cable TCP/La, TCP/Lb

Secondary sounder W / W1

Environmental conditions Phone LCD Unit
Ambient operating temp. -25°C...+55°C -10°C...+50°C

Degree of protection acc. to IEC60529 IP 65

Order information

Cable glands:

1x M20 cable gland 1x M20 blind plug 2x M12 blind plug



yellow red	
grey black	

~191/~7.5*

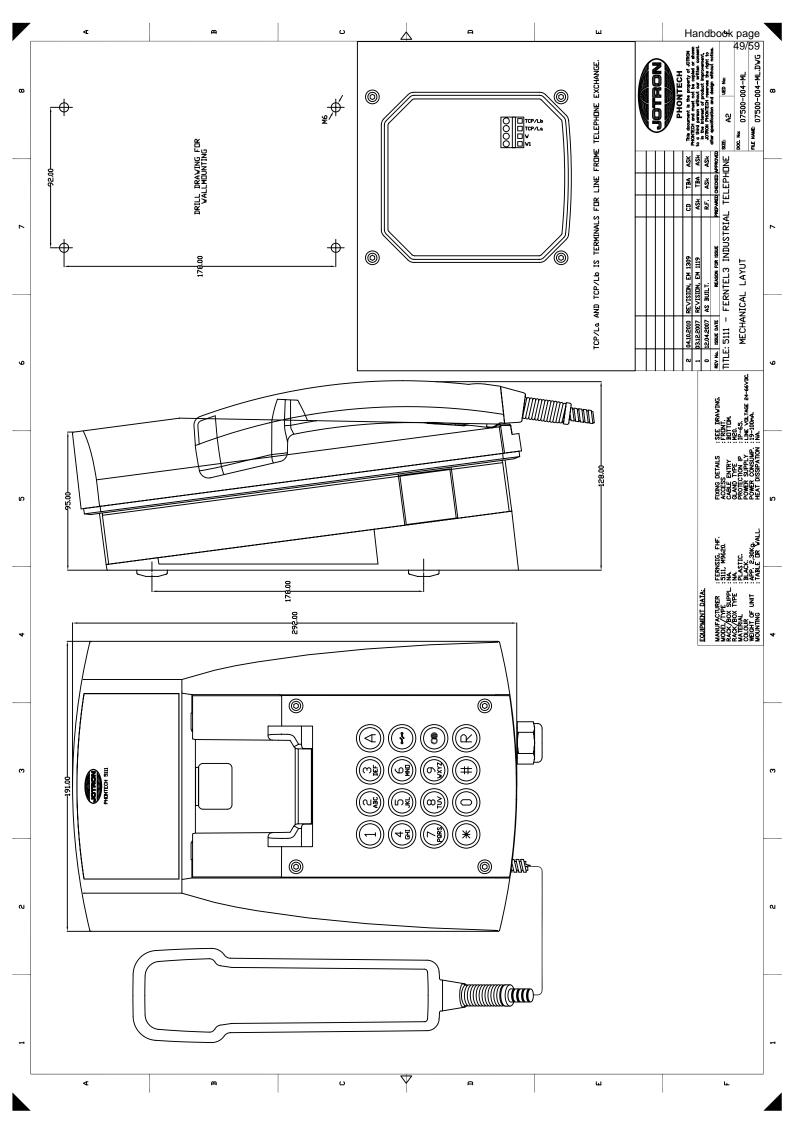
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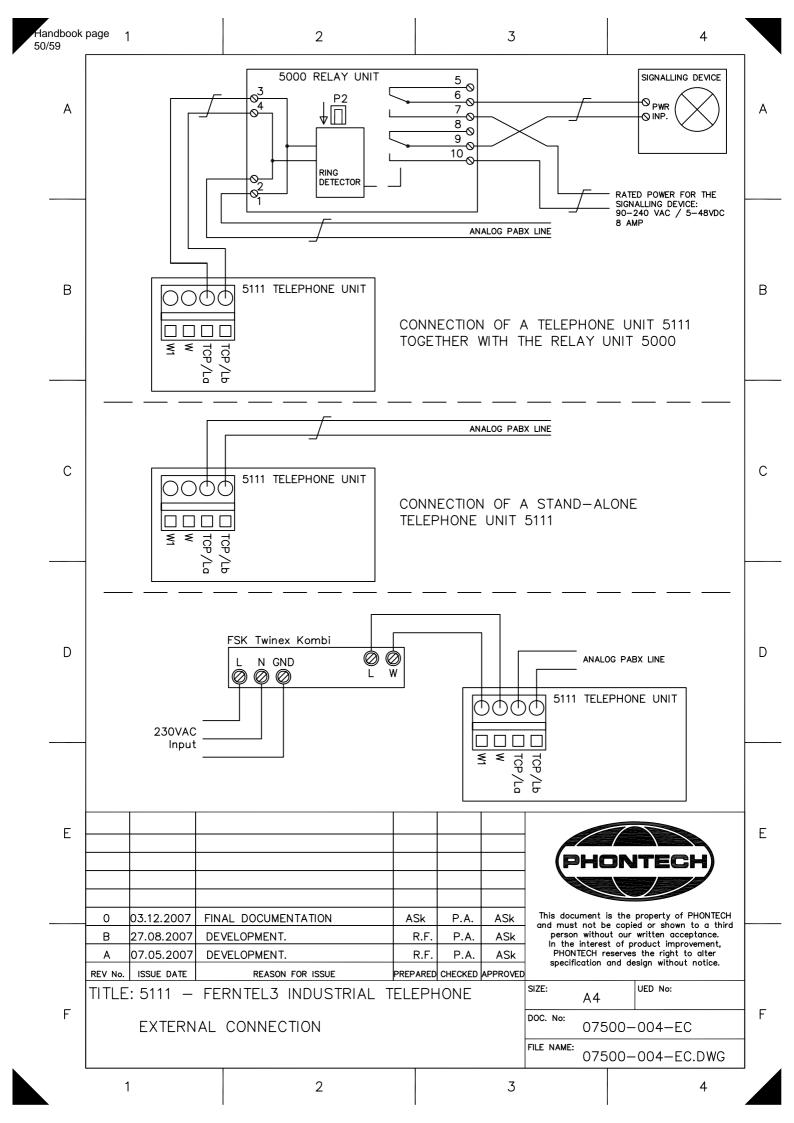
1
2
7
0

