

NT1+2a/b



Supplementary Services User Manual

Revision 1.3 2014/04/02

103987



Safety precautions

Before installing and commissioning the NT1+2a/b, please read these operating instructions thoroughly. These instructions will help you to use the device to its full extent and to avoid damage resulting from improper application.

- The NT1+2a/b has been manufactured according to state-of-the-art technology and corresponds to well-accepted safety regulations.
- Please make sure that the device is operated only in faultless condition and that these operating instructions are observed.
- Only authorized specialists are allowed to operate the NT1+2a/b and to open the device.
- After connecting the device, install the right supply voltage (\rightarrow Technical data)!
- Before opening the device, disconnect mains plug and remove cable from U interface!
- The NT1+2a/b has no individual component for interrupting the supply voltage. The device shall be placed within 2 meters from the mains socket.
- Care should be taken when installing the power supply and the connection cables to avoid the risk of accidents (such as tripping over the cables).
- The device shall be operated only within a temperature range of 0°C to +40°C.
- In case of device failure, please contact the manufacturer or the representative in your vicinity.
- Only such devices that will meet the electric safety requirements according to EN 60950 and that are marked with the CE symbol shall be connected to the interfaces of the NT1+2a/b.
- The terminals shall be equipped with appropriate connectors, otherwise adequate adapters may be used.
- The terminals shall be connected only to the corresponding interfaces that have been designed for them.

Apart from these safety instructions, installation and operation instructions are also given in the respective chapters.

We reserve the right to modify the contents of these operating instructions.



Regarding the configuration and installation of your NT1+2a/b, please ask your telecommunications network operator for further information.



Table of contents

1		Introduction	2
2		Symbols and abbreviations	2
3		Installation	3
4		Configuration	4
5		Service and configuration description	5
	5.1	MSN – Multiple Subscriber Number	5
	5.2	Terminal type	6
	5.3	Emergency terminal	6
	5.4	Device Type Configuration	7
	5.5	CH - Call Hold	7
	5.5	5.1 Standard Functions	7
	5.5	5.2 Extended functions	8
	5.6	CW – Call Waiting	9
	5.6	6.1 Activation and deactivation	9
	5.6	6.2 Active call phase	9
	5.7	3PTY – Three Party Conference	10
	5.8	CF - Call Forwarding	11
	5.8	6.1 CFU – Call Forwarding Unconditional	11
	5.8	3.2 CFB - Call Forwarding On Busy	11
	5.8	3.3 CFNR – Call Forwarding No Reply	12
	5.8	8.4 Call Forwarding control Interrogation	12
	5.9	CLIP – Calling Line Identification Presentation	13
	5.10	CLIR – Calling Line Identification Restriction	14
	5.11	COLR – Connected Line Identification Restriction	14
	5.12	CBT – Clearback Time Delay	15
	5.13	Metering Pulses AOC-D	15
	5.14	Malicious Call Identification (MCID)	16
	5.15	TP – Terminal Portability	16
	5.16	Return to default setting	17
	5.17	Configuration of the S/T-bus interface	17
	5.18	Incoming Call Management	17
6		Optical Signalling at NT1+2a/b	18
7		Connection configuration	18
8		Positions of switches and connectors	19
9		Bus configuration of S/T interface	19
1()	Technical Details	20
11	1	Block Diagram	21



1 Introduction

The NT1+2a/b is the joining element between the digital network of the telecommunications company and the ISDN respective analogue terminals.

Should any questions regarding installation or operation of your NT1+2a/b arise, please contact directly your telecommunications company. Competent specialists there will give you the required support. Do not start installation before your telecom access will be enabled. Your telecommunications company will inform you about the exact date when this will be effected!

This document describes handling of supplementary services and configuration procedures for the ELCON NT1+2a/b.

The content of this document is subject to change.

