



## austriamicrosystems AG

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The technical content of this austriamicrosystems datasheet is still valid.

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## Telephone Line Interface and Speakerphone Circuit AS2523, AS2524, AS2524B

DATA SHEET

## Key Features

- Line/Speakerphone circuit on a 28-pin CMOS-IC, simple inventory: same die for AS2523/24 and AS2523/24B Serial I/F on AS2523, parallel I/F on AS2524
- Enhanced voice switching
- Background noise monitoring
- Tx- and Rx-gain programmable on AS2523 only
- Digital volume control of Rx signals on AS2523 only
- DC characteristic programmable on AS2523 only
- Dual softclipping in handset mode on AS2523 only
- Dual softclipping in handsfree mode on AS2523 only
- Loudspeaker amplifier for loudhearing and handsfree
- Supply voltage generation for external circuitry
- Automatic line loss compensation on AS2523 / 24B only
  Real and complex impedance selectable by external components
- Side tone adaptation selectable by external components
- Unique EMC performance
- Operating range from 15mA to 100mA (down to 5mA with reduced performance)
  - Few external components

## General Description

AS2523/24 is a CMOS integrated circuit that incorporates DC and AC line adaptation (DC-mask and synthesized ACimpedance of  $1000\Omega$ ) as well as a speech circuit with softclipping, line loss compensation and Rx-volume control for handset and handsfree operation. It shall act as an a/bline powered device, which is controlled by a CPU via a serial interface on AS2523 or a standard dialler via a parallel interface on AS2524 and AS2524B.

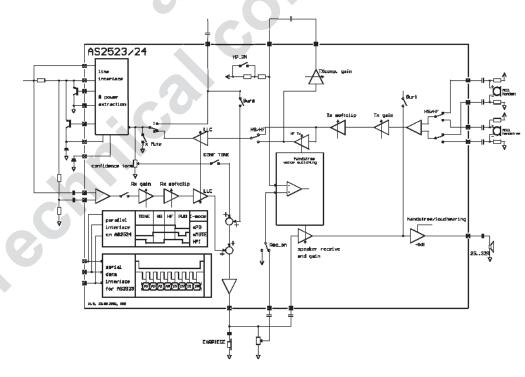
## Applications

Enhanced handsfree feature phones with CallerID and extended displays. The AS2524 and AS2524B are developed to interface with common Taiwanese dialers.

