**DATA SHEET** 

# **Key Features**

- Line/speech handsfree circuit, LD/MF repertory dialler, and tone ringer on one 44 pin CMOS chip
- Operating range from 15 to 100 mA line current (down to 5 mA with slightly reduced performance)
- All significant parameters programmable with external EEPROM
- Volume control of receiver signal
- Handsfree function with enhanced voice switching
- Low noise (max. -72 dBmp)
- Unique EMC performance
- LD/MF switchable dialing with temporary MF mode
- Repertory dialing with last number redial and memory dialing (8 x 2 direct and 12 indirect)
- Call restriction with PIN code
- Serial interface for EEPROM and LCD driver
- 2-tone/3-tone ringer with ring frequency detection

# Package

Available in 44 pin TQFP.

# **General Description**

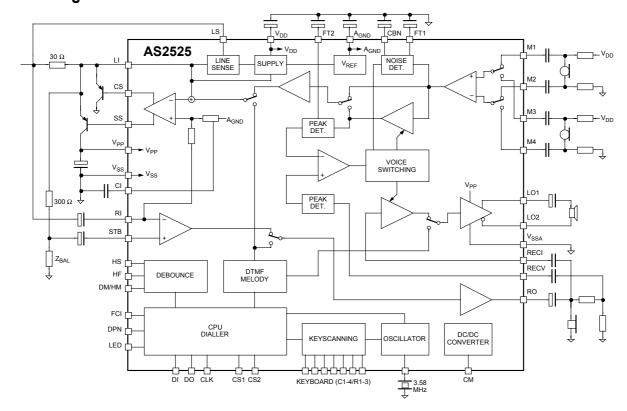
The AS2525 is a CMOS mixed-mode integrated circuit for use in feature phones, answering machines and fax machines. It contains an analogue line interface and speech circuit for a/b terminals, loudhearing, handsfree, enhanced LD/MF dialler, tone ringer with dc/dc converter and serial interface to EEPROM and LCD-driver (AS2591), all in a 44 pin package. The circuit is fully line powered.

The AS2525 uses an external EEPROM for a 32 digit last number redial storage and memories for 28 numbers each containing up to 24 digits/data.

The device provides a volume control for the earpiece and the loudspeaker. The volume can be controlled by pressing the [V-]/[V+] keys.

The versatility of the circuit is provided by programming all parameters through an external EEPROM. This allows easy adaptation to various PTT requirements worldwide.

# **Block Diagram**



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# Pin Description

Pin#	Symbol	Type	Description
1	CS2	DO	Chip Select 2
			Chip select signal for external LCD driver circuit or 2 mA LED
			If signal is not activated, a pull down resistor (100 $k\Omega$ ) is inserted.
2	DPn	DO	Dial Pulse Output
			Digital output that holds the line during off-hook operation or pulls low during break periods
			of pulse dialing and flash.
3	HF	DI	HandsFree Switch Input
			This is a Schmitt-trigger input that is pulled high to enable handsfree operation.
4	HS	DI	HandSet Switch Input
			This is a Schmitt-trigger input that is pulled high by the hook switch to enable handset or
			headset operation.
5	FT1	Al	Analogue input pin for connecting a capacitor for offset cancellation.
6	FT2	Al	Analogue input pin for connecting a capacitor for offset cancellation.
7	M4	Al	Microphone Input 4
			Differential input for the handsfree microphone (electret).
8	M2	Al	Microphone Input 2
			Differential input for the handset microphone (electret).
9	М3	Al	Microphone Input 3
			Differential input for the handsfree microphone (electret).
10	M1	Al	Microphone Input 1
			Differential input for the handset microphone (electret).
11	STB	Al	Side Tone Balance Input
			Analogue input for side tone cancellation network.
12	CS	AO	Current Shunt Control Output
			N-channel open drain output to control the external high power shunt transistor for
			synthesizing ac- and dc-impedance, modulation of line voltage and shorting the line during make periods of pulse dialing.
12	V	Commission	Voltage Source Source
13	V <sub>SS</sub>	Supply	Negative Power Supply.
14	LI	Al	Line Input
14	LI	AI	Analogue input used for power extraction and line current sensing.
1.5	RI	Al	Receive Input
15	KI	AI	Analogue input for ac-separated receive signal.
16	CC	40	Supply Source Control Output
16	SS	AO	N-channel open drain output to control the external high power source transistor for
			supplying (Vpp) the loudspeaker amplifier in off-hook loudspeaking/handsfree mode.
17	LS	Al	Line Current Sense Input
.,		/ 11	Analogue input for sensing the line current.
18	CI	AI	Complex Impedance Input
10		/ / /	Analogue input pin for the capacitor to program a complex impedance.

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Pin#	Symbol	Туре	Description			
19	RECI	Al	RECeive Input			
			Analogue input for the handsfree receive path. Should be connected to RO via a coupling capacitor.			
20	RECV	Al	RECeive Voice Switching Input			
			Analogue input for receive voice switching path.			
21	21 RO AO Receive Output to Handset					
			Output for driving a dynamic earpiece with an impedance from 150 $\Omega$ to 300 $\Omega$ .			
22	$V_{DD}$	Supply	Voltage Drain Drain			
			Positive Power Supply.			
23	AGND	Supply	Analogue Ground			
			Special ground for the internal amplifiers.			
24	CBN	Al	Analogue input pin for connecting a capacitor for background noise monitoring.			
25	CM	DO	Converter Make Output			
			This is an output for controlling the external switching converter. During ringing it converts			
			the ring signal into a 4V supply voltage.			
26	$V_{PP}$	Supply	Loudspeaker Power Supply High power supply for the output driver stage.			
27	LO1	AO	Loudspeaker Outputs 1/2			
28	LO2		Output pins for a 32 $\Omega$ (25 to 50 $\Omega$ ) loudspeaker.			
20	M	Cummlu				
29	Vssa	Supply	Power supply pin for LO1-LO2 output amplifier.			
30	DM/HM	HM DI	Dial Mode/Headset Mode Switch Input  Digital input for choosing between LD and MF dialing or choosing handset and headset			
			mode. See "Table 3: Parameter settings" for further details. This pin is scanned after manual off-hook only.			
31	FCI	DI	Frequency Comparator Input			
			This is a Schmitt-trigger input for ring frequency discrimination. Disabled during off-hook.			
32	LED	DO	LED Output			
			Digital output for connecting 2 mA LED to indicate handsfree mode.			
33	OSC		Oscillator Input			
			Input for ceramic resonator 3.58 MHz.			
34	R1	DI/O	Keyboard Rows			
35 36	R2 R3		(see Figure 1: Keyboard Layout).			
37	C1	DI/O	Keyboard Columns			
38	C2 C3		(see Figure 1: Keyboard Layout).			
39 40	C3					
41	CLK	DO	Clock			
			Clock output for 3-wire bus.			
			If signal is not activated, a pull down resistor (100 k $\Omega$ ) is inserted.			
42	DI	DI	Data Input			
			Data input of 3-wire bus.			
			If signal is not activated, a pull down resistor (100 k $\Omega$ ) is inserted.			

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Pin#	Symbol	Туре	Description			
43	DO	DO	Data Output			
			Data output of 3-wire bus.			
			A 100 k $\Omega$ resistor connected from this pin to VSS enables key locking.			
			If signal is not activated, a pull down resistor (100 k $\Omega$ ) is inserted.			
44	CS1	DO	Chip Select 1			
			Chip select signal for external EEPROM.			
			If signal is not activated, a pull down resistor (100 k $\Omega$ ) is inserted.			

