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 AC-impedance of 1000Ω) 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/b-line 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.

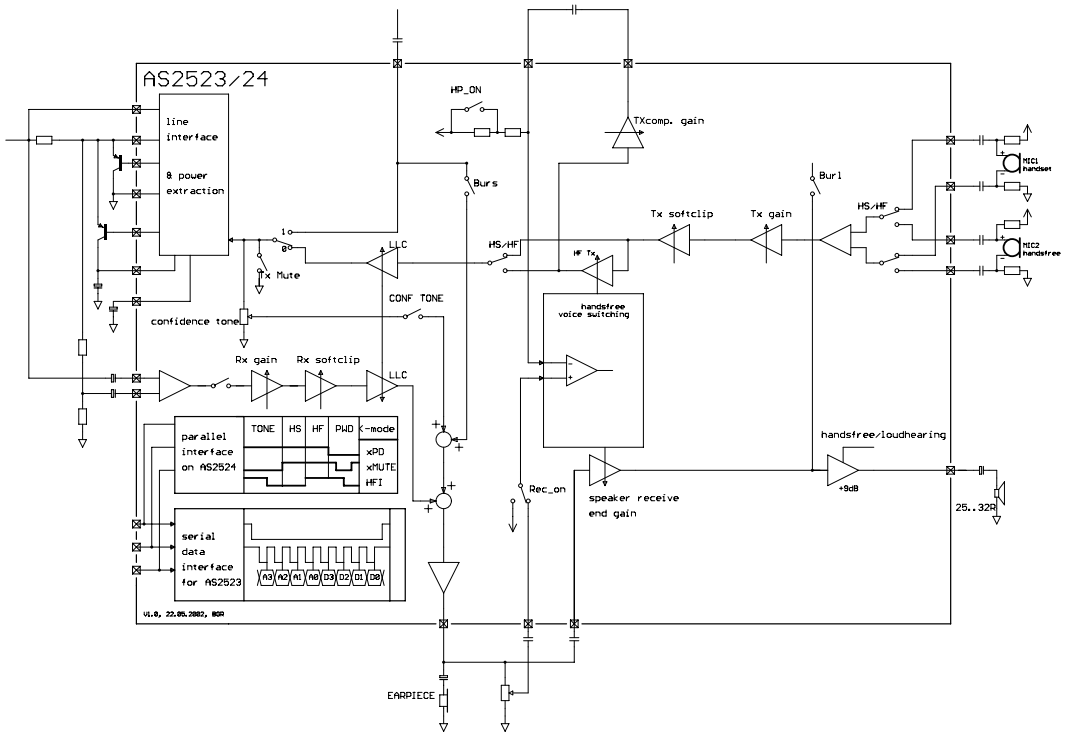
Applications

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

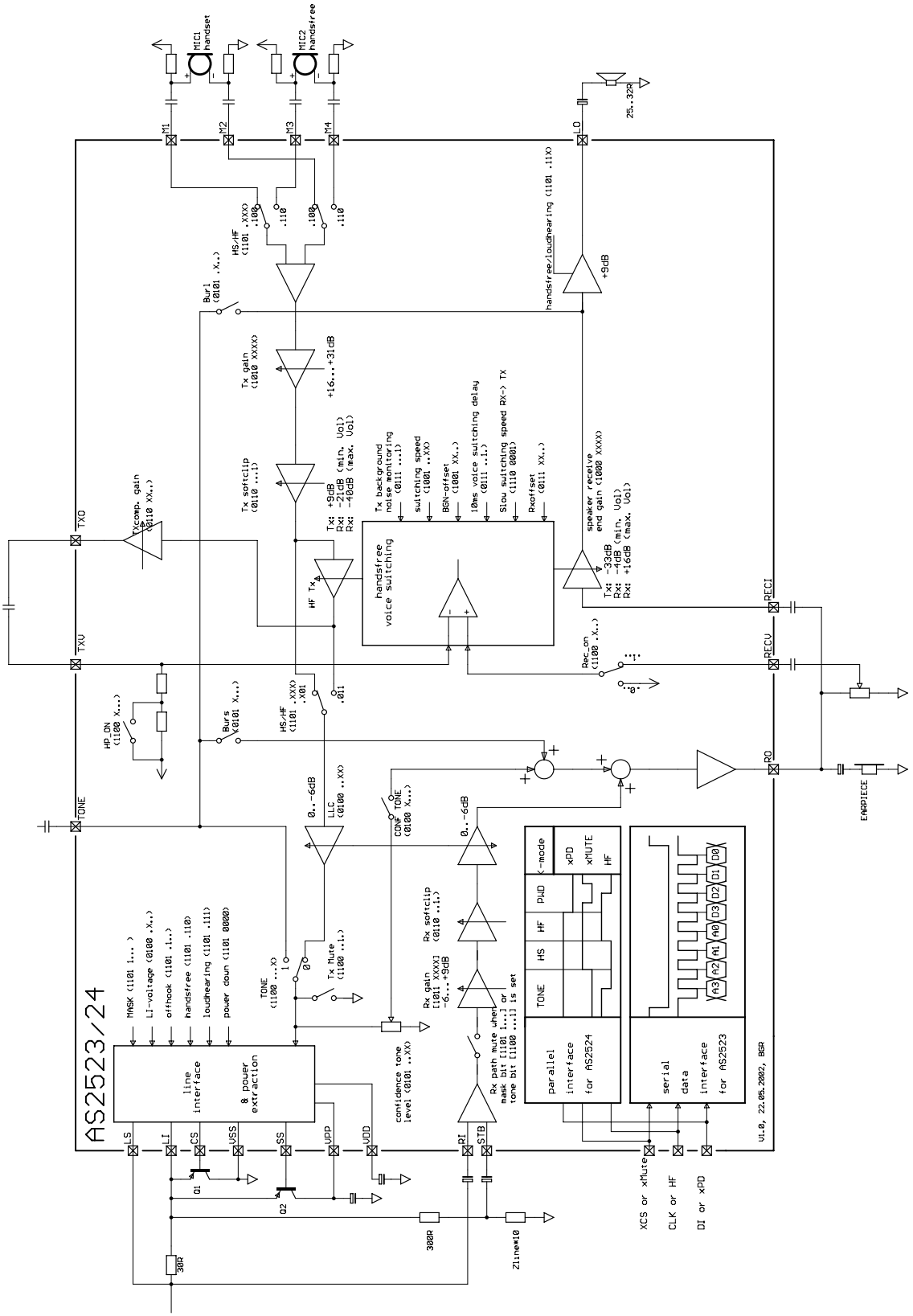


Pin description

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 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 150Ω to 300Ω

Pin #	Symbol	Function
25	xCS	Chip Select on AS2523 Chip select input of the serial interface. Internal pull-up resistor (100kOhm)
	/xMUTE	Mute transmit-, receive- path and activate tone path on AS2524 and AS2524B
26	DI	Data Input on AS2523 Data input of the serial interface. Internal pull-up resistor (100kOhm)
	/xPD	Power down mode for AS2524 and AS2524B (On-hook)
27	CLK	Clock on AS2523 Clock input of the serial interface. Internal pull-up resistor (100kOhm)
	/HFE	Handsfree mode operation on AS2524 and AS2524B
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 voice switching path
1	CBN	Analog input pin for connecting a capacitor for background noise monitoring.
20	RECI	Analog input for the handsfree receive path. Should be connected to RO via coupling capacitor.
21	RECV	Analog input for receive voice switching path.
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 LI-pin and a 30Ω resistor between LI- and LS-pin. It can be calculated by the following equation: $V_{LS} = V_{LI} + I_{Line} * 30\Omega$. V_{LI} 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 AC-impedance.

AC-impedance

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

Side Tone

A good side tone cancellation can be achieved by using the following equation:
$Z_{BAL}/Z_{LINE} = 10$

Transmit path

The gain of the M1/M2 → 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 10kΩ. The soft clip circuit limits the output voltage at LS to $2V_p$. There is LLC for this path.