## Package

Available in 28-pin SOIC or die

# Block Diagram

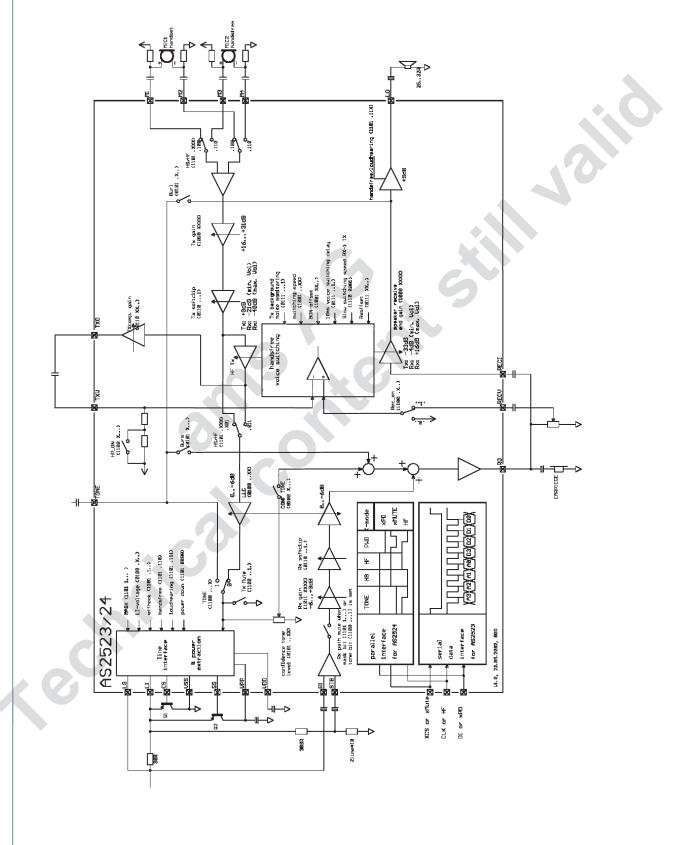


## Pin description

Pin #	Symbol	Function	
15	LS	Line Current Sense Input Analog input for sensing the line current	
12	LI	Line Input Analog input used for power extraction and line current sensing	
13	RI	Receive Input Analog input for ac-separated receive signal	
9	STB	Side Tone Balance Input Analog input for side tone cancellation network	
10	CS	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	
16	CI	Complex Impedance Input Analog input pin for the capacitor to program a complex impedance	
14	SS	Supply Source Control Output 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	
11	VSS	Voltage Source Negative Analog Power Supply	
18	VDD	Voltage Drain Drain Positive Analog Power Supply	
19	AGND	Analog Ground Special ground for the internal amplifiers	
8	M1	Microphone Input 1 Differential input for the handset microphone (electret)	
6	M2	Microphone Input 2 Differential input for the handset microphone (electret)	
7	M3	Microphone Input 3 Differential input for the handsfree microphone (electret)	
5	M4	Microphone Input 4 Differential input for the handsfree microphone (electret)	
17	RO	Receive Output to Handset Output for driving a dynamic earpiece with an impedance from 150 $\Omega$ to 300 $\Omega$	

25    xCS    Chip Select on AS2523 Chip select input of the serial interface. Intern      /xMUTE    Mute transmit-, receive- path and activate tone      26    DI    Data Input on AS2523 Data input of the serial interface. Internal pull-      /xPD    Power down mode for AS2524 and AS2524B (received)      27    CLK    Clock on AS2523 Clock input of the serial interface. Internal pull	e path on AS2524 and AS2524B 
/xMUTE  Mute transmit-, receive- path and activate tone    26  DI  Data Input on AS2523 Data input of the serial interface. Internal pull-    /xPD  Power down mode for AS2524 and AS2524B (received)    27  CLK  Clock on AS2523	e path on AS2524 and AS2524B 
26  DI  Data Input on AS2523 Data input of the serial interface. Internal pull- /xPD    27  CLK  Clock on AS2523	up resistor ( 100kOhm )
27  CLK  Clock on AS2523	
/xPD  Power down mode for AS2524 and AS2524B (r    27  CLK    Clock on AS2523	
27 CLK Clock on AS2523	On-hook)
Clock input of the serial interface. Internal pul	
/HFE Handsfree mode operation on AS2524 and AS	2524B
23 LO Loudspeaker Output. Output pin is for a 25Ω	loudspeaker
2 FT1 Analog input pin for connecting a capacitor for	offset cancellation.
3 TXO Analog output of the transmit signal	
4 TXV Analog input for the transmit signal in the voic	e switching path
1 CBN Analog input pin for connecting a capacitor for	r background noise monitoring.
20 RECI Analog input for the handsfree receive path. S capacitor.	hould be connected to RO via coupling
21 RECV Analog input for receive voice switching path.	0
24 VSSA Power supply pin for LO output amplifier.	
22 VPP Power supply pin for LO output amplifier.	
28 TONEIN Analog input for DTMF-signals.	

## Detailed Block Diagram



## Functional Description

#### DC conditions

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

The DC characteristic is determined by the voltage at LIpin and a  $30\Omega$  resistor between LI- and LS-pin. It can be calculated by the following equation: VLS = VLI + ILine \*  $30\Omega$ . VLI can be programmed to be 3.5V or 4.5V.

#### 2/4 wire conversion

AS2523/24 has built in two Wheatstone bridges 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 ACimpedance.

#### AC-impedance

The AC-impedance of AS2523/24 is set to t.m.  $1000\Omega$ . With the external capacitor at CI-pin it can be programmed complex. With an external resistor of approx.  $1.5k\Omega$ connected to the LS-pin it can be programmed to  $600\Omega$ .

#### Side Tone

A good side tone cancellation can be achieved by using the following equation: ZBAL/ZLINE = 10

#### Transmit path

The gain of the M1/M2  $\rightarrow$  LS is set to +37dB. This gain can be changed by programming on AS2523 from +30dB to +45dB in 1dB steps (Register *Txgain*), on AS2524 / 24B it is set to +37dB per default. The input is differential with an impedance of 10k $\Omega$ . The soft clip circuit limits the output voltage at LS to 2Vp. There is LLC for this path. The gain of the M3/M4  $\rightarrow$  LS is set to +46dB. This gain can be changed by programming from +39dB to +54dB in 1dB steps on AS2523. The input is differential with an impedance of 10k $\Omega$ . The soft clip circuit limits the output voltage at LS to 2Vp. There is no LLC for this path.

#### Receive path

The gain of the LS  $\rightarrow$  RO receive path is set to +7.5dB. This gain can be changed by programming from -7.5dB to +7.5dB in 1dB steps (Register *Rxgain*). The receive input is the differential signal of RI and STB. The soft clip circuit limits the output voltage at RO to 1Vp. It prevents harsh distortion and acoustic shock. There is LLC for this path.