2 Symbols and abbreviations

- …The receiver is hooked on
- \hat{U} ... The receiver is hooked off
- ① …Call is active
- \mathbb{O} \mathbb{Q} ... The receiver is hooked on
- ...A special tone in your receiver (dial tone, busy tone, error tone, ...)
- \bigcirc ... The phone is ringing
- •••• ...Error tone sequence (a single
 - ...sequence of 4 or 5 short tones)
- **629** ...Dialling the destination number
- ® ...Key ,R-Flash'
- [D] ...A digit in the range from (1) to (9)
- * ...Star key at your analogue terminal
- # ...Hatch key at your analogue terminal



3 Installation

Wall mounting

The NT1+2a/b unit is designed for wall mounting. The diagrams in figure 1, figure 2 and figure 3 show the connections to be made for the NT1+2a/b unit.

Mount the NT1+2a/b close to the transmission cable of the telecommunications operator. In addition, you will need a mains socket for the line cord (max. 2 m long).

CAUTION: Before defining the position of the drill holes make sure that there are no other concealed installations, e.g. water pipes, electrical power cables etc.

Mark the drill holes on the intended place of the wall. After drilling the holes (\emptyset 6 mm) and inserting the dowels, screw the 3.5 mm round-head screws into the wall as far as the distance between wall and screw head is only approx. 2 mm. Hang the NT1+2a/b over the screw heads and pull it down until it locks safely in its place.

IMPORTANT: Do not permanently expose the NT1+2a/b to the sun! Please keep it dry!

Connecting the NT1+2a/b to the ISDN line (U interface)

A RJ11 connector is available to connect the U interface directly. The position of the U interface connector is shown in figures 2 and 3.

Connecting the termination units

Connect your analogue terminal equipment, e.g. DTMF telephone, fax, etc. to the RJ11 connectors situated in the middle of the housing base or, alternatively, the connectors for a/b1 and a/b2 directly beside the RJ11 modular jacks where block wiring has been provided (Pins 3 and 4 are used in the RJ11 connector). **Connect the ISDN terminal equipment** by inserting the appropriate plug into the

RJ45 communication outlets on the right-side base of the NT1+2a/b. The 4 wire clamp connectors for bus installation at S/T interface are described in figure 1, so that wires for receive and transmit directions would not be mixed up.

Power supply

Now connect the 230 V a.c mains supply. Before switching on the 230 V a.c mains supply, please check that all other connections are correct and secured.

The installation is now complete.



4 Configuration

The NT1+2a/b is configured by means of DIP switches along with the DTMF signals of a connected analogue telephone. The DTMF settings are always available and not protected by a password.

DIP switch settings

The switch SW permits to set the basic configuration of the termination resistors at S/T interface.

Default settings is marked with **bold letters** here.

Switch	Default	Acts on	Effect
SW1	ON	S/T	ON : 100 Ω parallel SR
		Receive	OFF:
SW2	ON	S/T	ON : 100 Ω parallel SX
		Transmit	OFF:
SW3	OFF	S/T	ON: additional 100 Ω parallel SR
		Receive	OFF:
SW4	OFF	S/T	ON: additional 100 Ω parallel SX
		Transmit	OFF:

100 Ohm	ON 0FF 1 2 3 4	default setting	The pairs SW1, SW2 and SW3, SW4 are each
50 Ohm	ON 1 2 3 4		these combinations can be set: 50 Ω , 100 Ω or No
Termination resistors switched off	ON 0FF 1 2 3 4		Parallel Resistor.

Description of S/T interface connectors RJ45 modular jack and clamp connector



Figure 1



5 Service and configuration description

5.1 MSN – Multiple Subscriber Number

The NT1+2a/b can be configured with three telephone numbers (MSN1...MSN3) for each a/b port. The number should always be configured, because it is necessary for the Call Forwarding services. MSN1 is the number which is sent as the Calling Party Number. If you do not configure a number at MSN1 or configure a number which is not subscribed by your network provider, the provider will replace your Calling Party Number with the default one. Start always please with MSN1 for configuration. If you like to configure a/b1 please use analogue terminal at analogue port a/b1. For the other case to configure a/b2 please use the analogue telephone at a/b2.

