



## PS 29M DUAL CHANNEL BELTPACK



## USER MANUAL

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## 1.0 GENERAL DESCRIPTION

The PS 29 M is a portable dual channel intercom station. It is housed in a strong metal case provided with a steel belt clip.

On the front panel the unit has, for each channel A and B separately, a Volume (listen level) control, a TALK push button and a CALL push button, both with LED indicators.

On the rear panel are two XLR-3 connectors for the intercom connection cables (two 'party lines') and an XLR-4 headset connector.

As an option an XLR-5 headset connector can be fitted, allowing to hear channel A at the left headset can and channel B at the right headset can.

On the bottom panel are side tone trimmers (one for each channel), a buzzer volume trimmer and a mic gain trimmer.

Special attention has been paid to the intelligibility of speech. By applying low noise / high speed op-amps, a speech presence filter and propriety high power bridged headphone amplifiers, communication is very comfortable even in environments with a high level of background noise.

The unique ASL call system provides both a flashing red LED and a very distinctive sound signal. A momentary push of the Call button makes the red LED flash whilst holding the button for 2 seconds activates the Call sound signal. The volume of this sound signal (the buzzer) can be adjusted at the front panel.

Fully electronic switching increases reliability and allows for:

- 'soft' microphone On switching, latched or momentary
- Remote Mic Mute facility

## 2.0 UNPACKING

The shipping carton contains the PS 29 and this user manual. If any are missing, contact your dealer.

ASL has taken great care to ensure this product reaches you in flawless condition.

After unpacking the unit please inspect for any physical damage and retain the shipping carton

and relevant packing materials for use should the unit need returning.

If any damage has occurred, please notify your dealer immediately so that a written claim can be initiated. Please also refer to the warranty section of this manual.

## 3.0 INSTALLATION

This PS 29 M will form a part of an existing or new intercom system. Connection to it is straightforward. There are no separate power connections or batteries to install as the necessary DC voltages are derived from the intercom master station or power supply, via the intercom connection cables.

To connect the PS 29 M to the intercom system use professional flexible microphone cable with 2 wires and 1 shield only.

The PS 29 M is fully protected against mis-wiring (reverse power) or short circuit in the interconnecting cables.

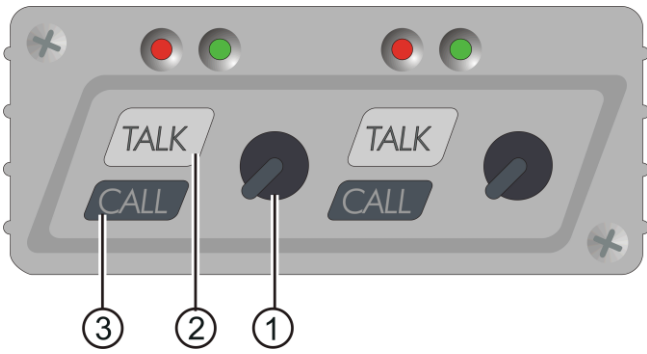
## 4.0 WARRANTY

This unit is warranted by ASL to the original end-user purchaser against defects in workmanship and materials in its manufacture for a period of 12 months from date of shipment to the end-user.

Faults arising from misuse, unauthorized modifications or accidents are not covered by this warranty.

If the unit is faulty it should be sent, in its original packing, to the supplier or your local ASL dealer, shipping costs prepaid. A note must be included stating the fault(s) found and a copy of the original suppliers invoice.

## 5.0 FRONT PANEL CONTROLS



### 1 VOLUME control knobs

These knobs are for adjusting the listen levels, for each channel separately.

### 2 TALK buttons

These buttons allow talking to each channel separately or to both channels simultaneously. When a TALK button is switched on, its large green LED is lit.

#### *Momentary switching:*

When a TALK button is pushed and held, the microphone signal is sent to the referring intercom channel until the button is released

#### *Latched switching:*

When a TALK button is pushed shortly it is electronically latched and the microphone signal is sent to the referring intercom channel. When pushed again the TALK button switches Off.