The gain of the LS  $\rightarrow$  LO receive path is set to +32dB. This gain can be changed by programming from +19dB to -34dB in 1dB steps. The user can also change the gain via Register *Handsfree receive end gain* (See section "Handsfree"). The receive input is the differential signal of RI and STB. The soft clip circuit limits the output voltage at LO to 1.1Vp. It prevents harsh distortion and acoustic shock. There is optional LLC for this path.

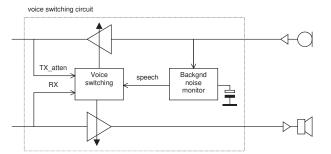
#### Line Loss Compensation (AS2523, AS2524B)

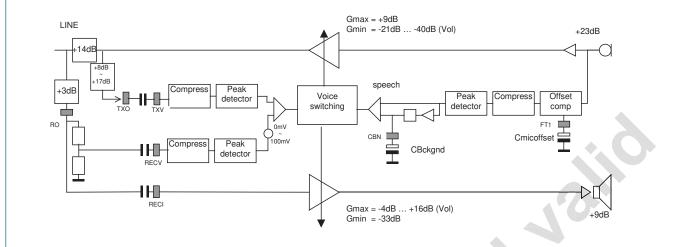
- programmable on AS2523 not activated at AS2524
- permanently activated at AS2524B

When it is activated, the transmit and receive gains for both I/O's are decreased by 6dB at line currents from 20mA to 50mA or from 45mA to 75mA.

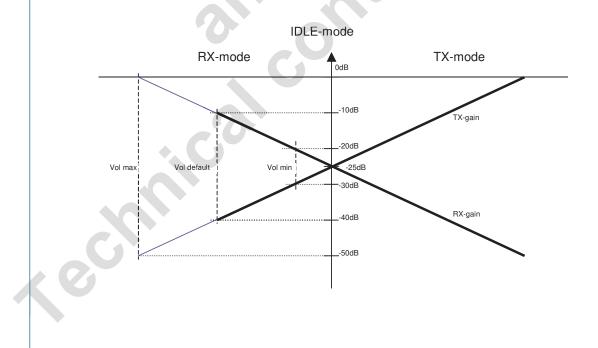
#### Handsfree

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 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 Register *Handsfree receive end gain* between 0dB and -20dB. The following table shows how voice switching is controlled



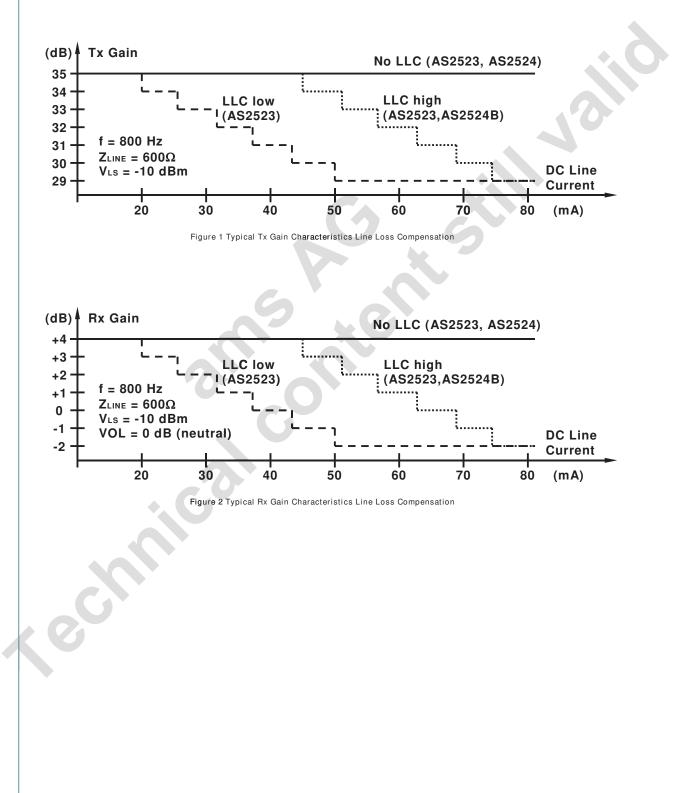


	Speech	Mode	Rx-g <b>ai</b> n	Tx-gain	Remark
Rx > Tx_atten	х	Receive	0db to -20dB	-50dB to -30dB	adjustable with VOL-setting
Tx_atten > Rx	NO	Idle	-25	-25	middle position
Tx_atten > Rx	YES	Transmit	-50dB	0dB	independent of VOL-setting



## Typical Characteristics of Line Loss Compensation (AS2523, AS2524B only)

The line-loss-compensation is programmable at the AS2523, not activated at the AS2524 and LLC high (45mA-75mA) activated at the AS2524B.