DI: Digital Input AI: Analogue Input

DO: Digital Output AO: Analogue Output

DI/O: Digital Input/Output AI/O: Analogue Input/output

## **Basic Definitions**

Password is set to 0123 and is used to access basic parameter settings (see Table 3: Parameter settings). The set-up programming is normally done in the factory during manufacturing of the telephone sets. The factory settings can be loaded into the EEPROM (see Figure 3: Memory allocation for external EEPROM).

PIN code is default set to **1111** and is used by the user to protect certain user settings (see Table 5: User programming with PIN code). The PIN code can be changed by the user.

Default settings are shown in bold.

Program procedures are entered with the [PG] key.

Pressing any invalid key or going on-hook during
programming will abort the program mode and no changes
will be stored. Ending a program procedure by pressing the
[PG] key will store the changes and exit the program mode.

# **Keyboard Layout**

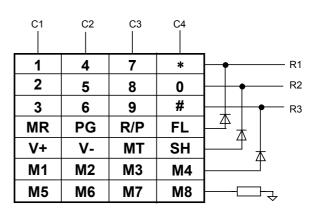


Figure 1: Keyboard Layout

Maximum RoN for key closure: 1 k $\Omega$  Minimum RoFF for open key: 1 M $\Omega$ 

See "Table 1: Key definitions" for description of key

functions

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# **Key Definitions and Procedures**

Table 1: Key definitions

Key	Condition	Function	Remark	
[0] - [9], [#], [*]	Off-hook	Digits	Function depending on mode	
[*]	LD mode	Temporary MF select	Providing mixed mode dialing (default LD selected)	
1001	Speech mode	Enter program state	To enter the program state	
[PG]	Program mode	Exit program state	To terminate/exit program state	
	Speech mode	Enter mute state	To enter mute state	
[MT]	Mute state	Exit mute state	To terminate/exit mute state	
	Program mode	Enter blank	To enter blank digit during programming	
[R/P]		Redial/Pause	Allows re-signaling of the last dialed number and insertion of pauses	
	Program state	Pause	Each key pressure inserts a pause	
[FL]		Flash function	Invokes a timed loop break	
[V+]	Speech mode	Volume control	To increase the receive gain	
[V-]	Speech mode	Volume control	To increase or decrease the receive gain	
[MR]		Memory redial key	To access memory by key codes	
[M1] - [M8]		Memory dial	Access keys to memories 1 to 8 and 9 to 16	
[SH]		Second function (shift)	To access second function of keys	

Table 2: Digits

Digit Key	DTMF Mode (fLow + fHIGH)	Pulse Mode	Programming Mode ([PG] + Digit)	Memory Location ([MR] + Digit)
[0]	941 + 1336 Hz	10 pulses		Memory 17
[1]	697 + 1209 Hz	1 pulse	To enter PIN protected programming	Memory 18
[2]	697 + 1336 Hz	2 pulses	Select flash duration	Memory 19
[3]	697 + 1477 Hz	3 pulses	Select dialing mode	Memory 20
[4]	770 + 1209 Hz	4 pulses	Select pause duration	Memory 21
[5]	770 + 1336 Hz	5 pulses	Key lock toggle	Memory 22
[6]	770 + 1477 Hz	6 pulses	Ringer volume	Memory 23
[7]	852 + 1209 Hz	7 pulses	Ringer melody	Memory 24
[8]	852 + 1336 Hz	8 pulses	LCD contrast	Memory 25
[9]	852 + 1477 Hz	9 pulses	Ringer on/off	Memory 26
[*]	941 + 1209 Hz	Temporary MF select		-
[#]	941 + 1477 Hz	-		-

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Table 3: Parameter settings

Parameter	Symbol	Procedure: [PG] + [0] + (Password) +	Default	Options
Inter-Digit Pause (LD)	IDP	[*] + [1] + [HEX code]	840 ms	1.72 ms 7.03 sec. (see Table 4)
MF Tone Duration	TD	[*] + [2] + [HEX code]	82 ms	1.72 ms 7.03 sec. (see Table 4)
MF Inter-Tone Pause	ITP	[*] + [3] + [HEX code]	82 ms	1.72 ms 7.03 sec. (see Table 4)
Flash 1 Duration	FL1	[*] + [4] + [HEX code]	101 ms	1.72 ms 7.03 sec. (see Table 4)
Flash 2 Duration	FL2	[*] + [5] + [HEX code]	285 ms	1.72 ms 7.03 sec. (see Table 4)
Pause 1 Duration	P1	[*] + [6] + [HEX code]	1 second	1.72 ms 7.03 sec. (see Table 4)
Pause 2 Duration	P2	[*] + [7] + [HEX code]	3 seconds	1.72 ms 7.03 sec. (see Table 4)
Pre-Digit Pause (LD)	PDP	[*] + [8] + [HEX code]	33 ms	1.72 ms 7.03 sec. (see Table 4)
Post Flash Pause	PFP	[*] + [9] + [HEX code]	274 ms	1.72 ms 7.03 sec. (see Table 4)
Tx Gain, Handset	A <sub>TX-HS</sub>	[0] + [1] + [HEX code]	37 dB	30 to 45 dB (see Table 8)
Rx Gain, Handset	A <sub>RX-HS</sub>	[0] + [2] + [HEX code]	1 dB	-6 to 9 dB (see Table 12)
Tx Gain, Handsfree	Атх-нғ	[0] + [3] + [HEX code]	46 dB	39 to 54 dB (see Table 10)
Rx Gain, Loudspeaker	A <sub>RX-HS</sub>	[0] + [4] + [HEX code]	29 dB	22 to 37 dB (see Table 14)
Line Loss Compensation	LLC	[0] + [5] +	[0] = off	[1] = range 1: I <sub>LINE</sub> = 20 - 50 mA [2] = range 2: I <sub>LINE</sub> = 45 - 75 mA
DTMF Level (Low Group)	V <sub>M</sub> F	[0] + [6] + [HEX code]	-6 dBm	-18 to -6 dBm in 1 dB steps (see Table 24)
Tx Gain, Headset	A <sub>TX-HEAD</sub>	[0] + [7] + [HEX code]	37 dB	30 to 45 dB (see Table 20)
Rx Gain, Headset	Arx-HEAD	[0] + [8] + [HEX code]	1 dB	-6 to 9 dB (see <b>Table 21</b> )
MF Comfort Tone, RO	V <sub>MF-CT/RO</sub>	[0] + [9] + [HEX code]	-30 dBr	-36 to -18 dBr in 6 dB steps (see Table 30)
MF Comfort Tone, LO1/2	V <sub>MF-CT/LO</sub>	[0] + [0] + [HEX code]	-9 dBr	-15 to 3 dBr in 6 dB steps (see Table 30)
Make/Break Ratio (LD)	M/B	[#] + [1] +	[1] = 2 : 3	[0] = 1 : 2
Dial Rate (LD)	DR	[#] + [2] +	[0] = 10 pps	[1] = 20 pps
DC Mask (LI Voltage)	V <sub>LI</sub>	[#] + [3] +	[1] = 4.5 Volt	[0] = 3.5 Volt
Tx Soft Clip (Handset)		[#] + [4] +	[1] = on (2 V <sub>PEAK</sub> )	[0] = off
Rx Soft Clip (Handset)		[#] + [5] +	[1] = on (1 V <sub>PEAK</sub> )	[0] = off
Noise Monitoring (HF)		[#] + [6] +	[1] = on	[0] = off
Loudhearing Mode		[#] + [7] +	[1] = HF only	[0] = LH and HF modes available
Ringer Melody		[#] + [8] +	[1] = 3-Tone	[0] = 2-Tone
Key-in Tone (Beep)	КТ	[#] + [9] +	[1] = on	[0] = off
Reset PIN Code		[#] + [0] +	[1] = 1111	Selected by user