The gain of the M3/M4 → 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 10kΩ. The soft clip circuit limits the output voltage at LS to $2V_p$. There is no LLC for this path.

Receive path

The gain of the LS → 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 $1V_p$. It prevents harsh distortion and acoustic shock. There is LLC for this path.

The gain of the LS → 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 $1.1V_p$. It prevents harsh distortion and acoustic shock. There is optional LLC for this path.

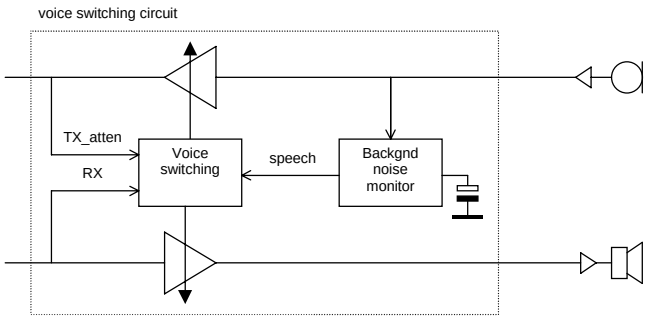
Line Loss Compensation (AS2523, AS2524B)

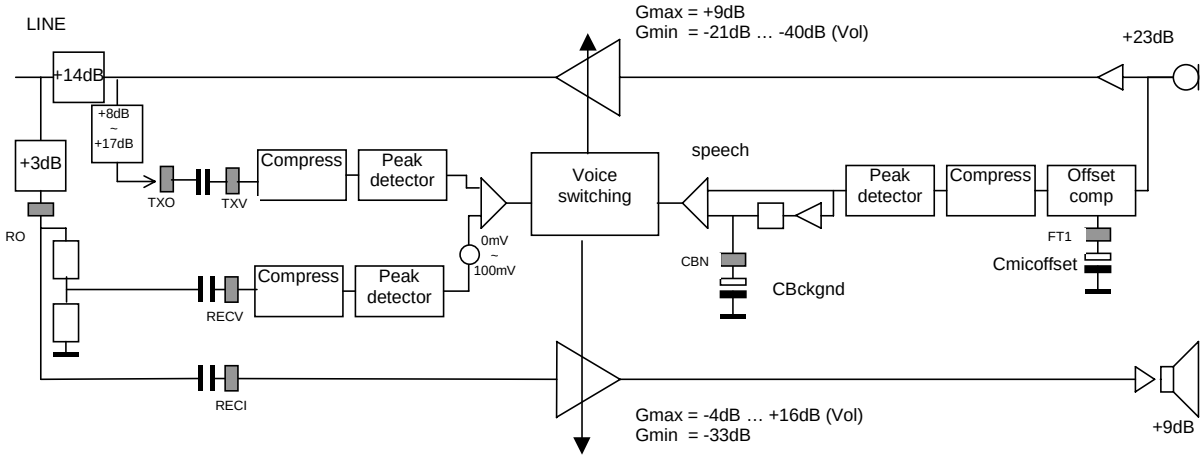
- programmable on AS2523
- not activated at AS2524
- permanently activated at AS2524B

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.