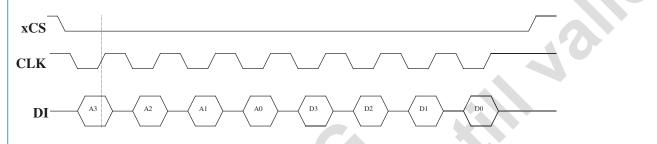
#### Serial Interface on AS2523

#### Registers

The settings of the AS2523 are stored in 16 registers. Each register has 4 bit data width.

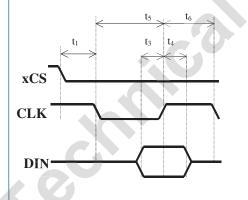
#### Timing

The data format for writing to a register has the following form:



Note: The pins xCS,CLK,DI have internal pull-up resistors.

Symbol	MIN	ТҮР	MAX
Т6	100ns	2	
T5	100ns		
T1	50ns		
Т3	50n <b>s</b>	$\mathbf{O}$	
Т4	50ns		
	T6 T5 T1 T3	T6  100ns    T5  100ns    T1  50ns    T3  50ns	T6  100ns    T5  100ns    T1  50ns    T3  50ns



#### Serial interface Registers

The following table shows the content of the 16 control registers.

Address	Data	Control registers AS2523		it value reset	
A A A A	DDDD		for AS2523,	for AS2524 LH bond	
3210			AS2524HF	option	
0 0 0 0	Nop		na	na	
0 0 0 1			na	na	
0 0 1 0			na na	na na	
0 1 0 0		Conf-Tone: 0 No Conf Tone 1 Conf Tone	0111	0111	
		LIV: Voltage at pin LI 03.5V, 14.5V LLC1 LLC0: Line loss compensation setting 0 0 20mA-50mA 0 1 NO Line loss compensation 1 0 45-75mA			
0 1 0 1	Conf Liv LIc1 LIc0	1 1 NO Line loss compensation	0001	0001	
	S2 T F 0	BURS: Analog tone at R0    0OFF,    1ON      BURL: Analog tone at LS    0OFF,    1ON      CT1 CT0: @RO    @LS    0    -36dB    -15dB      0    0    -36dB    -15dB    TONE-confidence level      0    1    -30dB    -9dB    rel. to pin LS      1    0    -24dB    -3dB      1    1    -18dB    +3dB			
0 1 1 0	× Burt × Burt × CT1 × CT0	Softclip-settings, Tx_comp_gain settings	1000	1000	
	tx_comp_gain Bit 1 > tx_comp_gain Bit 0 > Soft-RX > Soft-RX Soft-TX > Soft-TX > Soft-TX > 1	Bitl Bit0 :Tx_comp_gain      O      0      0      46dB      O      0      1 </td <td></td> <td></td> <td></td>			
0 1 1 1	00044		0101	1100	
	R xoffset Bit1 R xoffset Bit0 Delay_10ms Bnon	Bit1      Bit0:      Rx-offset setting        0      0      0mV        0      1      12mV        1      0      40mV        1      1      100mV        Delay_10ms      0 no delay      1 10ms delay        Bnon      0 EGN off      1 EGN on			
1000	16 gains	Handsfree receive endgain	1111	1011	
1001	x x x x	Data RXgain TXgain 0x0 -20dB -30dB Min. receive volume : : 0xF +0dB -50dB Max. receive volume off1 off0: BGN-offset	0000	1100	
	off1 off0 Hfs1 Hfs0	0 0 120mv 0 1 180mV 1 0 240mV 1 1 300mV Hfs1 HFs0: Speed of voice switching 0 0 max speed 0 1 1 0 1 1 min speed			
1 0 1 0	16 gains	Transmit gain [16], 16 steps, 1dB stepsize	0111	0111	
0 1 1	16 gains	Data      HS-mode      HF-mode        0x0      30dB      39dB      :        :      :      :      :        0xF      45dB      54dB        Receive gain [16], 16 steps, 1dB stepsize	1111	1111	
I V I I	io gailis	Data HS-mode HF-mode 0x0 -6dB 19dB : : :			
		0xF +9dB 34dB			
1 1 0 0	HP_on Rec_on Mute Fone	HP_on      0 fg=200Hz      1 fg=2000Hz        rec_on      0 Rxcomp=AGND      1 Rxcomp=RECV        Mute      0 No mute      1 Tx-path muted        Tone      0 Tx-signal path      1 Ton in signal path	0100	1000	
101	XXXX	Mask, Lh, Hf, Hook Switch Mask: 0 No mask 1 Mask acitivated	0100	0101	
	Mask His LH	His      HF      LH: Hook functions        0      x      x      Power down        1      0      0      Handset        1      0      1      not allowed        1      0      1      not allowed        1      1      0      Handsfree        1      1      1      Loudhearing			
1 1 1 0	(rat5 × est3 × est2 × low_rec ×	slow_rec: fast change to TX, slow change to RX, no IDLE Krat5,Test3,Test2: For factory test only !	0000	0001	
1 1 1 1		Reset to defaults	na	na	

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#### Programming Guidelines for the AS2523

The AS2523 is programmed by means of a serial 8-bit shift register. MSB is clocked in first, LSB last. The first four bits (7,6,5,4) are the addresses of the registers, the last four bits (3,2,1,0) are the data bits.