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Parameter	Symbol	Procedure: [PG] + [0] + (Password) +	Default	Options
Flash 1 During LD		[#] + [*] +	[1] = on	[0] = off
Flash 2 During LD		[#] + [#] +	[0] = off	[1] = on
Extended LCD Symbols		[9] + [1] +	[0] = off	[1] = on
Tx Soft Clip, Handsfree		[9] + [2] +	[1] = on	[0] = off
Rx Soft Clip, Handsfree		[9] + [3] +	[1] = on	[0] = off
Voice Switching, Speed		[9] + [4] +	[1] = x 2	[0]=x1 (slow); [2]=x4; [3]=x8 (fast)
Handset/Headset Volume		[9] + [5] +	[1] = reset by on/off-hook	[0] = setting is maintained
LCD Test		[9] + [6]		gments, pressing any further key turns and all symbols off.
Temporary MF by * with tone		[9] + [7] +	[0] = off	[1] = on
Key Lock Function		[9] + [8] +	[1] = available	[0] = not available
Pin Select "DM/HM"	DM/HM	[9] + [9] +	[1] = MF/LD	[0] = Handset/Headset
Ringer Off Function		[9] + [0] +	[1] = available	[0] = not available
Key Test		[9] + [*] +	Any key pressu	re turns the LED on
Volume Control on Handset		[9] + [#] +	[1] = enabled	[0] = disabled

The above parameters are programmable using a password (0123) through external EEPROM. All procedures must be terminated by pressing the [PG] key. Software for factory settings is available on request.

The programming of timings is done by using a HEX data code as follows:

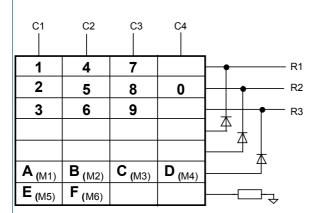


Figure 2: Keys for HEX-code entry

Data = 4096 - (fosc / (512 \* 12) \* T (T = Time; fosc = 3.58 MHz)

Table 4: Timing selection by HEX code

Parameter	Default	HEX	Binary Data
Inter-digit pause	840 ms	E17	1111 0001 0111
Tone duration	82 ms	FD0	1111 1101 0000
Inter-tone pause	82 ms	FD0	1111 1101 0000
Flash 1	101 ms	FC4	1111 1100 0100
Flash 2	285 ms	F5A	1111 0101 1010
Pause 1	1 sec	DB9	1101 1011 1001
Pause 2	3 sec	92C	1001 0010 1100
Pre-digit pause	33 ms	FED	1111 1110 1101
Post-flash pause	274	F60	1111 0110 0000

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Table 5: User programming with PIN code

Function	Symbol	Procedure	Options		
Call Restriction 1	CR1	[PG] + [1] + (PIN) + [*] + [1] +	Key in two digits, use [MT] for <b>blank</b>		
Call Restriction 2	CR2	[PG] + [1] + (PIN) + [*] + [2] +	Key in two digits, use [MT] for blank		
Key Lock		[PG] + [1] + (PIN) + [#] + [6] +	[0] = unlocked and [1] = locked		
Clear All Memories		[PG] + [1] + (PIN) + [#] + [7] +	Press [1] to clear		
Reset To Factory Settings		[PG] + [1] + (PIN) + [#] + [8] +	Press [1] to reset		
Change PIN Code	PIN	[PG] + [1] + (PIN) + [0] +	Key in new PIN + new PIN		
Press [PG] to store setting and terminate procedure					

Table 6: Mode programming

Function	Symbol	Procedure	Options	
Select Flash	FL	[PG] + [2] +	[0] = <b>Flash 1</b> and [1] = Flash 2	
Select Dialing Mode <sup>1</sup>	DM	[PG] + [3] +	[0] = <b>LD</b> (pulse) and [1] = MF (DTMF)	
Select Pause Time	Р	[PG] + [4] +	[0] = <b>Pause 1</b> and [1] = Pause 2	
Easy Key Lock <sup>2</sup>		[PG] + [5] + [3] + [9]	Toggles between locked and unlocked	
Ringer Volume		[PG] + [6] +	[1] = -16 dB, [2] = -7 dB, and [3] = <b>0 dB (maximum)</b>	
Ringer Melody		[PG] + [7] +	[0], [1], [2], [3], or [4]	
LCD Contrast	V <sub>33</sub>	[PG] + [8] +	[1], [2], or [3] (see also AS2591 data sheet)	
Bell On/Off		[PG] + [9] +	[0] = Off and [1] = <b>On</b>	
Press [PG] to store setting and terminate procedure				

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<sup>&</sup>lt;sup>1</sup> Only when DM/HM pin is set to HM

<sup>&</sup>lt;sup>2</sup> Also selectable with jumper (see application note AN525)

## **Memory Allocation**

The EEPROM memory consists of three areas:
A = factory settings; B = user settings; C = stored numbers

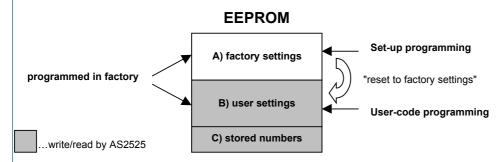


Figure 3: Memory allocation for external EEPROM

The AS2525 is writing and reading to and from area B and C during normal operation and user-code programming. Area A is protected during normal operation and contains a back-up copy of the factory settings. This back-up copy can be loaded into area B by the user with the user-code programming procedure "reset to factory settings". The only way to change the factory settings in area A is by set-up programming using password. After changing area A by set-up programming, the AS2525 automatically performs a "reset to factory settings" to load this changes also into area B.

# **Functional Description**

#### DC Conditions

The normal operating mode is from 15 mA to 100 mA. An operating mode with reduced performance is from 5 mA to 15 mA. In the line-hold range from 0 mA to 5 mA the device is in a power down mode.

The dc characteristic is determined by the voltage at LI pin and a 30  $\Omega$  resistor between pin LI and LS. It can be calculated by the following equation:

$$V_{LS} = V_{LI} + I_{LINE} * 30 \Omega$$

VLI can be programmed to be 3.5 V or 4.5 V.

Table 7: DC mask selection

[PG] + [0] + (Password) + [#] + [3] + (digit) + [PG]					
Digit	DC Mask (dc voltage at pin LI)				
[0]	Low dc mask, VLI = 3.5 Volt				
[1]	High dc mask, VLI = 4.5 Volt				

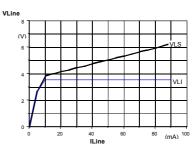


Figure 4: DC characteristic with low dc mask

#### 2/4 Wire Conversion

AS2525 has a built-in dual Wheatstone bridge with one common ground. This provides a maximum of independence of ac impedance and side-tone from each other. One can adapt side-tone without changing the ac impedance.

## **AC Impedance**

The ac impedance of AS2525 is set to 1000  $\Omega$ . A complex impedance can be adjusted by connecting an external capacitor to the CI pin. With an external resistor of approximately 1.5 k $\Omega$  connected to the LS pin, it can be programmed to 600  $\Omega$ .

$$Z_{AC(syn)} = 33 * 30 \Omega$$

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#### Side-Tone

A good side-tone cancellation can be achieved by using the following equation:

$$Z_{BAL}/Z_{LINE} = 10$$

This is assuming that the side-tone reference resistor, R2 (between LI and STB) is 300  $\Omega$  and R1 is 30  $\Omega$ .