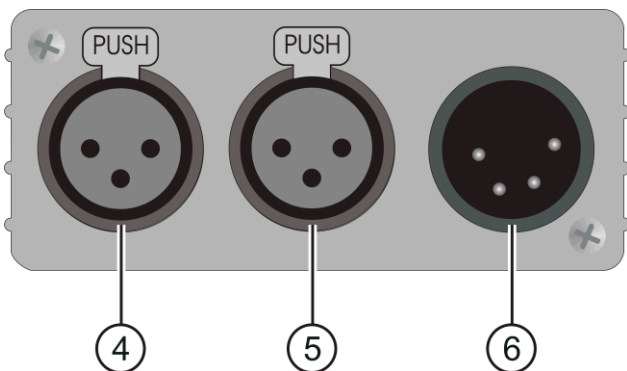
#### *Mic Mute when latched On:*

After a so-called Mic Mute signal has been received from a Pro Series master station or power supply, the connection between microphone and intercom channel is interrupted. By pushing the TALK button, the connection is restored and one can talk to the intercom channel again.

### 3 CALL buttons

By pushing one of this buttons the Call function is activated for the referring party line channel. With a momentary push a Call signal is sent to all intercom stations connected to that channel. The Call LED's on this PS 29 M and on all stations connected to the same party line start flashing. By pushing the Call button for 2 seconds the Call Buzzers are activated, provided the buzzers are not muted by a Buzzer Mute signal received from a Pro Series master station or power supply. After the CALL button is released the LED's continue to flash for a further 2 seconds.

## 6.0 REAR PANEL CONNECTORS



### 4 LINE connector channel A

### 5 LINE connector channel B

These XLR-3 connectors are for connecting the PS 29 M to intercom party lines.

#### Pin assignments:

Pin 1: 0V / ground shield  
Pin 2: +30V DC power wire  
Pin 3: audio wire

### 6 HEADSET connector

This XLR-4 connector is for connecting a headset to the PS 29 M beltpack.

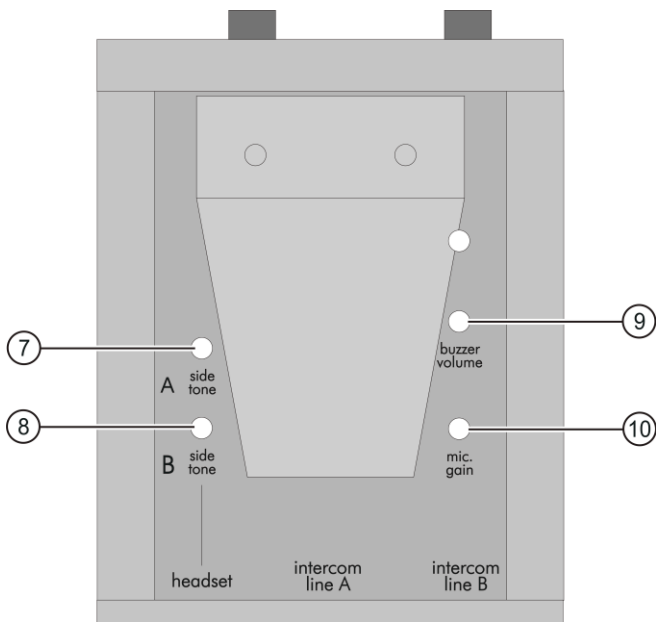
The impedance of the headset can must be minimum 200 ohms; in case the headset has 2 cans in parallel, the impedance of each can must be minimum 400 ohms.

#### Pin assignments:

Pin 1: shield mic. (GND)  
Pin 2: mic. +  
Pin 3: phones +  
Pin 4: phones -

As an option, the PS 29 M can be fitted with a XLR-5 headset connector, allowing to hear channel A on the left headset can and channel B on the right headset can.

## 7.0 BOTTOM PANEL CONTROLS



### 7 SIDE TONE trimmer channel A

### 8 SIDE TONE trimmer channel B

With the side tone trimmers #7 and #8 one adjusts the level of your voice as you hear in your headset, for each channel separately.

Adjustment procedure:

- Turn trimmer fully clockwise
- Switch the microphones of all (speaker) stations connected to the referring party line channel off

- Activate the Talk function of the referring channel
- Turn up the listen volume of this channel
- Speak into the headset microphone
- Adjust the listen level by turning the side tone trimmer

The operating area of the trimmers is between fully clockwise and minimum level. Adjusting the side tone does not affect the level of your voice as heard by the other stations on the referring party line channel.

### 9 BUZZER VOLUME trimmer

This trimmer adjusts the volume of the internal buzzer (located behind the front panel). The buzzer is activated when a CALL button on this beltpack (or on any other intercom station connected to one of both party lines) is pushed longer than 2 seconds, provided the buzzers are not muted by a Buzzer Mute signal received from a Pro Series master station or power supply.

### 10 MIC GAIN trimmer

This trimmer is to adjust the mic gain. To increase the gain, turn clockwise and to decrease the gain, turn counter clockwise.