Configure your telephone number MSN for analogue port a/b1 or a/b2: MSN1 for a/bx:

x = 1 for a/b port 1 and x = 2 for a/b port 2

620 is one of your subscribed telephone numbers you wish to configure (without area code)

Configuration example:

Note

- The right configuration of your telephone numbers is very important for the selection of your telephone when an incoming call occurs and for the ringing signal in that case.
- > If no number is configured in MSN 1, all incoming calls are accepted
- If a/b1 and a/b2 use same numbers, both telephones will ring (if they are free) and the one which hooked-off first, will receive the call.
- To disable your NT1+2a/b for incoming calls, configure a number not subscribed for the MSN (i.e. @@@).
- > To clear an MSN, use ,* #' instead of ,* **524** #' in the above sequence.



5.2 Terminal type

For each telephone number of your NT1+2a/b you can select three types of device type:

Select the device type for a telephone number:

☎ û)◀	# 9 7 * 1 0 # 4	🕽 🖟 🖀 Speech
☎ û)◀	# 9 7 * 1 1 # 4	🕽 🖟 🖀 FAX
☎ û)◀	# 9 7 * 1 2 # 4	🛈 🖓 🛣 Modem

Tip

- Default is Speech. Modem works for all combinations. Selection of another type exclude calls only if they come from or go to ISDN terminals
- If you select type facsimile for a telephone device you probably will receive no tone signalisation from the network (busy, disconnect).

5.3 Emergency terminal

Your NT1+2a/b is normally connected to your local power supply. In this case both a/b ports and the S/T port are available. In case of local power supply breakdown, the NT1+2a/b can still work, but in restricted mode only. In that case one device remains accessible. This could be either a/b1 or a/b2 or one ISDN telephone connected to the S/T port. You can

select the device for emergency mode at S/T interface:

Ŧ	⇧☽◀	#97	* 3 0 # 📢) () () A
---	-----	-----	-----------	------------------

In restricted mode the analogue ports are dynamical (a/b 1 or a/b 2), i.e. the connection is established to that a/b-port where the receiver is hooked off first. At this moment the other port is disabled. This is unavoidable, due to the fact that the exchange provides only a limited wattage.

Note

- > Default setting for emergency terminal is analogue port.
- Don't forget to select emergency mode on your connected ISDN telephone in this case.
- To avoid blocking emergency terminal selection will be rejected in emergency state.



5.4 Device Type Configuration

It is possible to select between different device type mode settings. You can select between 4 different modes (see figure 3 and 4 for description of port type)

NT1+2a/b (default): full functionality, S/T interface and both analogue ports ON #97*2**0# ① ↓ 鴌 NT1+a/b1: S/T interface and analogue port a/b1 ON; a/b2 OFF #97*2**1# ① ↓ 奮 NT1+a/b2: S/T interface and analogue port a/b2 ON; a/b1 OFF)) 🖟 🖀 **霍**介() • #97*2**2# NT1: only S/T interface ON; both analogue ports OFF #97*2**3# ① ↓ 奮

You can only switch off an analogue port if this port is not configured for emergency mode. In case only NT1-mode no incoming or outgoing calls were possible at analogue ports. Only configuration entry is in NT1-mode allowed.

5.5 CH - Call Hold

5.5.1 Standard Functions

If you have an active call and want to talk to another partner without disconnecting your first partner, you can interrupt your present call in order

To establish a second call: $\bigcirc \mathbb{R} \oslash \blacktriangleleft \heartsuit \oslash \spadesuit \oslash$

While the phone is ringing at the second subscriber or while you are talking to him (after he has hooked off), your first connection is on hold. You can (either)

Switch between two calls:

) R I 2)

When you try this and the subscriber on hold has already cleared his call you will

Receive the error tone sequence:	(if the call on hold was cleared)
••••	

If you now try it a second time, you can establish a new call as described in the first example of this paragraph.



If you have still a waiting partner and your active partner hooks on, you will hear a disconnect tone \blacktriangleleft . If you now hook on your receiver O, and the second partner is still waiting, the NT1+2a/b will

Remind you of the call in hold state:

AA

Your telephone is now ringing continuously for about 10 seconds. If you hook off $\widehat{1}$ within this time, you will be connected with the call on hold. Otherwise this call will be cleared.

If you have an active call and a call on hold and you hook on \mathbb{D} , your receiver first, then the active call will be cleared and the NT1+2a/b will remind you of the waiting call in the same way as described above.