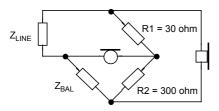


Figure 5: Side-tone balance

#### Transmit Path

The gain of the M1/M2 —> LS path is default set to 37 dB. This gain can be changed from 30 dB to 45 dB in 1 dB steps by set-up programming.

Table 8: Handset Tx gain setting

[PG] + [0] + (Password) + [0] + [1] + (x) + [PG]						
Key (x)	Tx Gain (dB) M1/2 to LS	Key (x)	Tx Gain (dB) M1/2 to LS			
[0]	30	[8]	38			
[1]	31	[9]	39			
[2)	32	[M1]	40			
[3]	33	[M2]	41			
[4]	34	[M3]	42			
[5]	35	[M4]	43			
[6]	36	[M5]	44			
[7]	37	[M6]	45			

The input is differential with an impedance of 10 k $\Omega$ . The soft clip circuit limits the output voltage swing at pin LS to 2 V<sub>PEAK</sub> when activated.

Table 9: Handset Tx soft clipping selection

[PG] + [0] + (Password) + [#] + [4] + (digit) + [PG]				
Digit	Handset Tx Soft Clipping			
[0]	Disabled			
[1]	Enabled (2 VPEAK)			

There is optional LLC for this path.

The gain of the M3/M4 —> LS path is default set to 46 dB. This gain can be changed from 39 dB to 54 dB in 1 dB steps by set-up programming.

Table 10: Handsfree Tx gain setting

[PG]	[PG] + [0] + (Password) + [0] + [3] + (x) + [PG]					
Key (x)	Tx Gain (dB) M3/4 to LS	Key (x)	Tx Gain (dB) M3/4 to LS			
[0]	39	[8]	47			
[1]	40	[9]	48			
[2]	41	[M1]	49			
[3]	42	[M2]	50			
[4]	43	[M3]	51			
[5]	44	[M4]	52			
[6]	45	[M5]	53			
[7]	46	[M6]	54			

The input is differential with an impedance of 10 k $\Omega$ .

The soft clip circuit limits the output voltage swing at pin LS to 2  $V_{\text{PEAK}}$  when activated.

Table 11: Handsfree Tx soft clipping selection

[PG] + [0] + (Password) + [9] + [2] + (digit) + [PG]					
Digit	Handsfree Tx Soft Clipping				
[0]	Disabled				
[1]	Enabled (2 V <sub>PEAK</sub> )				

There is no LLC option for this path.

### **Receive Path**

The gain of the LS —> RO receive path is default set to 1 dB. This gain can be changed from -6 dB to 9 dB in 1 dB steps by set-up programming.

Table 12: Handset Rx gain setting

[PG] + [0] + (Password) + [0] + [2] + (x) + [PG]						
Key (x)	Rx Gain (dB) LS to RO	Key (x)	Rx Gain (dB) LS to RO			
[0]	-6	[8]	2			
[1]	-5	[9]	3			
[2]	-4	[M1]	4			
[3]	-3	[M2]	5			
[4]	-2	[M3]	6			
[5]	-1	[M4]	7			
[6]	0	[M5]	8			
[7]	1	[M6]	9			

When the handset volume control is enabled (see Table 3: Parameter settings), the user can also change the gain within this range in 4 dB steps via the [V+]/[V-] keys. The default receive gain in handset and headset mode must lie between position 5 and 9 for correct operation of the volume control. The receive input is the differential signal of RI and STB. The soft clip circuit limits the output voltage at RO to 1 VPEAK. It prevents harsh distortion and acoustic shock.

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Table 13: Handset Rx soft clipping selection

[PG] + [0] + (Password) + [#] + [5] + (digit) + [PG]				
Digit	Handset Rx Soft Clipping (Earpiece)			
[0]	Disabled			
[1]	Enabled			

There is LLC option for this path.

The gain of the LS —> LO1/LO2 receive path is set to 29 dB.

This gain can be changed by set-up programming from 22 dB to -37 dB in 1 dB steps.

Table 14: Loudspeaker gain setting

[PG] + [0] + (Password) + [0] + [4] + (x) + [PG]					
Key (x)			Rx Gain (dB) LS to LO1/2		
[0]	22	[8]	30		
[1]	23	[9]	31		
[2]	24	[M1]	32		
[3]	25	[M2]	33		
[4]	26	[M3]	34		
[5]	27	[M4]	35		
[6]	28	[M5]	36		
[7]	29	[M6]	37		

The user can also change the gain using the [V+]/[V-] keys (see section "Handsfree"). The volume range is 20 dB (8 steps, each 2.5 dB).

The receive input is the differential signal of RI and STB. The soft clip circuit limits the output voltage swing at LO1/LO2 to 1 VPEAK.

Table 15: Loudspeaker soft clipping selection

[PG] + [0] + (Password) + [9] + [3] + (digit) + [PG]					
Digit	Handsfree Rx Soft Clipping (Loudspeaker)				
[0]	Disabled				
[1]	Enabled				

There is optional LLC for this path.

## Line Loss Compensation (LLC)

The line loss compensation is set-up programmable and effects only the handset mode.

Table 16: Line loss compensation selection

[PG] + [0] + (Password) + [0] + [5] + (digit) + [PG]				
Digit	LLC Range			
[0]	No LLC (gain independent of line current)			
[1]	-1 dB/5 mA from 20 to 50 mA			
[2]	-1 dB/5 mA from 45 to 75 mA			

When it is activated, the transmit and receive gains for both channels are decreased by 6 dB at line currents from 20 mA to 50 mA or from 45 mA to 75 mA depending on selected range.

#### Handsfree

The handsfree function allows voice communication without using the handset (full 2-way speaker phone). Two voice controlled attenuators prevent acoustic coupling between the loudspeaker and the handsfree microphone.

The voice switching circuit has three states, namely idle, transmit or receive. In receive mode the attenuation of the receive path and the transmit path can be controlled by the volume keys between 0 dB and -20 dB. The following table (Table 18: Handsfree attenuation scheme) shows how voice switching is controlled.

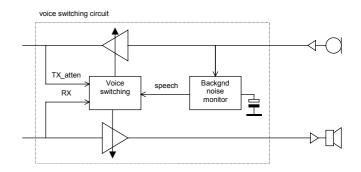


Figure 6: Handsfree voice switching

A background monitoring circuit is incorporated. This circuit can be enabled/disabled as required.

Table 17: Background noise monitoring selection

[PG] + [0] + (Password) + [#] + [6] + (digit) + [PG]				
Digit	Digit Background Noise Monitoring			
[0]	Disabled			
[1]	[1] Enabled			

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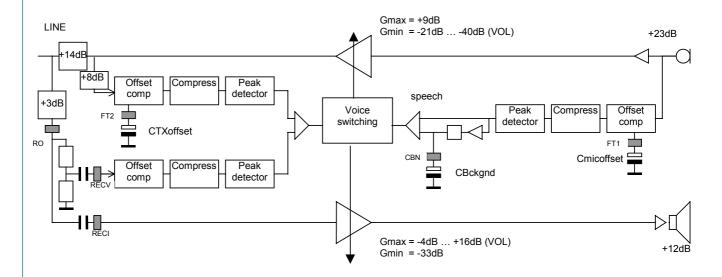


Figure 7: Signal path of handsfree circuit

Table 18: Handsfree attenuation scheme

	Speech	Mode	RX-gain	TX-gain	Remark
RX > TX_atten	Х	Receive	0 dB to -20 dB	-50 dB to -30 dB	adjustable with [V+]/[V-] keys in 8 steps
TX_atten > RX	No	ldle	-25 dB	-25 dB	middle position
TX_atten > RX	Yes	Transmit	-50 dB	0 dB	independent of [V+]/[V-] keys

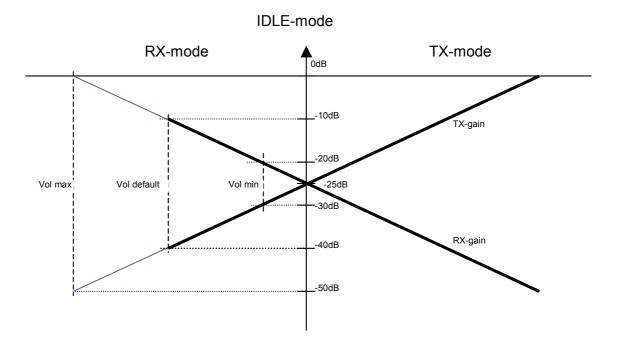


Figure 8: Gain transition of voice switching

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#### Loudhearing

A loudhearing mode can be enabled/disabled through the set-up programming.