## 8.0 TECHNICAL SPECIFICATIONS PS 29 M

### Intercom System

Dynamic range: 80 dB (1 kHz, THD 1%)  
Call signal: 2.8 mA  
Call signal threshold (receive): +2.4V DC  
Supply voltage: + 30V DC (12 – 32V DC)  
Power interrupt time (mic mute): 0.1 sec  
Line Impedance: 350  $\Omega$  (1 kHz) / 2.2 k $\Omega$  (DC)  
Audio level: nom. -18 dBu, max. +4 dBu

### Side Tone

Rejection: 0 – 30 dB adjustable

### Buzzer

Max. SPL: 85 dBA

### Mic pre-amp

Headset mic. impedance: 200  $\Omega$   
Gain: 35 – 60 dB (adjustable internally)  
Presence filter: +6 dB at 5 kHz  
Frequency response: 200Hz - 15 kHz (-3 dB)  
Power to electret mic: +9V DC

### Headphone Driver Amps

Max. load: 200  $\Omega$   
Max output level:  
monaural: 14 Vrms @ 200  $\Omega$   
binaural: 2x 10,6 Vrms @ 400  $\Omega$   
Max output power:  
monaural: 1 Wrms @ 200  $\Omega$   
binaural: 2x 0.28 Wrms @ 400  $\Omega$

### PS 29 M Power Consumption

Current (at 30V DC):  
30 mA quiescent,  
45 mA signaling  
170 mA at max. output + signaling

### PS 29 M Dimensions & Weight

Width: 90 mm / Height: 48 mm (incl. clip)  
Depth: 124 mm / Weight: 425 grams

0 dB defined as 775 mV into open circuit.

ASL reserves the right to alter specifications without prior notice

## 9.0 PART LINE, TECHNICAL CONCEPT

User stations in an ASL intercom system are interconnected via one or several 'party lines'. A party line offers two way ('full' duplex) communication and consists of standard microphone (multi-pair) cable. One wire is used for the audio signal and one wire for the DC power. The screen of the microphone cable functions as earth/return. Current drive is used for signal transfer. Each intercom station utilizes a current amplifier to amplify the microphone signal. That signal is put on the common audio line. Due to the constant line impedance, a signal voltage is developed which can be further amplified and sent to headphones or loudspeakers. (the line impedance is situated in the power supply between XLR pin 3 and pin 1)

This principle has the following advantages:

- The use of a single audio line allows several intercom stations to talk and listen simultaneously
- Due to the high bridging impedance offered by each intercom station. The number of stations on the party line has no influence on the level of the communications audio signal.
- Power and audio to the intercom stations use the same cable

Also the Call signal is sent as a current, on the audio wire. It develops a DC potential over the line impedance which is sensed by each intercom station and interpreted as a Call signal.

## 10.0 CABLING

The intercom lines (the 'party lines') are of the shielded two-conductor microphone cable type. The intercom line connectors are of the XLR-3 type. Audio and Call signals are on pin 3. DC power is on pin 2. Pin 1 is connected to the shield of the cable which functions as the common return for audio and power.

The audio signal is transferred in an unbalanced way (see Party Line, Technical Concept). To avoid earth loops (hum), the possible effect of electromagnetic fields and to minimize power loss, certain rules have to be obeyed when installing the cabling of an intercom system:

### **Use high quality cable**

Use high quality microphone cable only (shielded two conductor cable, minimum 2x 0.30 mm<sup>2</sup>). In case multi-pair cable is used, there should be an overall shield and each pair should consist of two conductors (minimum 2x 0.15 mm<sup>2</sup>) with separate shield.

### **Use flexible cable**

Use flexible single and multi-pair microphone cable instead of cable with solid cores, especially when the cable is subjected to bending during operation or installation.

### **Cable screens to XLR pin 1**

The screen of each separate microphone cable and/or the screen of each single pair in a multi-pair cable, should be connected to pin 1 of each XLR-3 connector. Do not connect these screens to the metal housing of ASL units or XLR-3 wall boxes). See Earthing Concept.

### **Connect cable trunks, connection boxes and overall multi-pair cable screens to clean earth**

Metal cable trunks, metal connection boxes and overall multi-pair cable screens should be interconnected and, at one point (the 'central earth point') in the intercom network only, be connected to a clean earth or a safety earth (see Earthing Concept).

### **Keep metal connection boxes and trunks isolated from other metal parts**

Metal trunks or pipes for intercom cables and metal connection boxes should be mounted in such a way that they are isolated from any other metal housing or construction parts.