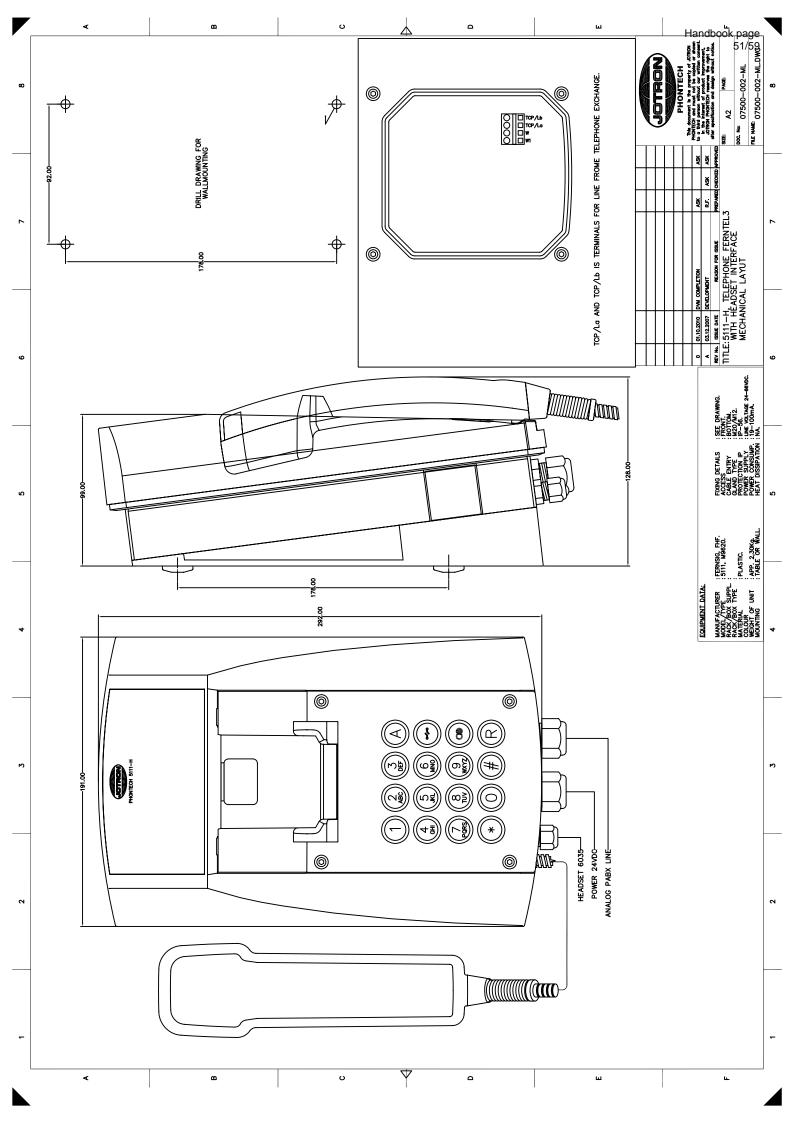
Stabiliser bracket

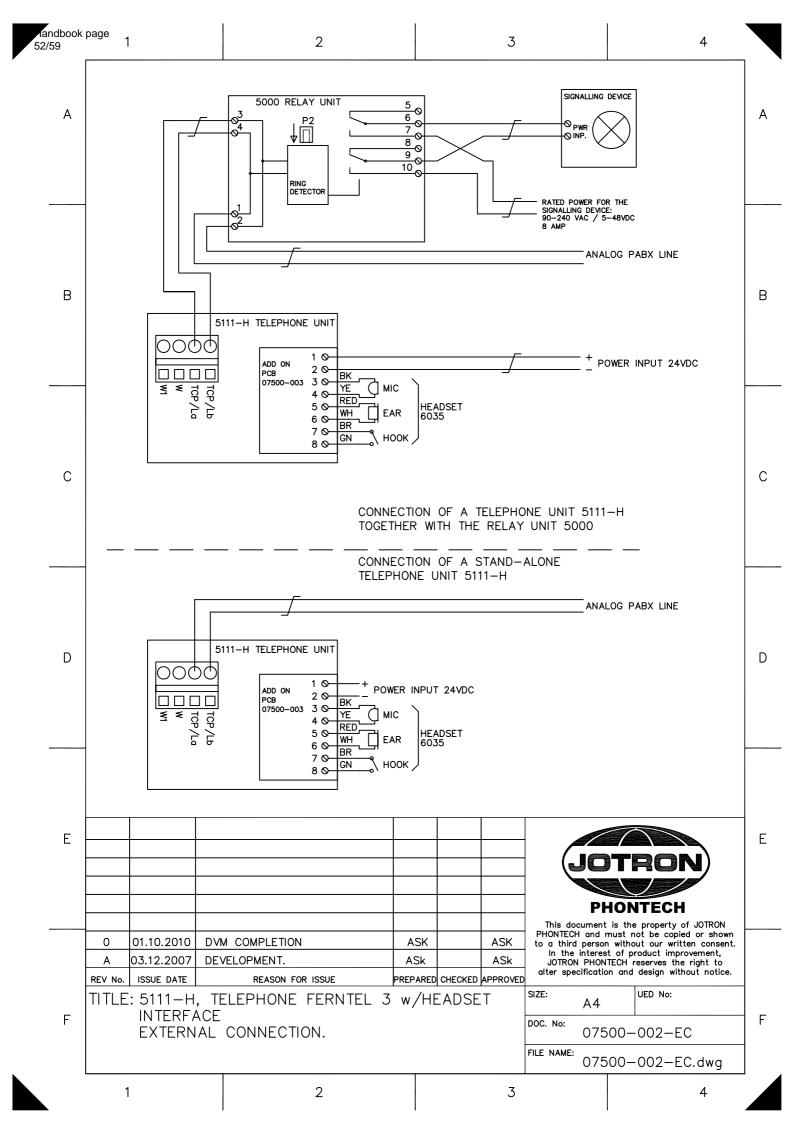
Туре	Name	Version		Art. no. *	
FernTel 3	Desk/Wall Telephone	without display	with spiral cord	112 300 2	
FernTel 3	Desk/Wall Telephone	with display	with spiral cord	112 310 2	
FernTel 3	Desk/Wall Telephone	without display	with flexible steel armoured cord	112 320 2	
FernTel 3	Desk/Wall Telephone	with display	with flexible steel armoured cord	112 330 2	
FernTel 3 ZB	Desk/Wall Telephone	without keypad	with flexible steel armoured cord	112 350 2	
Accessories	Stabiliser bracket			112 390 00	