There is still another way to retrieve a waiting call after your active partner has hooked on and you hear the disconnect tone. If you haven't hooked on yet, you can

Switch to the waiting call:

(with no call active)

) R 2)

If the waiting call was already cleared, you will hear the disconnect tone for a short time overlaid with $\bullet \bullet \bullet \bullet$ and you should hook on \mathbb{O} \mathbb{Q} at this point.

5.5.2 Extended functions

There are still some functions for clearing calls if you have an active and a call on hold. But note the following.

Disconnect the hold call:

() ◀ ℝ () ()

The partner still on hold will now hear the disconnect tone. You can continue your present call.

Terminate the active call and switch to the one on hold:

 $\bigcirc \blacksquare \blacksquare \bigcirc \bigcirc \blacksquare$

You are now connected to the call on hold. Your first partner will now hear the disconnect tone.



Note

- > Although you are allowed to clear a call on hold, you should do this only with active calls by 𝔅𝔅. If there is still a call on hold, your NT1+2a/b will remind you of that by 𝔅𝔅. If you want to clear the call on hold, first switch to that call with 𝔅 ◀
 𝔅 (say ,goodbye') and 𝔅𝔅.
- Key strokes for one active call and one incoming call are described in paragraph 5.6 CW - Call Waiting.

5.6 CW – Call Waiting

This service allows the indication of an incoming call when another call is already active.

5.6.1 Activation and deactivation

Activation and deactivation of this service are set with a configuration key sequence during an inactive call phase of your NT1+2a/b.

Activation of CW:

☎ û) • * ④ ③ # •) ↓ ☎

Deactivation of CW:

☎ û) • # ④ ③ # •) ↓ ☎

Note

- Activation and deactivation are valid only for that a/b interface, where that telephone is connected to.
- > Key strokes for one active call and one in hold position are described in 5.5
- > Default setting: CW is activated for both analogue ports.

5.6.2 Active call phase

If you are calling with one partner and another subscriber dials your number, he normally will receive a busy tone. With an activated CW service you will hear a special tone in your receiver. You now can decide either to switch to the waiting call (with and without termination of the present call), or to reject the waiting call, or to suppress the CW tone only.

Reject a waiting call:

() ◀ ℝ () ()



The calling partner will now receive a busy tone or a rejection message from your network provider. You can continue your actual call. New incoming calls will now be rejected until you hook on.

Terminate an active call and switch to a waiting one:

You are now connected to the waiting call. Your first partner will receive a disconnect tone.

Put the actual call into hold state and switch to the waiting one:

③ ◀ ℝ ② ③

The partner in hold will receive a special tone or message from the network provider. You are now in a state described in paragraph 'CH'-Call Hold and have to follow the instructions there to handle your two calls. It is possible to establish a Three Party Conference described in 5.7.

Suppress the CW tone:

) 🕇 🌀)

Pressing of any digit (without leading \mathbb{B} !) will suppress the CW tone only. The calling subscriber will furthermore receive the call tone and you have still the opportunity to switch to the incoming call.

5.7 3PTY – Three Party Conference

If you want to talk with two external subscribers at the same time, you can establish a "Three-party conference".

If one call is on hold (see 5.5) and one call is active, you can

Establish a Three Party Conference:

⑦ ℝ ◀ ③ ①
If you talk with two subscribers at the same time you can

End a Three Party Conference:

) ℝ ◀ ②)

After execution of the last command you are connected to the same partner as before the conference was activated. The other one is on hold again. If one partner hooks on while conference is active, a normal connection to the remaining subscriber is established. If you want to clear both connections, you simply have to hook on.