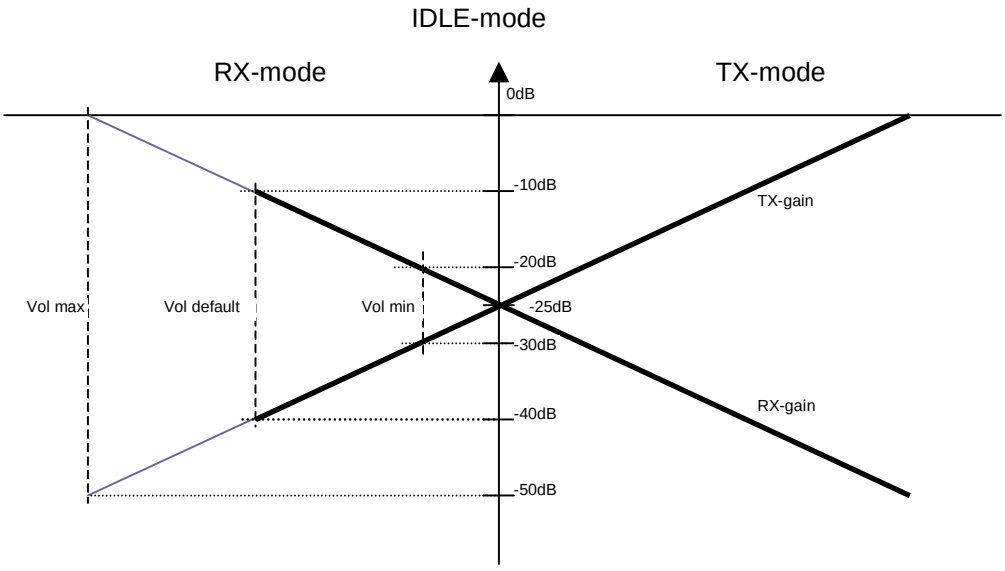
Handsfree

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-gain	Tx-gain	Remark
Rx > Tx_atten	X	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.