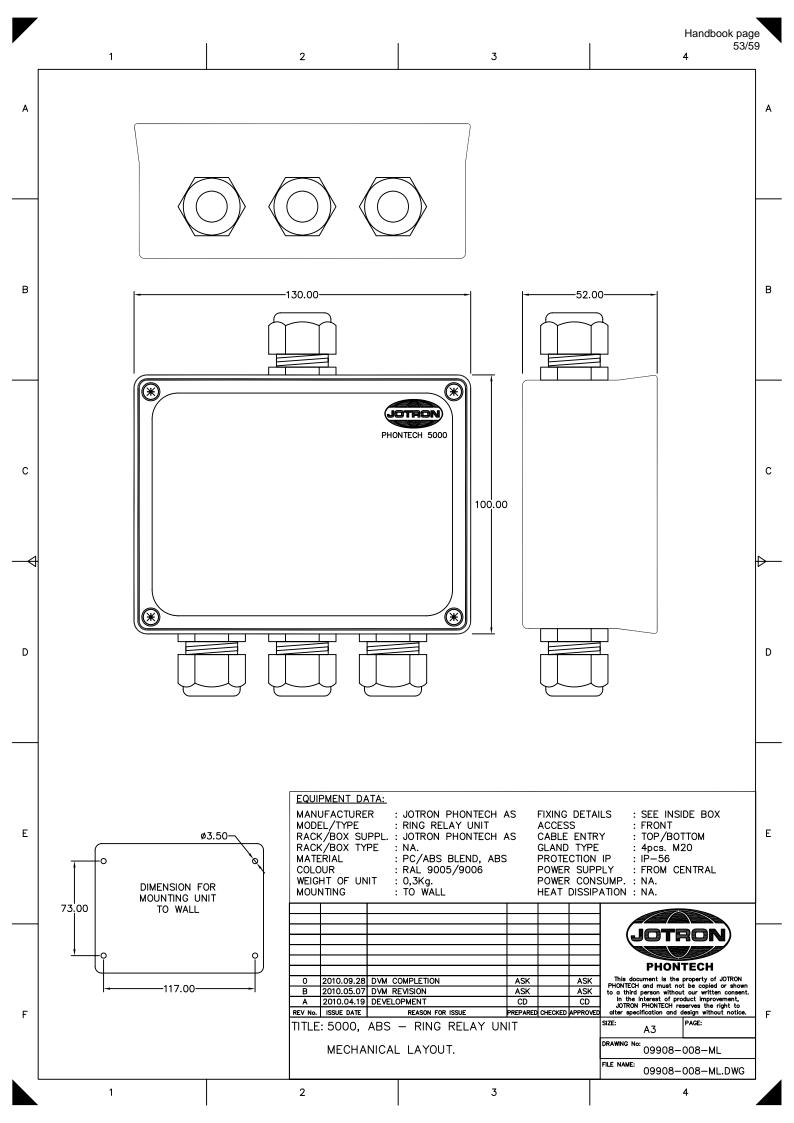


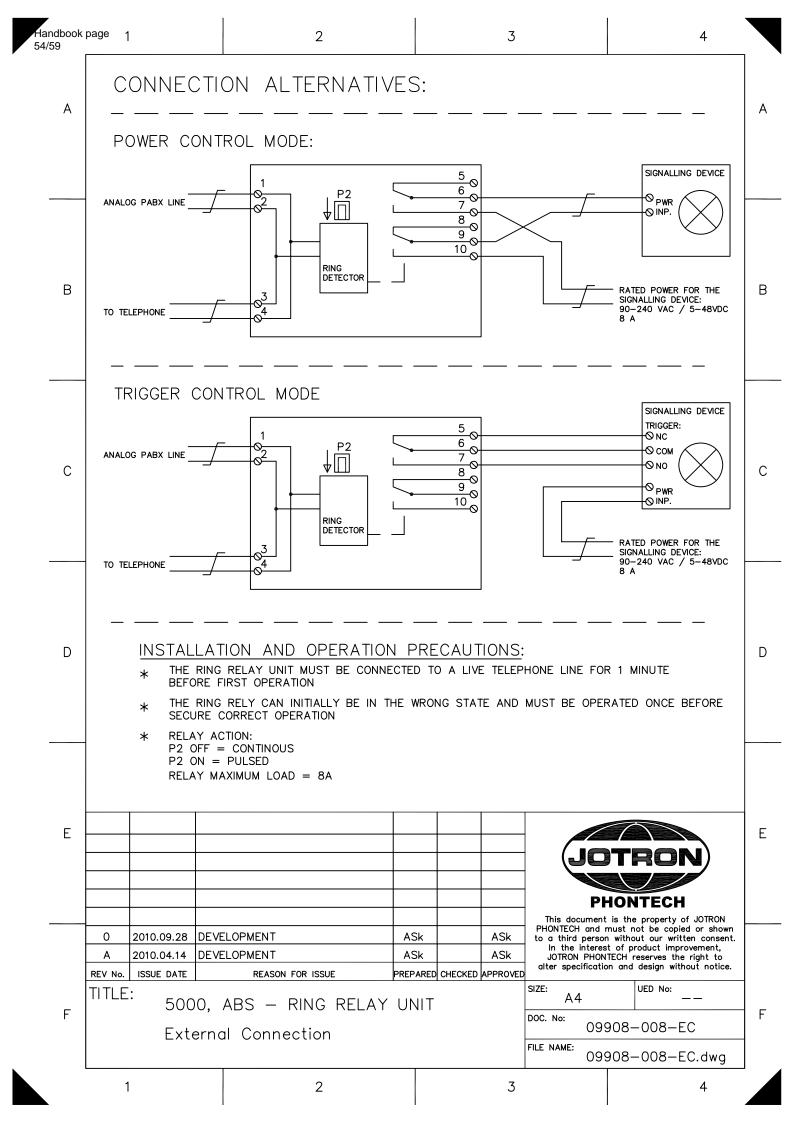
















ExII-Telephone ExResistTel

Explosion-proof, weatherproof Industrial telephone

- Handsfree
- ➤ Temperature range -25°C to +60°C
- Certified for dust and gas atmospheres
- Display
- ▶ IP 66 EN 60529
- different housing colours

Application

Communication devices for use in hazardous areas in the industry have to be especially well adapted to the extreme operating conditions they will be exposed to.

Our Ex-telephone has been developed for operation in the petro-chemical industry, on off-shore plants, in mills and harbours, which means it is resistant to large temperature differences, air humidity, sea water, dust and strong mechanical wear and tear. It is certified for use in hazardous dust and gas atmospheres.

The ExResistTel is completely programmable, and has been equipped with a 21-piece stainless (V4A) steel keypad designed for use with gloves. Letters and figures are presented clearly on the alphanumerical display.

The ExResistTel also boasts all the convenient features that have become standard in the field of office communication.