5.8 CF - Call Forwarding

Call Forwarding is a feature that allows you to forward incoming calls to another telephone number. You must configure the MSN (see point 5.1) for the analogue port which should be used for Call Forwarding. The activation for a Call Forwarding service is for all numbers which are configured at this analogue port. These services CFU, CFB, CFNR work on MSN1 of the served port without SUN. [*SUN] is optional. SUN = Served user number is MSN1, MSN2 or MSN3 for the number which is programmed in point 5.1. It means the place of the MSN. SUN =1, 2, or 3 and inserts the configured MSN 1, 2, or 3 in the outgoing command. So can for example choose only calls to MSN2 for Call Forwarding to an other access. Your NT1+2a/b supports three different kinds of Call Forwarding: CFU, CFB, CFNR

5.8.1 CFU – Call Forwarding Unconditional

When you activate this feature, every call to your telephone number will not reach you, but the number you selected before when activating this service. You can use this service, for example, if you want to redirect every call from your home telephone to your office or vice versa.

Activation CFU: for one telephone number MSN1 O \rule{O} O \rule{O} $\rule{O$

5.8.2 CFB - Call Forwarding On Busy

This service works in the same way as CFU, but redirection only takes place when your telephone is busy. You can use this service, for example, when you own two telephone numbers (one official and one private) and your first number is often occupied (by your daughter or someone else), but you want to receive all the calls to your first number.

Note

If you want to use this service, you should disable Call Waiting service (see 5.6)



5.8.3 CFNR – Call Forwarding No Reply

This service works in the same way like CFU, but redirection only takes place when nobody answers the phone and your telephone is already ringing for about 20 seconds (depends on your provider). You can use this service just like CFU, but it offers you the opportunity to answer when you are at home.

☎ û 🕽 � # ⑥ ① [*SUN] #

Note

① ↓ 奮

- > If you have forgotten which forwarding was enabled, try to disable all of them.
- Keep in mind that after activation of CFU no call will be put through to you anymore.

5.8.4 Call Forwarding control Interrogation

In principle, it is possible to select several Call Forwarding services for MSN1 (e.g. CFNR + CFB). The signalling, however, indicates only one existing Call Forwarding function. Besides, an eventual replacement of the NT1+2a/b, a reset or a mal-function may result in certain inconsistencies within the Call Forwarding data. This is the reason why an inquiry of the Call Forwarding procedure has been implemented.



Upon entering the command, below tone signals can be heard:				
No tone	Exchange does not react to the inquiry			
Accept tone for CFX	Call Forwarding enabled (80 ms Tone 800 ms Pause)			
-	tone tone tone tone			
Disc tone for CFX	Call Forwarding not enabled (500 ms Tone 100 ms Pause)			
	tone tone •••			
Reject tone	Inquiry rejected (8 x {100 ms Tone 100 ms Pause}) if Service			
-	is not available, wrong MSN saved under MSN 1,2 or 3, etc.			
	tone tone tone tone tone tone			
Description of the tone signals				

Description of the tone signals

Accept tone	700 ms Tone 600 ms Pause 700 ms Tone <only one="" time=""></only>
	tone tone
Reject tone	8 x {100 ms Tone 100 ms Pause} Pause <only one="" time=""></only>
	tone tone tone tone tone tone

After the Reject tone please start your program sequence again. In such a case there sequence was faulty or spurious. After the error tone special tones can be heard which may differ according to the device state and the national ISDN configuration. If any service cannot be enabled, please ask your network provider.

5.9 CLIP – Calling Line Identification Presentation

While you send an information with your telephone number to the subscriber you called, the number of a calling subscriber is delivered with a call. This number can be transmitted to your telephone in form of FSK procedures. To visualize the number of the caller you need a special CLASS or CLIP telephone device which support this service.

To dea	ctivate	the CLIF	functio	n:		
2 10	₽◀ #	90*	5 0 #	◀	ĴĴ	Ŧ

To activate the CLIP function:

☎ ৫ 🕽 🖣 🛛 # 🤋 🗇 🛪 🔄 🛈 # 🖣 🔹 🕄 🖓 🖀

The transmission was only ON HOOK tested. (with handset Hang Off)

Note

- > For the special telephone set ask your service provider.
- > Some network providers transmit identity although restriction is activated.
- > Default setting: CLIP enabled



5.10 CLIR – Calling Line Identification Restriction

If you call another subscriber, your telephone number is transmitted with the call. If the other subscriber uses an ISDN equipment or a special device for CLIP, your number is presented to him. If you do not want your number to be transmitted, you can activate the identification restriction in two ways. Either you hide your number permanently or you do it only for the next call.