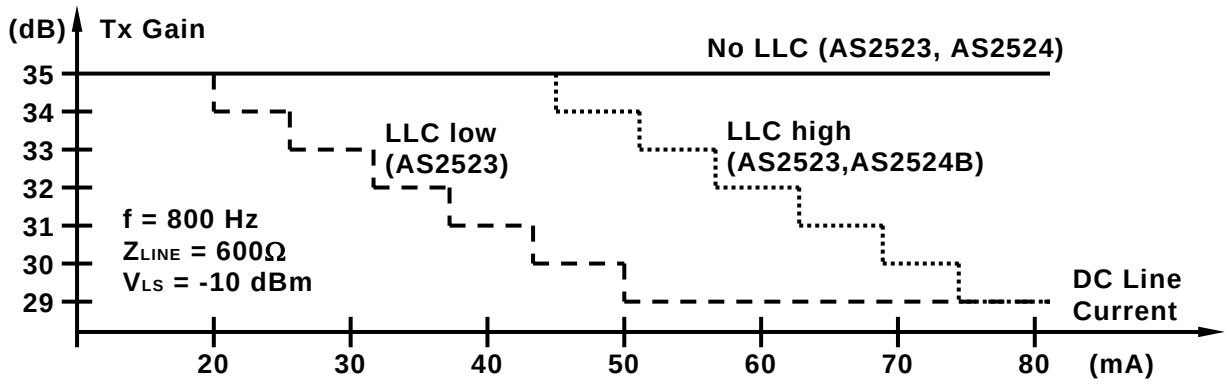


Figure 1 Typical Tx Gain Characteristics Line Loss Compensation

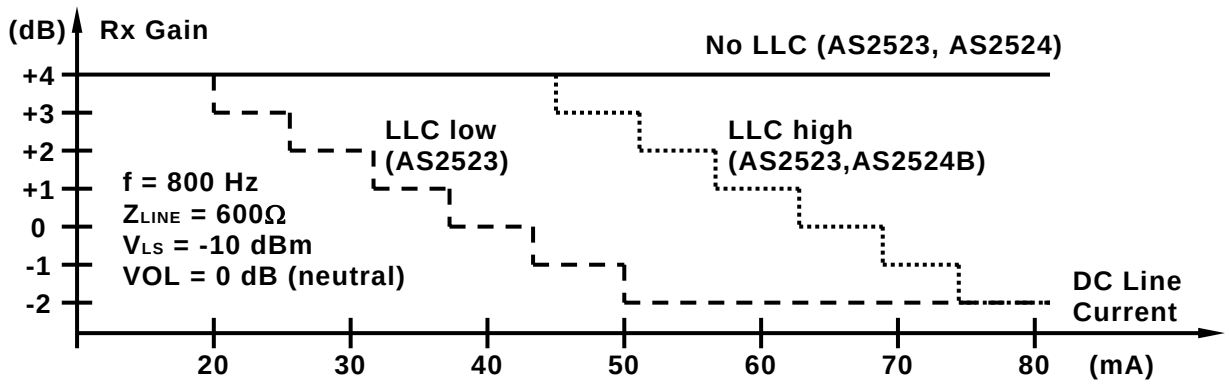


Figure 2 Typical Rx Gain Characteristics Line Loss Compensation

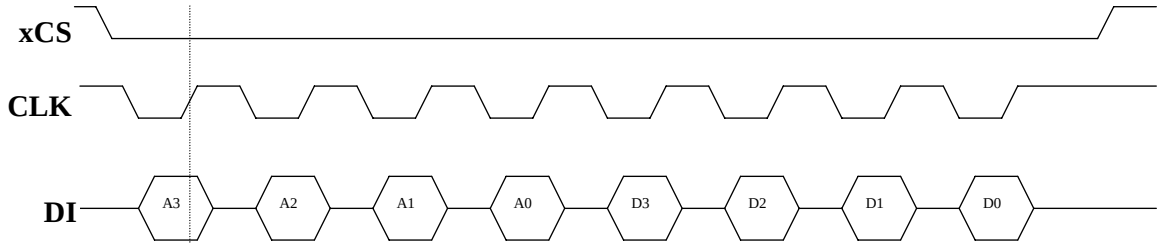
Serial Interface on AS2523

Registers

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

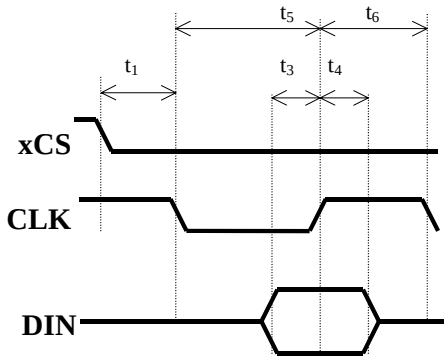
Timing

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



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

Parameter	Symbol	MIN	TYP	MAX
CLK Pulse width HIGH	T6	100ns		
CLK Pulse width LOW	T5	100ns		
xCS to first falling CLK-edge setup time	T1	50ns		
DIN to CLK setup time	T3	50ns		
DIN to CLK hold time	T4	50ns		



Serial interface Registers

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

Address				Data				Control registers AS2523	Default value after reset	
A3	A2	A1	A0	D3	D2	D1	D0		for AS2523, AS2524HF	for AS2524 LH bond option
0	0	0	0	Nop					na	na
0	0	0	1						na	na
0	0	1	0						na	na
0	0	1	1						na	na
0	1	0	0	x	x	x	x	**Conf-Tone:** 0... No Conf Tone 1... Conf Tone **LIV:** Voltage at pin LI 0...3.5V, 1...4.5V **LLC1 LLC0:** Line loss compensation setting 0 0 20mA-50mA 0 1 NO Line loss compensation 1 0 45-75mA 1 1 NO Line loss compensation	0111	0111
				Conf	Liv	Llc1	Llc0			
0	1	0	1					**BURS:** Analog tone at RO 0...OFF, 1...ON **BURL:** Analog tone at LS 0...OFF, 1...ON CT1 CT0: @RO @LS 0 0 -36dB -15dB TONE-confidence level 0 1 -30dB -9dB rel. to pin LS 1 0 -24dB -3dB 1 1 -18dB +3dB	0001	0001