A string of optional extras and components – especially certified for hazardous areas – makes our telephone even more functional.

Telephoning in the off-shore

The housing is made of impact-resistant and shockproof glass-fibre-reinforced polyester and is resistant to acids, sea water, alkalis, moisture and grease.



Technical specifications

Connections

Types of protection II 2G Ex e mb [ib] IIC T5 Gb II 2G Ex e mb [ib] IIC T6 Gb

II 2D Ex tb [ib] IIIC T100 °C Db $\,$ II 2D Ex tb [ib] IIIC T80 °C Db $\,$

 $-25\,^{\circ}\text{C} \le T_a \le 60\,^{\circ}\text{C}$ $-25\,^{\circ}\text{C} \le T_a \le 40\,^{\circ}\text{C}$

Approval DMT 02 ATEX E 183 Line voltage 24 V_{DC} to 66 V_{DC} Line current 15 mApc to 100 mApc

Ringing alternating current 24 V_{AC} to 90 V_{AC} (at 21...54 Hz ringing frequency)

30 V_{AC} to 90 V_{AC} (at 16,6...54 Hz ringing frequency)

Ringing impedance greater than 6,0 K Ω at 25 Hz and 24...90 V_{AC}

greater than 4,0 K Ω at 50 Hz and 24...90 V_{AC} Flash function adjustable from 40 ms to 399 ms

Inquiry key Flash function adjustable from 40 ms to 399 m

Dialling procedure PD-DTMF operation to be set in the menu.

PD operation where the pulse/pause ratio can be set to 1.5:1 or 2:1 in the menu.

W-conductor Connection for external secondary sounder Screw terminals up to 4 mm² rigid. Up to 2.5 mm² flexible

Housing

Material Glass-fibre-reinforced polyester

Height x Width x Depth approx. 266 mm x 227 mm x 135 mm

Weight approx. 5.5 kg

Display 2-line alphanumerical display with pictograms,

Visible area approx. 78 mm x 26 mm

Keypad With ice protection,

21 keys with ABC lettering for name entries

Receiver

Stabilizer bracket Integrated, adjustable stabilizer bracket

Handset cord Stainless steel (V4A) armoured handset cord

Receiver inset Dynamic receiver inset with leakage field spool

for inductive coupling of hearing aids

Mouthpiece Electret-foil microphone

Noise suppression greater than 3dB due to integrated mouthpiece

horn mouth

Environmental conditions

Degree of protection IP 66 according to EN60529 Impact protection IK 09 according to EN50102

Operation temperature -25 °C to +60 °C for temperature class T5

-25°C to +40°C for temperature class T6

Storage temperature: -25°C to +70°C

Further characteristics

Optical call signalling Display shows (((🛕)))

Ringing sound pressure level approx. 90 dB(A) at 1m distance

Ringing melodies 10 melodies selectable

Listening by loudspeaker Maximum sound pressure level

approx. 68 dB (A) at 1m distance

Handsfree operation Maximum sound pressure level

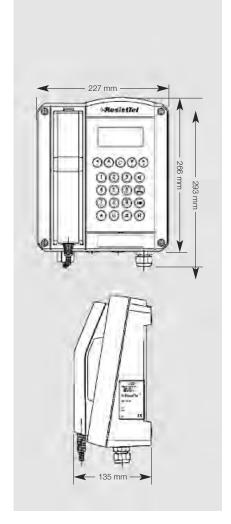
approx. 68 dB (A) at 1m distance

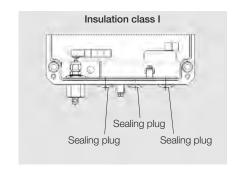
Amplified listening in receiver Receiver volume can be boosted,

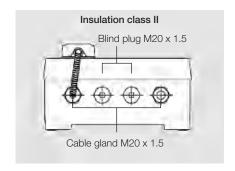
in 7 steps from 0-12 dB(A)

Menus in several language

Telephone directory max. 50 entries (names and numbers)