Each register has a default setting (see data sheet), which is set after power-up of the chip.

The internal registers are RAM-cells. When the AS2523 loses VDD (as in on-hook state) it also loses the register contents.

It is therefore necessary to re-write the affected registers after each hook event. The register contents cannot be read, they can only be written.

Registers need to be re-written after each power-up and after each hook event (handset / handsfree / on-hook).

It is recommended to always initialize the chip with a "reset to defaults command" (Fx) first and then write the appropriate registers that need to be changed from the default setting.

Alternatively, all registers (4 to 16) may be re-written in a bulk at each power-up and hook event (handset / handsfree / on-hook).

This guarantees safe operation in case of unexpected loss of power during normal operation. Re-writing all registers also eases later software updates, as only register contents need to be changed, but no additional command lines need to be inserted.

#### Example of a typical power-up sequence

A typical power-up sequence will require the following programming.

- a) determine the cause of power-up (handset / handsfree mode) and set register D accordingly.
   Default = handset mode
- b) Set the LI voltage to 3.5V or 4.5V (default = 4.5V) and Line Loss Compensation (default = off) in register 4
- c) Set the Confidence Tone level (e.g. the DTMF level which is audible in the handset) and path in register 5 Default = off
- d) Set the required Tx and Rx gains in Registers A and B
  Default = 37dB Tx gain, 1dB Rx gain

#### Additional settings in Handsfree mode

Note: change the default settings only if necessary:

- A click-free startup can be achieved by starting up in handset mode (=default) and then switching to handsfree, once the chip has stabilized.
- b) Change the Tx comparator preamplifier gain in register 6
  Default = 14dB
- c) Change the Receive DC offset and Background noise monitoring in Register 7 default = 0mV offset (higher val. puts more weight on Rx) switching delay default = on Background Noise Monitoring = on
- d) Set the speaker amplifier volume in register 8
  Default = 0B
- e) Set the handsfree voice switching speed and Background Noise monitoring offset in register 9: default voice switching speed = 1ms/6dB (fastest) default BGN monitoring offset = 240mV (determines the Tx level required to switch from idle to Transmit)

#### Additional settings during Flash

During a Flash, the AS2523 should be powered down to avoid discharge of VDD:

Set the MASK bit (bit3 in register D) to 1 After the line current is restored, the MASK bit must be cleared again. Other settings do not need to be re-written, as VDD has not discharged. As a safety margin however, it is recommended to re-write all registers after a flash

#### Additional settings during line breaks

If a line-break-detection (brief interruption of the line while in off-hook state) is implemented, the same rules apply as for a Flash: set the MASK bit to 1 to avoid discharge of VDD. It is recommended to re-write all registers after a line-break, as a line break may take long enough to discharge VDD, even when the MASK bit was set.

#### General Rules

The serial interface may be programmed at any time, it does not affect the speech quality, e.g. if a register is overwritten with the same value.

It is also possible to re-write all registers periodically. The register is static, therefore it can be clocked at any speed up to 5MHz.

However, electromagnetic pulses on the clock and data lines may cause unwanted programming of the chip. It is therefore recommended to keep these lines short, filter them by a discrete lowpass filter and reduce the clock speed accordingly.

#### Parallel Interface on AS2524 and AS2524B

Input Signal on Pin						
XPD	xMute	HF	Mode			
0	0	0	Power down			
0	0	1	Power down			
0	1	0	Power down			
0	1	1	Power down			
1	0	0	Tone mode			
1	0	1	Tone mode			
1	1	0	Handset mode			
1	1	1	Handsfree mode			

#### Power down mode

For low power consumption the analog part is turned off during activated power down mode. The Power down mode is used during Flash and Pulse dialing and for On-hook operation.