Table 19: Loudhearing mode selection

[PG] + [0] + (Password) + [#] + [7] + (digit) + [PG]				
Digit	Loudhearing			
[0]	Enabled (LH + HF provided)			
[1]	Disabled (only HF is provided)			

## Headset mode

Head set mode is selected by connecting pin "HEAD" to VDD and enabling the head function by service code programming (see also Table 3: Parameter settings).

Table 20: Headset Tx gain setting

[PG] + [0] + (Password) + [0] + [7] + (x) + [PG]						
Key (x)	Tx Gain (dB) M1/2 to LS	Key (x)	Tx Gain (dB) M1/2 to LS			
[0]	30	[8]	38			
[1]	31	[9]	39			
[2]	32	[M1]	40			
[3]	33	[M2]	41			
[4]	34	[M3]	42			
[5]	35	[M4]	43			
[6]	36	[M5]	44			
[7]	37	[M6]	45			

Table 21: Headset Rx gain setting

[PG] + [0] + (Password) + [0] + [8] + (x) + [PG]				
Key (x)	Rx Gain (dB) LS to RO	Key (x)	Rx Gain (dB) LS to RO	
[0]	-6	[8]	2	
[1]	-5	[9]	3	
[2]	-4	[M1]	4	
[3]	-3	[M2]	5	
[4]	-2	[M3]	6	
[5]	-1	[M4]	7	
[6]	0	[M5]	8	
[7]	1	[M6]	9	

If this mode is entered both hook inputs HS and HF act as electronic off-hook for handset operation. The gain in transmit and receive can be programmed by service code programming. The pin "HEAD" is only scanned once after off-hook and any change on this pin during one session will be ignored.

# **Ring Frequency Detection**

The ring frequency detector assures that only signals with a frequency between 13 Hz and 70 Hz are regarded as valid ring signals.

When a valid ring signal is present for 100 ms continuously, the melody generator is activated and remains active as long as the ring signal is present.

Once the melody generator has been started, the ring signal is continuously monitored and the melody generator is instantly turned on or off according to the momentary presence of a valid or invalid ring signal respectively (until next POR or off-hook).

## DC/DC converter

When a valid ring signal is detected, the dc/dc converter is started.

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# **Dialing and Control Functions**

#### Valid Keys

The key scanning is enabled in off-hook mode with  $V_{DD}$  above  $V_{REF}$ . A valid key is detected from the keyboard by connecting the appropriate row to the column ( $R_{ON} < 1$  k $\Omega$ ). This can be done using an n x m keyboard matrix with single contacts. Three diodes and a resistor are used to extend the number of rows (see Figure 1: Keyboard Layout).

#### PIN Code

A PIN code is available for protecting various user settings. The default PIN code is set to 1111 and can be change as follows:

- 1. Press [PG] to enter program mode
- 2. Press [1] + (PIN) + [0]
- 3. Key in new PIN code (valid digits: 0 9, #)
- 4. Key in new PIN code again
- 5. Press [PG] (= store and exit) or go on-hook to abort

It is also possible to reset the PIN code through set-up programming:

- 1. Press [PG] to enter program mode
- 2. Press [0] + (Password) + [#] + [0]
- 3. Press [1] to reset
- Press [PG] (= store and exit) or go on-hook to abort

#### **Call Restriction**

A call restriction function is accommodated. This function can be activated using the PIN code. One or two digits can be programmed to inhibit dialing starting with said digit(s). If first digit(s) of a manual entered number or a number retrieved from a memory matches the content of the call restriction memory, dialing will be inhibited.

Storing digits for call restriction:

- 1. Press [PG] to enter program mode
- 2. Press [1] + (PIN) + [\*]
- 3. Press [1] for entering first digit of call restriction or [2] for entering second digit
- 4. Key in digit for call restriction
- 5. Press [PG] (= store and exit) or go on-hook to abort.

#### Example 1:

Prohibit all calls with numbers starting with 0.

Activate: [PG] + [1] + (PIN) + [1] + [0] + [PG]

Deactivate: [PG] + [1] + (PIN) + [1] + [MT] + [PG]

#### Example 2:

Prohibit all calls with numbers starting with **09** (numbers starting with 00..., 01..., etc. will be allowed; pauses will be ignored, i.e. numbers starting with 0 PS 9 ... will be prohibited).

Activate: [PG] + [1] + (PIN) + [1] + [0] + [PG] + [PG] + [2] + (PIN) + [1] + [9] + [PG]

Deactivate: [PG] + [1] + (PIN) + [1] + [MT] + [PG] + [PG] + [2] + (PIN) + [1] + [MT] + [PG]

Pressing the [MT] key in place of the digit during programming will cancel said call restriction. The first and second digit can be stored, changed or deleted independent of each other.

## **Key Lock**

A key lock function is provided to allow the user to protect the telephone from any misuse. When this function is available through the set-up programming, it is possible to enable/disable the key lock using the user PIN code.

Enabling/disabling key lock:

- 1. Press [PG] to enter program mode
- 2. Press [1] +(PIN) + [#] + [6]
- 3. Press [1] for enable or [0] for disable
- Press [PG] (= store and exit) or go on-hook to abort

Also an easy key lock function is provided. This function can be activated/deactivated when the key lock function is available through set-up programming and the previous mentioned key lock through PIN code is disabled.

### **Clear Memory Procedure**

It is possible to clear all stored numbers by using the PIN code as follows:

- 1. Press [PG] to enter program mode
- 2. Press [1] + (PIN) + [#] + [7]
- 3. Press [1] for clear

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Press [PG] (= store and exit) or go on-hook to abort

### Reset to Factory Settings

It is possible to reset all settings to the default factory settings with following procedure:

- 1. Press [PG] to enter program mode
- 2. Press [1] + (PIN) + [#] + [8]
- 3. Press [1] for reset
- Press [PG] (= store and exit) or go on-hook to abort

#### Dial Mode Selection

The default signaling mode (LD or MF) is selectable through following procedure:

- 1. Press [PG] to enter program mode
- 2. Press [3] to enter dial select mode
- 3. Press [0] for LD and [1] for MF
- Press [PG] (= store and exit) or go on-hook to abort

When default LD mode is selected, a temporary change to MF can be invoked by pressing the [\*] key. Temporary MF select with tone on first key pressure can be selected as option (see also Table 3: Parameter settings). The circuit will revert to LD by pressing the [FL] key or by next onhook.

## Re-Dialing (Last Number Redial)

Re-dialing is a facility that allows re-signaling of the last dialed number without keying in all the digits again.

Numbers dialed out from any memory will also be stored in the LNR memory. The re-dialing works in following manner:

- Any digit (including Flash, Pause, and MF select) being dialed is storable in the LNR memory up to a total of 31 digits.
- If more than 31 digits are entered, re-dialing will be inhibited.
- Pressing the [R/P] key as first entry after off-hook (LNR cursor at first digit) will invoke dialing the content of the LNR memory.
- If the [R/P] key is pressed after entering one or more digits, the key will function as a pause key and each pressure will insert a pause.