Accessories



ExII-Additional headset



ExII-Additional earpiece



ExII-Loudspeaker set



ExII-TWIN



Protection hood



ExII-Secondary sounder

* The full article number is made up by appending the colour code for the coloured cap to the article number given below.

transp. 11 red 12 amber 13

green 14 blue 15

Order information

Туре	Name	Versio	n	Article no.
ExResistTel	ExII-Telephone	black		112 861 01
ExResistTel	ExII-Telephone	red		112 861 0102
ExResistTel	ExII-Telephone ZB	black	 without keypad and display 	112 861 02
ExResistTel	ExII-Telephone ZB	red	 without keypad and display 	112 861 0202
ExResistTel	ExII-Telephone	black	- Protection class I	112 862 01
Accessories	ExII-Additional earpiece			112 861 03
Accessories	ExII-Additional headset			112 861 04
Accessories	ExII-Loudspeaker set			112 861 05
Accessories	ExII-Secondary sounder			211 842 06
Accessories	Protection hood	hot galv	vanized, yellow	118 901 01
Accessories	Protection hood	stainles	s steel	118 901 11
Accessories	ExII-TWIN			118 833*

Subject to change without notice \cdot Printout 07/14





POSSIO SVEA II GSM Connector

- **▶** Connect your Analogue Telephone Equipment
- **▶** Fixed Landline Replacement
- Handles Fax
- ▶ Easy to Install and Manage
- Optimized for Every Country
- ▶ High Quality Voice with Cinterion Module



About SVEA II

SVEA II GSM Connector is a Fixed Cellular Terminal (FCT) that converts standard analogue voice, fax & data services into GSM services.

SVEA II fullfils all requirements for fixed line such as: national, international and mobile calls, sending and receiving fax, data transmission and dialup connection to Internet.

SVEA II makes all well-known fixed-line services available over GSM and at the same time in an intelligent way eliminates differences between GSM and PSTN networks.

SVEA II is an easy to use product, with built-in internal GSM antenna (with possibility to connect external antenna), built-in battery backup and external power supply. It can easily be wall mounted and requires only SIM card installation.

SVEA II, among other services, includes an echo cancelling system for adaptation to different line impedance, a fixed dial system for minimising the waiting time when dialing out and support for GSM 850/900/1800/1900 technology. CDMA version is also available.

SVEA II has a remote management system, dedicated for Network Operators, to keep control of installed units in the field and receive alarms from them. SVEA II is also equipped with a tamper alarm to prevent unauthorized usage of the unit.

LED indicator for battery backup, line status and signal level.

SVEA II is used by major operators and is perfect for mobile offices, temporary buildings and remote areas.

SVEA II as fixed line

- Two independent lines:
 - a. RJ11 analogue port for ordinary phones
 - b. RJ11 analogue port for fax and data (optional)
- GPRS/EDGE and CSD support
- National alignments: tones, cadence, etc
- Data transmission services V.90 and V.92-optional (ISP/server mode), V.34
- FSK/CLIP presentation
- Intelligent Quick Dial System, that route the dialed number immediately through the GSM net at the last dialed digit

Additional Features ■ Additiona

- Easily programmable locally via PC and/or DTMF
- Protection system 3 level security access system
- Remote Possio management system for configuration and service of all units in the field
- Fraud protection and detection system
- Built-in GSM antenna, TNC plug for external antenna attachment
- PIN code management
- Service & Network operator SIM card lock system
- Built-in battery backup
- LED indicator for battery backup, line status and signal level
- Build in Echo cancellation system for adaptation to different line impedance
- Polarity reversal line when B-answer



Telephone

Fax

Connectors

- 1. Power plug for power supply, converts 220V/230V voltage to 18V voltage
- 2. Telephone plug RJ11 for analog telephone with DTMF or Pulse
- 3. Fax/modem plug RJ11 for analog fax machine G3 or analog modem V90
- 4. RS232 port for programming SVEA II GSM Connector. Only for authorized person.
- 5. USB port Data port GPRS/EDGE. Contact Possio for more information.