### **Keep cables parallel as much as possible**

When two (multi channel) units in a network are connected by more than one cable, make sure that these cables are parallel to each other over the whole distance between those units. When using multi-pair cable, parallelism is ensured in the best possible way.

### **Avoid closed loops**

Always avoid that intercom cables are making a loop. So-called 'ring intercom' should not physically be cabled as a ring.

### **Keep cables away from electromagnetic sources**

Keep intercom cables away from high energy cables, e.g. 115/230/400V mains power or dimmer controlled feeds for spotlights. Intercom cables should cross high energy cables at an angle of 90° only. Intercom cables should never be in the same trunks as energy cables.

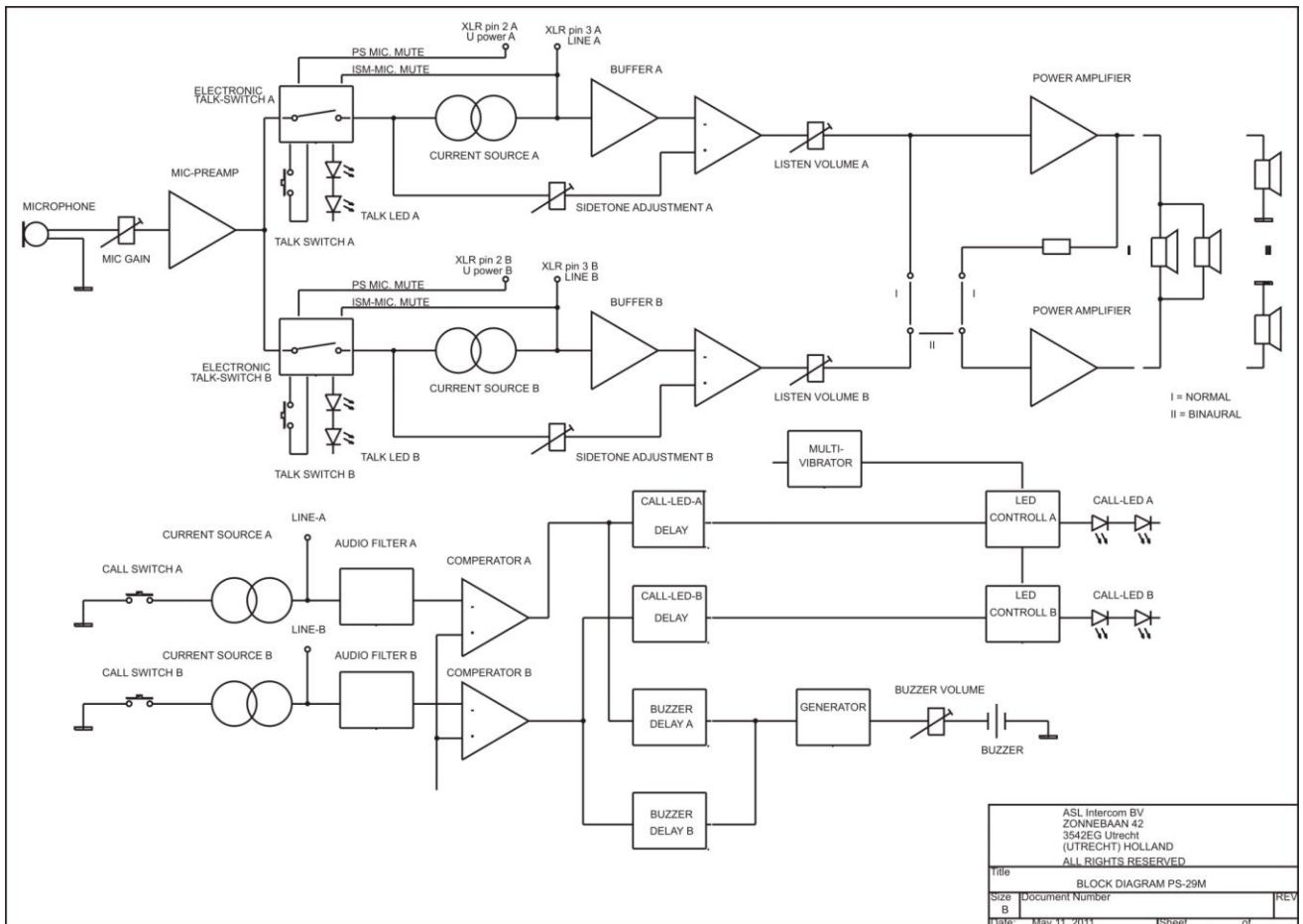
### **Place power supply in a central position**

In case of a system powered by a separate power supply: In order to diminish power losses, place the power supply as close as possible to where most power consumption occurs, in other words most user stations are placed.