Pauses can also be inserted by pressing the [R/P] key in program mode.

#### Flash Function

A flash, **[FL]** key, activation will invoke a timed loop break. The flash duration can be programmed through set-up (see Figure 2: Keys for HEX-code entry and Table 4: Timing selection by HEX code):

Table 22: Flash duration setting

[PG] + [0] + (Password) + [*] + (digit) + (HEX) + [PG]			
Digit	Timing	Flash Duration	
[4]	HEX code	Flash 1	
[5]	HEX code	Flash 2	

One of two pre-programmed can be selected as follows:

- 1. Press [PG] to enter program mode
- 2. Press [2] to enter flash select mode
- 3. Press [0] for duration 1 and [1] for duration 2
- 4. Press [PG] (= store and exit) or go on-hook to abort

A flash entered will be stored in the LNR memory together with entered digits. After a recall a 270 ms pause will automatically be executed.

#### **Pauses**

Pauses can be inserted if the [R/P] key is pressed after entering one or more digits. Also during programming, pauses can be inserted by pressing the [R/P] key. Default pause duration is 1 second (P1) and an alternative pause of 3 second (P2) can be selected. These pauses can be changed through the set-up programming (see Table 3: Parameter settings)

## **Memory Keys**

The keys [M1] to [M8] are direct memory access keys, [SH] key is used to access second level of direct memories, and the [MR] key is used for key-code dialing. Up to 26 numbers can be stored in the on-chip RAM. Each number can contain up to 21 digits (including pauses). During programming multiple pauses can be inserted by pressing the [RD] key.

Example (45678P123 is stored in M1, where P is a pause and 123 the extension number):

- 1. Off-hook, wait for dial tone
- 2. Press [M1]
  - 45678 is dialed out
  - Dialing halts (pauses)
  - Dialing is resumed, 123 is dialed out.

Memory dialing is cascadable up to a the maximum of 31 digits/data.

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## **Memory Storing Procedure**

Up to 28 numbers, each with maximum 24 digits, can be stored into the external EEPROM. The store procedure is as follows:

- 1. Press [PG] to enter program mode
- 2. Enter location (MR + digit<sup>3</sup>; M1 to 8; or SH + M1 to 8)
- 3. Enter number
- 4. Press [PG] to store and exit or go on-hook to abort
- 5. Go to 1 for storing further numbers

[Mute] starts and stops security storing

(LCD-Symbols [ and ])

[R/P] inserts a pause (LCD-Symbol P)

## **Tone Generator**

The tone generator incorporates the DTMF tones, 3 frequencies for the tone ringer and pacifier tones.

#### DTMF

The DTMF generator provides 7 frequencies.

Table 23: DTMF frequencies

Digit	Frequency
Rows	Low Group
1 - 2 - 3	697 Hz
4 - 5 - 6	770 Hz
7 - 8 - 9	852 Hz
* - 0 - #	941 Hz
Columns	High Group
1 - 4 - 7 - *	1209 Hz
2 - 5 - 8 - 0	1336 Hz
3 - 6 - 9 - #	1477 Hz

The MF output level can be set through set-up programming. The range is from -18 dBm to -6 dBm in 1 dB steps.

Table 24: DTMF level settings

S Contract of the contract of				
[PG] + [0] + (Password) + [0] + [6] + (x) + [PG]				
Key (x)	Level (dBm) Low Group	Key (x)	Level (dBm) Low Group	
[0]	-18	[7]	-11	
[1]	-17	[8]	-10	
[2]	-16	[9]	-9	
[3]	-15	[M1]	-8	
[4]	-14	[M2]	-7	
[5]	-13	[M3]	-6	
[6]	-12			

 $<sup>^3</sup>$  Digit includes 0 – 9, \* and #

The pre-emphasis is 2.6 dB. The DTMF tones are according to CEPT recommendations.

## Tone Ringer (Melody/Volume)

The tone ringer incorporates two basic selection, namely 2-tone or 3-tone melodies. The default set-up selection is the 3-tone ringer.

Table 25: Basic ringer melody selection

[PG] + [0] + (Password) + [#] + [8] + (digit) + [PG]		
Digit	Ringer Melody	
[0]	2-tone	
[1]	3-tone	

The user can chose the following melodies as shown in below table.

Table 26: 3-tone melody programming

Procedure	Digit	Sequence	Rate
[PG] + [7] +	[0]	F1 F2 F3 F1 F2 F3	1
	[1]	F1 F2 F3 F1 F2 F3	2
	[2]	F1 F2 F3 F1 F2 F3	4
	[3)	F1 F2 F3 F1 F2 F3	6
	[4]	F1 F2 F3 F1 F2 F3	10
Press [PG] to store and terminate			

The chosen melody will be heard in the loudspeaker during programming. Pressing the [PG] key will end the procedure and the last selected melody will be stored.

When 2-tone melody is selected through the set-up procedure, it is possible to chose the following melodies:

Table 27: 2-tone melody programming

Procedure	Digit	Sequence	Rate	
[PG] + [7] +	[0]	F1 F2 F1 F2 F1 F2	2	
	[1]	F1 F2 F1 F2 F1 F2	5	
	[2]	F1 F3 F1 F3 F1 F3	2	
[3) F1 F3 F1 F3 F1 F3 5			5	
[4] F2 F3 F2 F3 F2 F3 2				
Pr	Press [PG] to store and terminate			

Repetition rate means that a sequence of 6 frequencies is repeated 1, 2, 4, 5, 6 or 10 times within 1 second.

## **Pacifier Tone**

Any key entry is acknowledged by a pacifier tone (key-in tone) of 1333 Hz. The pacifier tone can be enabled/disabled through set-up programming.

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Table 28: Pacifier tone selection

[PG] + [0] + (Password) + [#] + [9] + (digit) + [PG]		
Digit	Pacifier Tone (Key-in Tone)	
[0]	Disabled	
[1]	Enabled	

Table 29: Pacifier tones

Key Entry	Frequency	Duration
Acknowledge	1333 Hz	40 ms
Reject/Terminate	1333 Hz	4 x 40 ms

An invalid key entry will be indicated by a rejection tone of 4 times 40 ms.

By MF dialing the DTMF tones are provided to the earpiece as comfort tone. The level of the comfort is selected through the set-up programming.

Table 30: DTMF comfort tone level setting

[PG] + [0] + (Password) + [0] + (y + x) + [PG]				
Key (y + x)	Level (dBr) Earpiece	Key (y + x)	Level (dBr) Loudspeaker	
[9] + [0]	-36	[0] + [0]	-15	
[9] + [1]	-30	[0] + [1]	-9	
[9] + [2]	-24	[0] + [2]	-3	
[9] + [3]	-18	[0] + [3]	3	

#### Serial Bus

The AS2525 incorporates a microwire<sup>™</sup> compatible serial bus (DO, DI, and CLK) with two chip select outputs (CS1 and CS2). The serial bus is used to:

- sending data to the LCD driver (AS2591) when a key is pressed or a number is dialed out
- reading and writing to the external EEPROM under following conditions:

Table 31: EEPROM read/write timings

ACTION	DURATION
After off-hook (debounced)	50 ms
After on-hook (debounced)	200 ms
After first rising edge at FCI	10 ms
After PG is terminated	12600 ms
After pressing [V+]/[V-]	20 ms

See also AS2591 data sheet for further information.

## **LCD Control**

The AS2525 is designed to work together with the LCD driver AS2591 (see also AS2591 data sheet).