#### Handset mode

#### Speech mode is Handset mode (M1-M2, RO active; LO deactivated)

Parameter	AS2524	AS2524B
LI Voltage	4.5V	4.5V
LLC	No LLC	LLC high (45mA to 75mA)
TX-Gain	+37dB	+43dB
RX-Gain	+7.5dB	+7.5dB
Rx-Softclip	Off	Off
Tx-Softclip	Off	Off
Tone	Off	Off

#### Handsfree mode

#### Speech mode is Handsfree mode (M3-M4, RO active; LO active)

Parameter	<b>AS25</b> 24	AS2524B
LI Voltage	4.5V	4.5V
LLC	No LLC	LLC high (45mA to 75mA)
HF TX-Gain	+46dB	+52dB
HF RX-Gain	+34dB	+34dB
Rx-Softclip	Off	Off
Tx-Softclip	Off	Off
Tone	Off	Off
BNM	On	On

#### Tone mode

This mode is used to send DTMF and FSK data to the line.

Speech mode is Tone mode (M1-M2;M3-M4 muted, Rx path muted, RO and LO active)

Parameter	AS2524	AS2524B
LI Voltage	4.5V	4.5V
LLC	No LLC	LLC high (45mA to 75mA)
Tone TX-Gain	+14dB	+14dB
Confidence tone level	-36dB@Ro -15dB@Lo relative to Pin LS	-36dB@Ro -15dB@Lo relative to Pin LS

#### Loudhearing AS2524 and AS2524B (feature only available as DIE)

LI Voltage 4		.5V	4.5V		
LLC	C No LLC		LLC high (45mA to	75mA)	
Tone TX-Gain	+1	4dB	+14dB		
Confidence tone	-36dB@Ro	o −15dB@Lo	-36dB@Ro -15d	B@Lo	
level	relative	to Pin LS	relative to Pin	LS	
oudbearing AS2P	524 and AS25	24B (feature on	ly available as DI	=)	
beech mode is Loud		,	-	_/	
	0.		document "AS2523/24	4 Delivery a	as Dice").
Parame	ter	A	S2524		AS2524B
LI Voltage			4.5V		4.5V
LLC		No LLC		LL	C high (45mA to 75mA)
TX-Gain		+	37dB		+43dB
RX-Gain		+	7.5dB		+7.5dB
Tone		Off			Off
Slow_rec		0	On		On
RECV pin		Dea	Deactivated		Deactivated
Highpass 2000kHz		2000kHz (10n	F from TXO->TXV)	2000k	Hz (10nF from TXO->TXV)
Voice switching spe	ed	120	ms/6dB		120ms/6dB
RX-Offset		3	00mV		300mV

## Electrical characteristics

Electrical characteristics are measured with the Test Circuit application. Typical mean values will not be tested.

#### Absolute maximum ratings

Positive Supply Voltage	-0.3V <= VDD <= 7V
Input Current	+/- 25mA
Input Voltage (LS)	-0.3V <= Vin <= 12V
Input Voltage (LI, CS)	-0.3V <= Vin <= 8V
Input Voltage (STB, RI)	-2V <= Vin <= VDD+0.3V
Digital Input Voltage	-0.3V <= Vin <= VDD+0.3V
Electrostatic Discharge (HBM 1.5k $\Omega$ -100pF)	+/- 1000V
Storage Temperature	-65°C to +125°C

#### Recommended operating conditions

Supply Voltage (generated internally)	3V <= VDD <= 5V
Operating Temperature	-25°C to +70°C

#### DC characteristics

ILine=15mA w/o operation of any additional external circuitry, unless other specified

Symbol	Parameter	Conditions	Min	Туре	Max	Units	Test
Idds	Operating Current	Speech Mode		3	6	mA	Y
Iddh	Operating Current	Handsfree Mode		7	10	mA	Y
Vli	Line Voltage LIVolt=0 Line Voltage LIVolt=1	15mA<=ILINE<=100mA	3.2 4.2	3.5 4.5	3.8 4.8	V V	Y Y
Iol	Output Current, Sink CS,SS	VOL=0.4V		1		mA	N

#### Transmit characteristics

ILine=15mA f=800Hz, default settings unless other specified

Symbol	Parameter	Conditions	Min	Туре	Max	Units	Test
AM1/2TX	Transmit Gain M1/M2 → LS AS2524 AS2524B	ZAC(syn)=1000Ω	35 39	+36.5 +42.5	38 44	dB dB	Y Y
Амз/4тх	Transmit Gain M3/M4 → LS AS2524 AS2524B	ZAC(syn)=1000Ω	43 49	+45.0 +51	47 53	dB dB	Y Y
Atone TX	Transmit Gain Tone $\rightarrow$ LS	ZAC(syn)=1000Ω	12	+14dB	16	dB	Y
ΔΑΤΧ	Variation with frequency	f = 500Hz 3.4kHz		±0.8		dB	Ν
AVRM12	Control range $M1/M2 \rightarrow LS$	16 steps		-7/+8		dBr	Ν
Avrm34	Control range M2/M4 → LS	16 steps		-7/+8		dBr	Ν
THD	Distortion	VLS=0.25VRMS			2	%	Y
VAGC1	Soft Clip Level M1/M2 $\rightarrow$ LS at LS			2		VP	Ν
VAGC2	Soft Clip Level M3/M4 $\rightarrow$ LS at LS			2		VP	Ν
Asco	Soft Clip Overdrive M1/M2 - M3/M4			20		dB	Ν
tattack	Attack time			70		us/6dB	Ν
tdecay	Decay time			100		ms/6dB	Ν