 Activation of permanent suppression of your identification:

 [^]
 [^]

If your identity presentation is activated permanently and you wish to make an Activation of identification suppression for the present call only (CLIR per Call):

☎ û) • * 3 ① * • 524)

Note

> Default setting: CLIR deactivated.

5.11 COLR – Connected Line Identification Restriction

In the same way your number is transmitted to the network when you receive a call. (This is to show the real receiver when a Call Forwarding is established.) If you don't like to show your real telephone number to the subscriber for incoming calls select:



5.12CBT – Clearback Time Delay

This service allows you to change your terminal equipment at the a/b interface without clearing an active connection. The service is active for incoming calls only. A possible application is the situation when you receive an analogue fax on a line that has been established for telephony connection. After you hear the specific fax sound, you can disconnect your phone and connect an analogue fax instead of it. For this operation you have 2 minutes time. After this time the call is disconnected by your NT1+2a/b.

Configuration is only valid for the own (configuring port).

Note

- If the service is activated, it works for all incoming calls, i.e. if you hook on ① ↓ during an active call, the call will still remain active for 2 minutes (unless your partner does not hook on). If you want to establish a new call within this time ① ①, you will still be connected to your first connection. So you should be careful about using this service.
- Default setting: CBT activated

5.13 Metering Pulses AOC-D

This function enables tax metering pulses 16 kHz if your analogue terminal supports the notification of this service. The configuration is valid only for that analogue port, where that telephone is connected to.

Activation of AOC-D:				
☎ û) •	# ⑨ ⑦ ∗ ④ ① # ◀)\$ 7		
Deactivatio	on of AOC-D:			
☎ û) •	# 9 7 * 4 0 # 4)↓ ☎		

Note

> Default setting: metering pulses deactivated.



5.14 Malicious Call Identification (MCID)

If your telecommunications network supports the MCID service attribute you can activate MCID with the following sequence. This service supports only incoming calls.

Activation:

....) **R** * 3 9 **#**)...

5.15TP – Terminal Portability

To change the terminal equipment or to change to another telephone connected to your NT1+2a/b during an active call you can use this service if supported by your provider. If you wish to

Suspend an active call:

) **R**6 4)**2**

whereby the park code is fixed automatically at default value 8. You need the same park code to identify the parked call at resume time.

After you have hooked on your telephone you can disconnect it and plug another one or you can change to another one.

To get you suspended call back you have to

Resume a suspended call:

whereby the same park code must be used as in the suspended call. Now you can continue your interrupted call.

If there was an other park code from your ISDN terminal used you can change the park code as described in the following sequence:

☎↑♫◀ #⑨⑦★⑥⑨★[D]#◀ℑϟ☎

[D] ... new reserved park code, which is used automatically for the next Suspend/Resume Procedure

Tip

- If another call is active while you try to suspend a call suspension is rejected by the network. You will hear the error tone sequence for a short time and your present call will stay active.
- If you suspend a call and before you resume another call is established by another telephone you will hear the busy tone after you hooked off. Nether less you can resume the call with the above sequence.
- You can also hand over a call between analogue and ISDN telephone. You only have to select the same park code. Please look inside the user manual of your ISDN terminal.



5.16 Return to default setting

In some cases it may be useful to set all functions back to the default settings of our company. For this function dial:

☎ û) • # 9 7 * 888 * 888 # •) ↓ ☎

After 2 seconds the green light-emitting diode starts with fast flashing. The device performs a RESET and start a new synchronisation with U interface. Please wait 20 seconds and then you can start with re-programming the device.

5.17 Configuration of the S/T-bus interface

It is possible to select between Short passive bus and Extended passive bus / Point-to-Point connection (cf. figure 4).

Extended passive bus / Point to Point:

☎ û) ◀ # ⑨ ⑦ * ④ ⑧ ① # ◀ 〕↓ ☎

Default: Short passive bus

5.18 Incoming Call Management

For use of a PABX at analogue ports it is possible to set all incoming calls send only to a/b1 for all incoming calls.