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# **Electrical Characteristics**

## Absolute maximum ratings

$-0.3 \le V_{DD} \le 7V$
+/- 25 mA
$-0.3V \le V_{IN} \le 12V$
$-0.3V \le V_{IN} \le 8V$
$-2V \le V_{IN} \le V_{DD} + 0.3V$
$-0.3V \le V_{IN} \le V_{DD} + 0.3V$
+/- 1000V
-65 to +125°C
500mW
·

<sup>\*</sup> Exceeding these figures may cause permanent damage. Functional operation under these conditions is not permitted.

## Recommended Operating Range

Symbol	Parameter	Conditions	Min.	Typ.*	Max.	Units
V <sub>DD</sub>	Supply Voltage (internally generated)	Speech mode	3.0		5.0	V
T <sub>AMB</sub>	Ambient Operating Temp. Range		-25		+70	°C
fosc	Oscillator Frequency	Resonator: Murata CSA 3.58M G312AM		3.58		MHz

<sup>\*</sup> Typical figures are at 25°C and are for design aid only; not guaranteed and not subject to production testing.

## DC Characteristics

(ILINE = 15 mA, recommended operating conditions unless otherwise specified)

Symbol	Parameter Conditions		Min.	Тур.	Max.	Units
I <sub>DD</sub>	Operating Supply Current	Speech mode		4	6	mA
		Handsfree mode		11	13	mA
		LD dialing, V <sub>DD</sub> = 2.5V		200		μΑ
		Ring mode, V <sub>DD</sub> = 2.5V		300		μΑ
I <sub>DD-EXT</sub>	Available Supply Current for Peripheral Circuits	Speech mode or MF dialing	2			mA
I <sub>DD0</sub>	Idle Current	Idle mode, V <sub>DD</sub> = 2 V, T <sub>AMB</sub> = 25°C		2		μA
$V_{LI}$	Line Voltage, LIVolt = 0	13 mA< I <sub>LINE</sub> < 100 mA	3.2	3.5	3.8	V
	Line Voltage, LIVolt = 1		4.2	4.5	4.8	V
loL	Output Current, Sink Pin CS, SS	V <sub>OL</sub> = 0.4V		1		mA
VIL	Input Low Voltage	T <sub>AMB</sub> = 25°C	Vss		0.3 V <sub>DD</sub>	V
Vih	Input High Voltage	ge T <sub>AMB</sub> = 25°C			$V_{DD}$	V

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## AC Electrical Characteristics

 $I_{LINE}$  = 15 mA; f = 800 Hz; ZAC = 1000 $\Omega$ ; recommended operating conditions and default settings unless otherwise specified.

## Transmit

Symbol	Parameter Conditions		Min.	Тур.	Max.	Units
Атх	Transmit Gain, default	M1/M2 to LS	34.3	35.8	37.3	dB
		M3/M4 to LS	42.0	44.0	46.0	dB
	Variation with Frequency	Variation with Frequency f = 500 Hz to 3.4 kHz		+/- 0.8		dB
THD	Distortion	V <sub>LS</sub> < 0.25 V <sub>RMS</sub>			2	%
V <sub>AGC</sub>	Soft Clip Level	V <sub>LS</sub> =		2		V
Asco	Soft Clip Overdrive			20		dB
tDECAY	Attack Time			70		µs/6dB
tDECAY	Decay Time			100		ms/6dB
Zin	Input Impedance; M1/M2			10		kΩ
	Input Impedance; M3/M4			10		kΩ
Амите	Mute Attenuation	Mute activated	60			dB
V <sub>NO</sub>	Noise Output Voltage	T <sub>AMB</sub> = 25°C, handset mode A <sub>TX</sub> = 37 dB			-72	dBmp
VIN MAX	Input Voltage Range; M1/M2 and M3/M4	Differential		+/- 1		V <sub>РЕАК</sub>

## Line Driver

Symbol	Parameter	Test Conditions	Min.	Typ.	Max.	Units
VIN MAX	Input Voltage Range; LI			+/- 2		VPEAK
V <sub>TX</sub>	Dynamic Range; LI			+/- 2		VPEAK
RL	Return Loss	$Z_{RL} = 1000 \ \Omega; T_{AMB} = 25^{\circ}C$				dB
$\Delta Z_{\text{AC/TEMP}}$	Temperature Variation			0.5		Ω/°C

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## Receive

Symbol	Parameter	Conditions	Min.	Тур.	Max.	Units
A <sub>RX</sub>	Receive Gain, Default	LS to RO; volume reset	-2.0	-0.5	1.0	dB
		LS to LO1/LO2; volume max.	25.0	27.0	29.0	dB
A <sub>RX-VOL</sub>	Volume Control Range	8 steps through [V-]/[V+] keys		15		dB
$\Delta A_RX$	Variation with Frequency	f = 500 Hz to 3.4 kHz		+/- 0.8		dB
THD	Distortion, RO	V <sub>RO</sub> = 0.25 V <sub>RMS</sub>			2	%
	Distortion, LO1/LO2	$V_{LO1/2} = 0.25 V_{RMS}$			5	%
V <sub>AGC</sub>	Soft Clip Level	V <sub>RO</sub> =		1		$V_{PEAK}$
Asco	Soft Clip Overdrive			10		dB
tdecay	Attack Time	V <sub>RI</sub> > 0.8 V <sub>RMS</sub>		70		µs/6dB
tdecay	Decay Time			100		ms/6dB
V <sub>NO</sub>	Noise Output Voltage, RO	T <sub>AMB</sub> = 25°C, A <sub>RX</sub> = 3 dB			-72	dBmp
Vufc	Unwanted Frequency Components	50 Hz20 kHz			-60	dBmp
Zin	Input Impedance, RI			8		kΩ
VIN-MAX RI	Input Voltage Range, RI			+/- 2		VPEAK
Ast	Side Tone Cancellation	V <sub>RI</sub> < 0.25 V <sub>RMS</sub>	26			dB
Z <sub>IN</sub>	Input Impedance, STB			80		kΩ
V <sub>IN-MAX</sub> ST	Input Voltage Range, STB			+/- 2		V <sub>PEAK</sub>

## General Timings

Symbol	Parameter	Conditions	Min.	Тур.	Max.	Units
td	Key Debounce Time			15		ms
ths-L	Low/High Debounce	HS Input (going off-hook) 15		15		ms
tнs-н	High/Low Debounce HS Input (going on-hook) 210			ms		
tfD	Flash Duration	FL 1 (default)	100		102	ms
		FL 2 (default)	275	285	300	ms
t <sub>PFP</sub>	Post Flash Pause			274		ms
t <sub>AP</sub>	Access Pause	P1 (default)	0.9	1.0	1.1	sec.
		P2 (default)		3.0		sec.