Symbol	Parameter	Conditions	Min	Туре	Max	Units	Test	
ZIN-M1/2	Input Impedance M1/M2			10		kΩ	N	
ZIN-M3/4	Input Impedance M3/M4			10		kΩ	N	
VINmax	Input Voltage Range M1/2 - M3/4	differential		±1		Vp	N	2
VNO	Noise Output Voltage LS	TAMP=25°C Handset mode Gain = 36.5dB Gain = 42.5dB		-72	-66	dBmp dBmp	N Y	
Амите	Mute Attenuation	Mute activated	60			dB	Y	

#### Receive characteristics

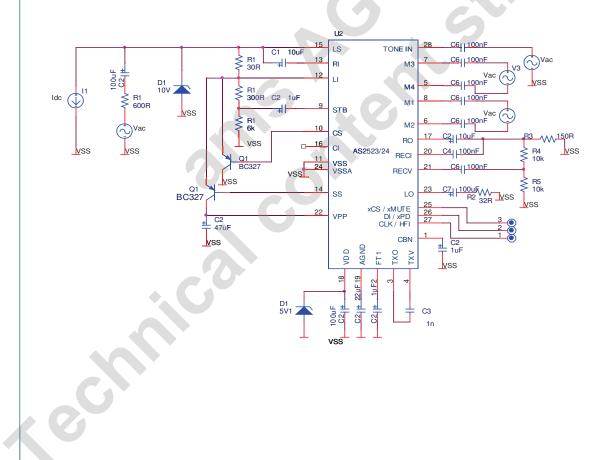
ILine=15mA f=800Hz, default settings unless other specified

Symbol	Parameter	Conditions	Min	Туре	Max	Units	Test
Aro	Receive Gain LS $\rightarrow$ RO	ZAC(syn)=1000Ω Vol default	6	+7.5	9	dB	Y
ALO12	Receive Gain LS $\rightarrow$ LO	ZAC(syn)=1000Ω Vol maximum	+32.0	+34	+36.0	dB	Y
Δ <b>A</b> RX	Variation with frequency	f=500Hz to 3.4kHz		±0.8		dB	Ν
Avrro	Control range $LS \rightarrow RO$	16 steps		-7/+8		dBr	Ν
AVRLS	Control range $LS \rightarrow LO$	16 steps		-7/+8		dBr	Ν
THDLS	Distortion LO	VLS=0.25VRMS			5	%	Y
THDRO	Distortion RO	VLS =0.25VRMS			2	%	Y
Vufc	Unwanted Freq. Cmp.	f = 200 20kHz			-60	dBm	Y
VAGCRO	Soft Clip Level RO			1		VP	Ν
VAGCLS	Soft Clip Level LO			1.1		VP	Ν
ASCLS	Soft Clip Overdrive LO			10		dB	Ν
Tattack	Attack time			70		us/6dB	Ν
Tdecay	Decay time			100		ms/6dB	Ν