				Burs	Burl	CT1	CT0			
0	1	1	0	x	x	x	x	Softclip-settings, Tx_comp_gain settings **Bit1 Bit0** :Tx_comp_gain 0 0 +8dB 0 1 +11dB 1 0 +14dB 1 1 +17dB **SOFTRX:** Softclip RX 0...OFF, 1...ON **SOFTTX:** Softclip TX 0...OFF, 1...ON	1000	1000
				Tx_comp_gain Bit 1	Tx_comp_gain Bit0	Soft-RX	Soft-TX			
0	1	1	1					**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... BGN off 1... BGN on	0101	1100
				Rxoffset Bit1	Rxoffset Bit0	Delay_10ms	Bnon			
1	0	0	0	16 gains				Handsfree receive endgain **Data** RXgain TXgain 0x0 -20dB -30dB Min. receive volume : : 0xF +0dB -50dB Max. receive volume	1111	1011
1	0	0	1	x	x	x	x	**off1 off0**: BGN-offset 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	0000	1100
				off1	off0	Hfs1	Hfs0			
1	0	1	0	16 gains				Transmit gain [16], 16 steps, 1dB stepsize **Data** HS-mode HF-mode 0x0 30dB 39dB : : : 0xF 45dB 54dB	0111	0111
1	0	1	1	16 gains				Receive gain [16], 16 steps, 1dB stepsize **Data** HS-mode HF-mode 0x0 -6dB 19dB : : : 0xF +9dB 34dB	1111	1111
1	1	0	0					**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
				HP_on	Rec_on	Mute	Tone			
1	1	0	1	x	x	x	x	Mask, Lh, Hf, Hook Switch **Mask:** 0... No mask 1... Mask acitivated **His HF LH:** Hook functions 0 x x Power down 1 0 0 Handset 1 0 1 not allowed 1 1 0 Handsfree 1 1 1 Loudhearing	0100	0101
				Mask	His	Hf	LH			
1	1	1	0	x	x	x	x	**slow_rec:** fast change to TX, slow change to RX, no IDLE **Krat5,Test3,Test2**: For factory test only !	0000	0001
				Krat5	Test3	Test2	slow_rec			
1	1	1	1	x	x	x	x	Reset to defaults	na	na