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## DTMF

Symbol	Parameter	Conditions	Min.	Тур.	Max.	Units
DF	Frequency Deviation	Deviation of ceramic resonator is not included			1.2	%
$V_{MF}$	MF Tone Level at LS (Low Group)	Default	Default -4.5 -5.5		-6.5	dBm
V <sub>MF-RANGE</sub>	MF Tone Level Range at LS (Low Group)	13 steps		-18/-6		dBm
$\Delta V_{L_{-}H}$	Pre-emphasis, Low to High 2.0 2.6		2.6	3.2	dBr	
Vufc	Unwanted Frequency Components	300 Hz4.3 kHz 4.3 kHz7 kHz 7 kHz10 kHz 10 kHz14 kHz 14 kHz28.5 kHz 28.5 kHz40 kHz			-40 -46 -52 -58 -70 -80	dBm dBm dBm dBm dBm dBm
tтD	Tone Duration, Minimum	Default	80	82.3	85	ms
t <sub>ITP</sub>	Inter Tone Pause, Min.	Default	80	82.3	85	ms
t <sub>TR</sub>	Tone Rise Time	From 10% to 90% amplitude			5	ms
t <sub>TR</sub>	Tone Fall Time	From 90% to 10% amplitude			5	ms
V <sub>CT</sub>	Comfort Tone Level, RO	Relative to LS, 4 steps		-36 -18		dBr

## LD Dialing

Symbol	Parameter	Conditions	Min.	Тур.	Max.	Units
tor	Dialing Rate	Default	9.53	10	10.5	pps
		Option		20		pps
t <sub>/B</sub>	Break Period	Default	57	61.2	63	ms
		Option		66		ms
		Option		33		ms
t <sub>M</sub> /	Make Period	Default	38	40.8	42	ms
		Option		33		ms
		Option		16.5		ms
tppp	Pre-Digit Pause			35		ms
tide	Inter-Digit Pause		800		880	ms

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## Tone Ringer

Symbol	Parameter	Conditions	Min.	Тур.	Max.	Units
F1	Frequency 1		770	800	830	Hz
F2	Frequency 2		1025	1067	1110	Hz
F3	Frequency 3		1280	1333	1385	Hz
t <sub>DT</sub>	Detection Time Initial			100		ms
f <sub>MIN</sub>	Min. Detection Frequency		12	13	14	Hz
f <sub>MAX</sub>	Max. Detection Frequency		68	70	75	Hz

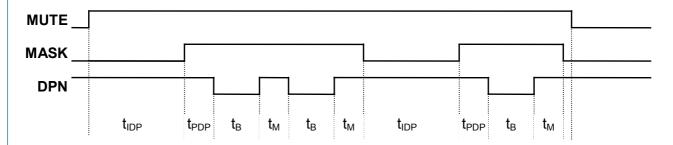
## Pacifier Tone

Symbol	Parameter	Conditions	Min.	Typ.	Max.	Units
VPARO	Tone Output Level, RO	RL = 150		30		mV <sub>P-P</sub>
	Tone Output Level, LO1/2	RL = 32		300		mV <sub>P-P</sub>
f <sub>PT</sub>	Frequency	requency Key entry		1333		Hz
tтвр	Tone Burst Duration	Key entry		40		ms
		Acknowledge, terminate PG mode		140		ms
		Rejection, invalid key entry		4x40		ms
titp	Inter-Tone Pause			80		ms

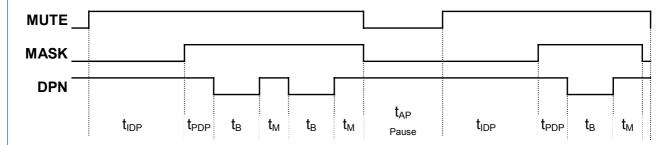
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# **Timing Diagrams**

# LD dialling



## LD dialling with access pause



## MF dialling

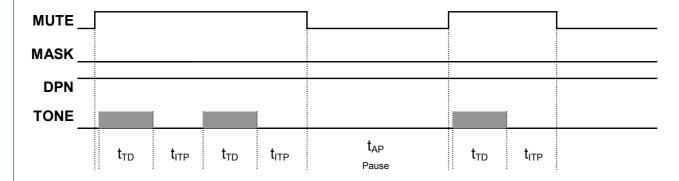


Figure 9: Timing diagrams

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# **Test Circuit**

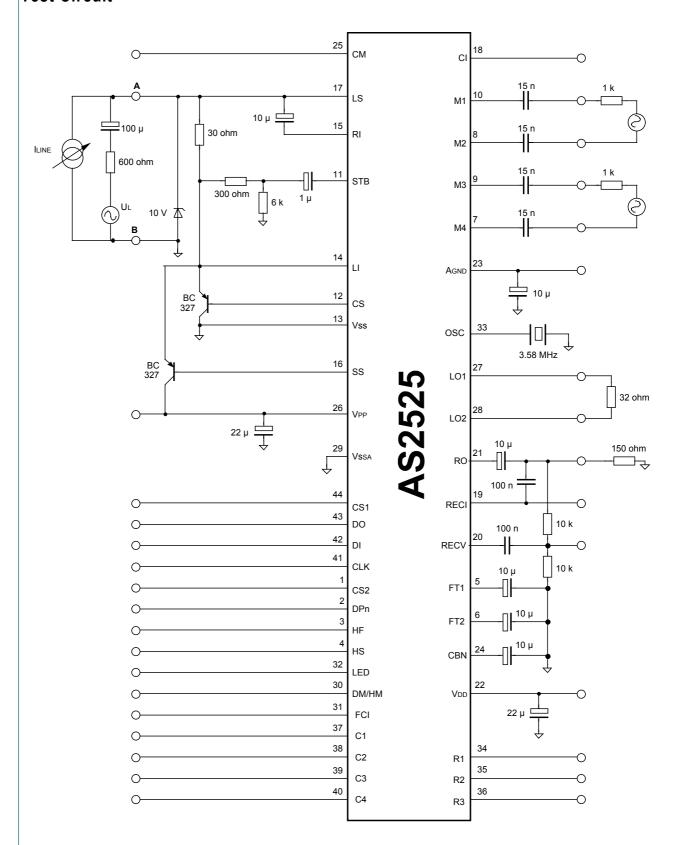


Figure 10: Test circuit

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# Typical Application

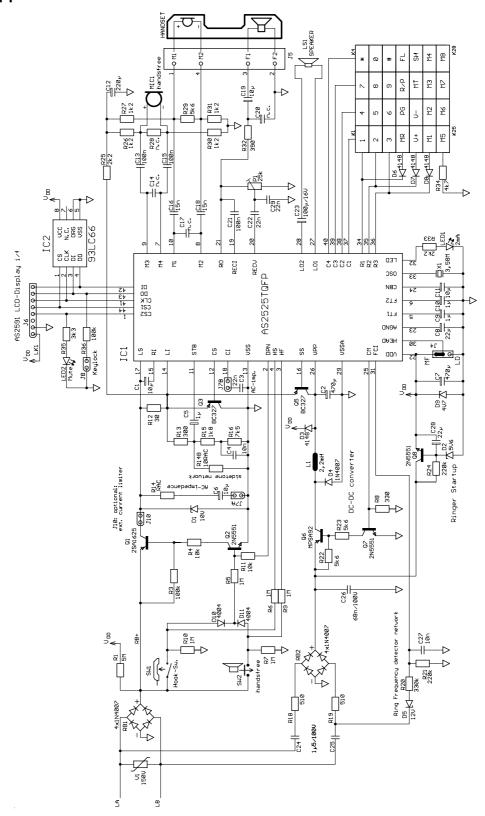


Figure 11: Typical application

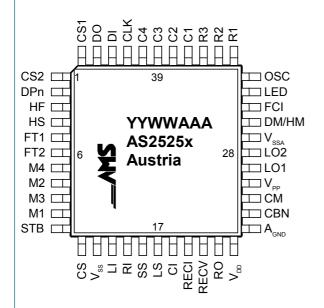
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## **Packaging**

44-in plastic TQFP (suffix Q)

For exact mechanical package dimensions please see austriamicrosystems<sub>AG</sub> packaging information.

# Marking, Pin-out



YY year of production

WW calendar week of production

AAA austriamicrosystems<sub>AG</sub> assembly ID

X revision

## **Ordering Information**

Number	Package	Description
AS2525 Q	TQFP	plastic thin quad flat package –
		44 leads (suffix T)
AS2525 F	DOF	Dice-on-Foil

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