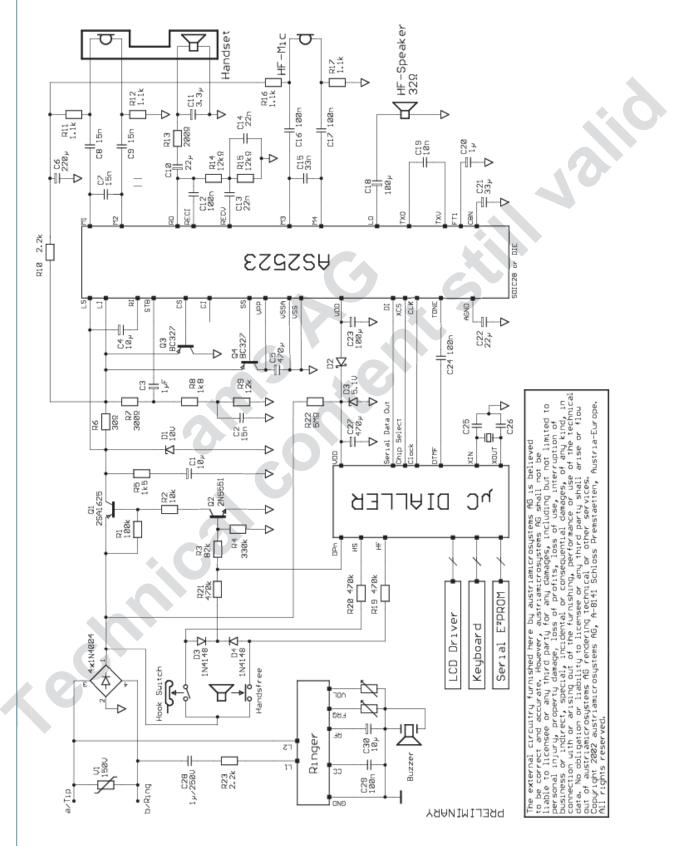
#### Data Sheet AS2523, AS2524, AS2524B

Symbol	Parameter	Conditions	Min	Туре	Max	Units	Test
Vno	Noise Output Voltage RO	TAMP=25°C Gain = 7.5dB			-68	dBmp	Y
ZIN-RI VINmax	Input Imp. RI Input Voltage Range RI			8 ±2		kΩ Vp	N
ZIN-STB VINmax	Input Imp. STB Input Volt. Range STB			80 ±2		kΩ Vp	N
ST	Side tone	VRI<=0.25VRMS	26			dB	Y
RL ∆Zac/°c	Return Loss Temp. Variation	$ZAC(syn)=1000\Omega$	18	0.5		dB Ω/°C	Y

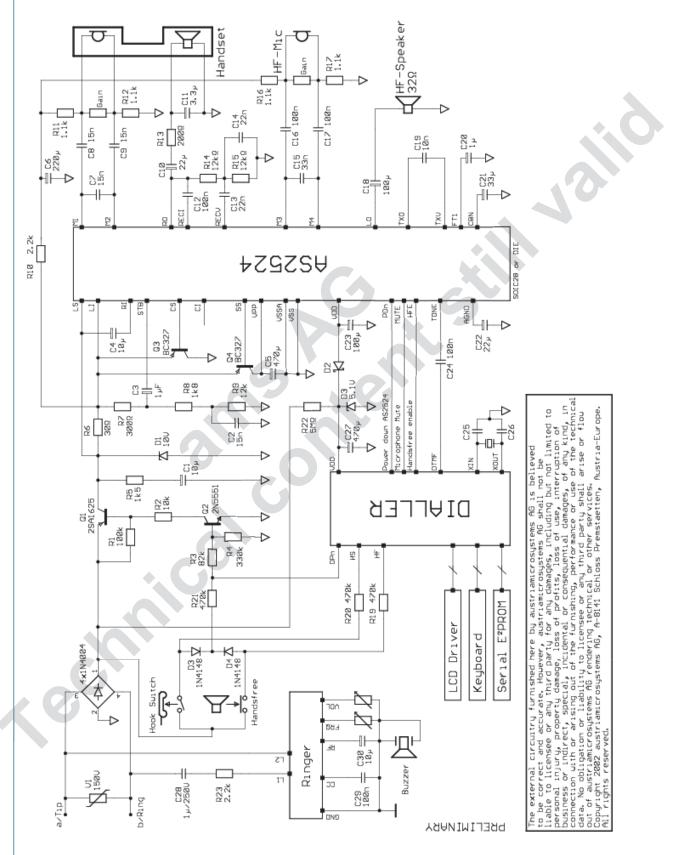
## Test circuit



## Typical application AS2523



## Typical application AS2524 and AS2524B

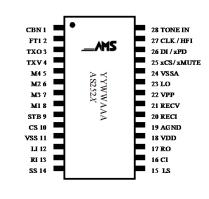


## Packaging

28-pin plastic SOIC

For exact mechanical package dimensions please see austriamicrosystemsAg packaging information.

## Pin-out, Marking



- YΥ year of production
- ww calendar week of production
- AAA austriamicrosystemsAG assembly ID
- 3 ... AS2523 Version Х
  - 4 ... AS2524 Version
  - 4B .. AS2524B Version

## Ordering Information

				6
ΥY	year of	production		
ww	calenda	r week of pro	duction	
ΑΑΑ	austrian	nicrosystemsA	<sub>IG</sub> assembly ID	
x	4 AS	S2523 Versior S2524 Versior S2524B Versio		N.C
Orde	0	nformatic Package	Description	
AS25	523 T	SOIC28	plastic surface mounted package – 28 leads	
AS25	524 T	SOIC28	Plastic surface mounted	
			package – 28 leads	
AS25	24B T	SOIC28	package – 28 leads Plastic surface mounted package – 28 leads	
AS25 AS25		SOIC28	Plastic surface mounted	
	523 F		Plastic surface mounted package – 28 leads	-

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