In case of compatible MSN and Terminal Type the function could be activated with:

Incoming calls are sent only to a/b1. If a/b1 is busy the incoming call will be sent to a/b2:

☎ û) ◀ # ⑨ ⑦ ★ ⑧ ① # ◀) ↓ ☎

Note

- > Default setting: Incoming calls are sent to both a/b1 and a/b2.
- For this service Call Waiting should be deactivated. Further should Clear Back Time Delay switched off.



6 Optical Signalling at NT1+2a/b

The equipment has 2 LED to display various operating states. The states marked in bold letters denote the display state in normal mode.

LED ~ (yellow)	ON OFF	local main power supply (230 V ~) remote power (no local power supply)
LED U (green)	ON	U interface is activated, an ISDN device is connected at S/T interface, Layer 1 is activated
	Flash slow (1 Hz)	U interface is activated, no ISDN device is connected at S/T interface; Layer 1 is activated via the analogue ports
	Flash fast (8 Hz)	U interface in synchronisation state,
	OFF	Layer 1 is deactivated (in "sleep" mode)

7 Connection configuration



Figure 2



8 Positions of switches and connectors





9 Bus configuration of S/T interface



Figure 4



10 Technical Details

U interface according to ETSI TS	102 080
Transmission method	Two-wire transmission, echo cancellation
Channel structure	2B+D, synchronization and service data
Line code	2B1Q
Transmission rate	80 kBd
Net data rate	144 kbps (2 × 64 kbps + 16 kbps)
Range	4.8 km with 0.4mm Cu pairs
	9.0 km with 0.6mm Cu pairs
Max. power consumption	1.4 W in emergency mode
S/T interface according to ETS 30	00 012
Transmission method	Four-wire duplex
Channel structure	2B+D, synchronization and monitoring
Line code	AMI (modified)
Total bit rate	192 kbps
Useful bit rate	144 kbps
Bus configuration:	- Short passive bus:
(default setting)	range approx. 150 m 220 m
	- Point-to-point:
	range approx. 1,000 m
a/b Ports according to national req	uirements
Powering current	≥ 20 mA
Line resistance	< 600 Ω
a/b voltage	$U \ge 44 V$
Call signals	\geq 40 V_{rms} at 2.2 k Ω + 3 μF
Call frequency	25 Hz
Charging level	> 400 mV _{rms}
Charging frequency	16 kHz
Impedance Z	600 Ω
Output level	- 7 dBr
Input level	0 dBr
Power supply according to 1.430	
Nominal voltage	230 V A.C. ± 10 %
Voltage range	196 V 254 V, 48 Hz 52Hz
Power consumption	Max. 20 VA locally powered
Dimensions and ambient temperation	ature
Storage/transport	-30 °C to +55 °C
Operation	0 °C to +55 °C
Casing dimensions	$195 \times 150 \times 44 \text{ mm}^3$
Weight	900 g



11 Block Diagram

Block diagram with functional description and connections for the NT1+2a/b.



Our products are subject to continuous further development and improvement.

Subject to technical changes without prior notice. 1681 V3 (especially the national requirements of the clients)

© ELCON Systemtechnik GmbH 2014

Declaration of Conformity We ELCON Systemtechnik GmbH (Manufacturer or his Authorised Representative established within the EAA) Obere Hauptstraße. 10 09232 Hartmannsdorf (Address) declare under our sole responsibility that the product NT1+2ab / 2B1Q (PCB identification number 6084) (Name, type or model) to which this declaration relates is in conformity with the following standard(s) or other normative document(s) EN 60950-1:2001 EN 55022:1998+A1:2000+A2:2003 ETSI EN 300 386 (2003-05) EN 61000-3-2:2000 EN 61000-3-3:1995+A1:2001 (Title and/or number and date of issue of the standard(s) or other normative document(s) if applicable) following the provisions of directive 89/336/EEC **EMC-Directive** 73/23/EEC Low Voltage Directive (Number and title of the European Council Directives which applies) Registration number of Type Examination Certificate: 06/05 Year in which the CE marking was affixed: 2006 Rademacher Hartmannsdorf, 2006-06-13 Managing Director (Place and date of issue) (Name and signature or equivalent marking of authorized person