▶ Tech Spec

General

Quad Band GSM850/950/1800/1900 Mhz GPRS, EDGE and CSD support Fax Group 3, Class 1

GSM performance

Voice and emergency calls (TCH / FS) Half Rate (HR), Full Rate (FR), Enhanced Full Rate (EFR) and Adaptive Multi Rate (AMR)

Output power

Class 4 (2W) for EGSM850/900

Class 1 (1W) for GSM1800/1900 Class E2 (0,5W) for EDGE850/900 Class E2 (0,4W) for EDGE1800/1900

Data connectivity

EDGE class 12, max 236,8 kbps GPRS Class 12, max 86 kbps Input AC: 110-240V AC, 50/60 Hz Output DC: 18V DC, 0,8AUSSD

Mechanical features

2 independent RJ11 connectors

Internal GSM antenna, 50 Ohm External antenna connector TNC (f) Input AC: 110-240V AC, 50/60 Hz

Output DC: 18V DC, 0,8A

Physical description

[H] 210 mm x [W] 200 mm x [D] 55 mm 1540g (incl. ext. power supply)

Operating temperature

from -10° C to $+55^{\circ}$ C

SVEA is RoHS compliant

P.O. BOX 1236 SE-164 28 Kista **SWEDEN**

VISITING ADDRESS

Isafjordsgatan 35,8th fl. Kista (Stockholm) **SWEDEN**

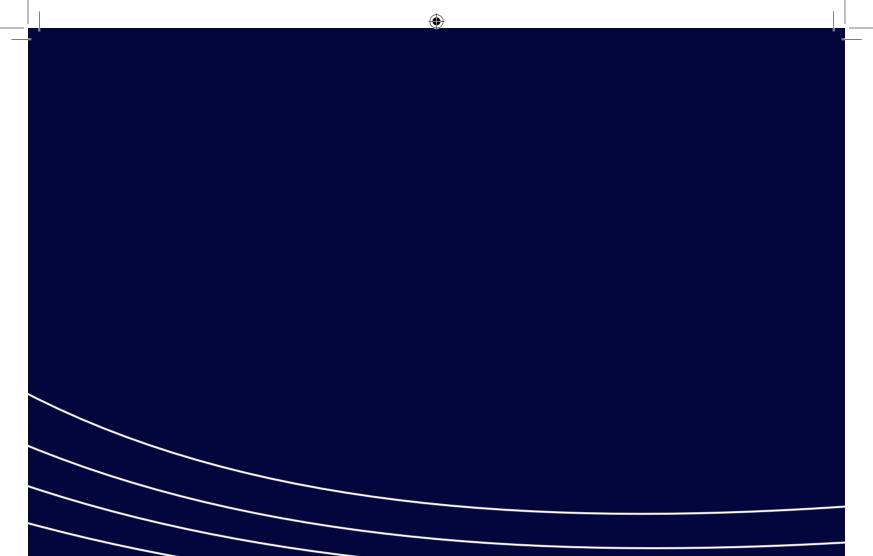
WEB

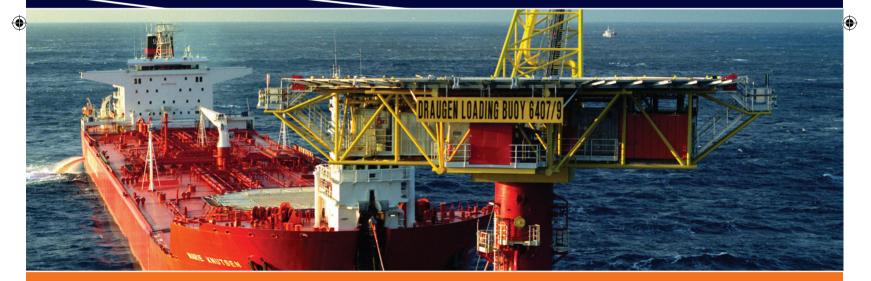
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www.jotron.com