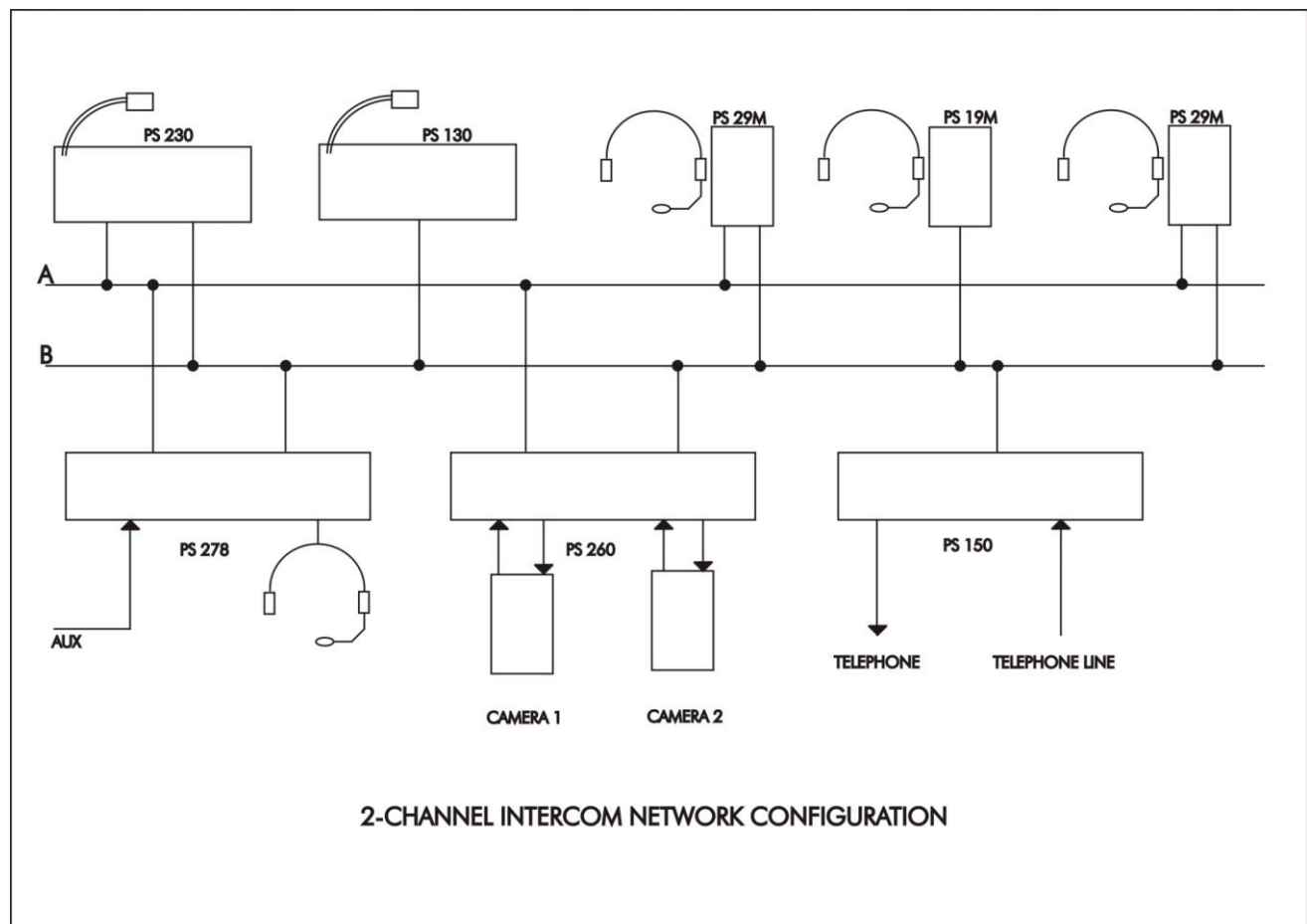
### **Connect ASL powered units to a 'clean' mains outlet**

Master stations or power supplies should be connected to the mains outlet with a clean earth. Other audio equipment may be connected to this mains outlet, but avoid using mains outlets which also power dimmer controlled lighting systems.

## 11.0 BLOCK DIAGRAM PS 29 M



## 12.0 SYSTEM CONFIGURATION



# 13.0 EARTHING CONCEPT