For a detailed description of the commands see Application note AN523/24.

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) 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	AS2524	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)

Speech mode is Loudhearing (M3-M4, RO active; LO activated).
Loudhearing can be activated via a bond-option (see document “AS2523/24 Delivery as Dice”).

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
Tone	Off	Off
Slow_rec	On	On
RECV pin	Deactivated	Deactivated
Highpass	2000kHz (10nF from TXO->TXV)	2000kHz (10nF from TXO->TXV)
Voice switching speed	120ms/6dB	120ms/6dB
RX-Offset	300mV	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 <= V_{DD} <= 7V
Input Current	+/- 25mA
Input Voltage (LS)	-0.3V <= V_{in} <= 12V
Input Voltage (LI, CS)	-0.3V <= V_{in} <= 8V
Input Voltage (STB, RI)	-2V <= V_{in} <= V_{DD}+0.3V
Digital Input Voltage	-0.3V <= V_{in} <= V_{DD}+0.3V
Electrostatic Discharge (HBM 1.5kΩ-100pF)	+/- 1000V
Storage Temperature	-65°C to +125°C

Recommended operating conditions

Supply Voltage (generated internally)	3V <= V_{DD} <= 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	Type	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	Type	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
AM3/4TX	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 → LS	ZAC(syn)=1000Ω	12	+14dB	16	dB	Y
ΔATX	Variation with frequency	f = 500Hz ... 3.4kHz		±0.8		dB	N
AVRM12	Control range M1/M2 → LS	16 steps		-7/+8		dBr	N
AVRM34	Control range M2/M4 → LS	16 steps		-7/+8		dBr	N
THD	Distortion	VLS=0.25VRMS			2	%	Y
VAGC1	Soft Clip Level M1/M2 → LS at LS			2		VP	N
VAGC2	Soft Clip Level M3/M4 → LS at LS			2		VP	N
ASCO	Soft Clip Overdrive M1/M2 - M3/M4			20		dB	N
tattack	Attack time			70		us/6dB	N
tdecay	Decay time			100		ms/6dB	N

Symbol	Parameter	Conditions	Min	Type	Max	Units	Test
$Z_{IN-M1/2}$	Input Impedance M1/M2			10		kΩ	N
$Z_{IN-M3/4}$	Input Impedance M3/M4			10		kΩ	N
V_{INmax}	Input Voltage Range M1/2 - M3/4	differential		±1		Vp	N
V_{NO}	Noise Output Voltage LS	T_{AMP}=25°C Handset mode Gain = 36.5dB Gain = 42.5dB		 -72	 -66	 dBmp dBmp	 N Y
A_{MUTE}	Mute Attenuation	Mute activated	60			dB	Y

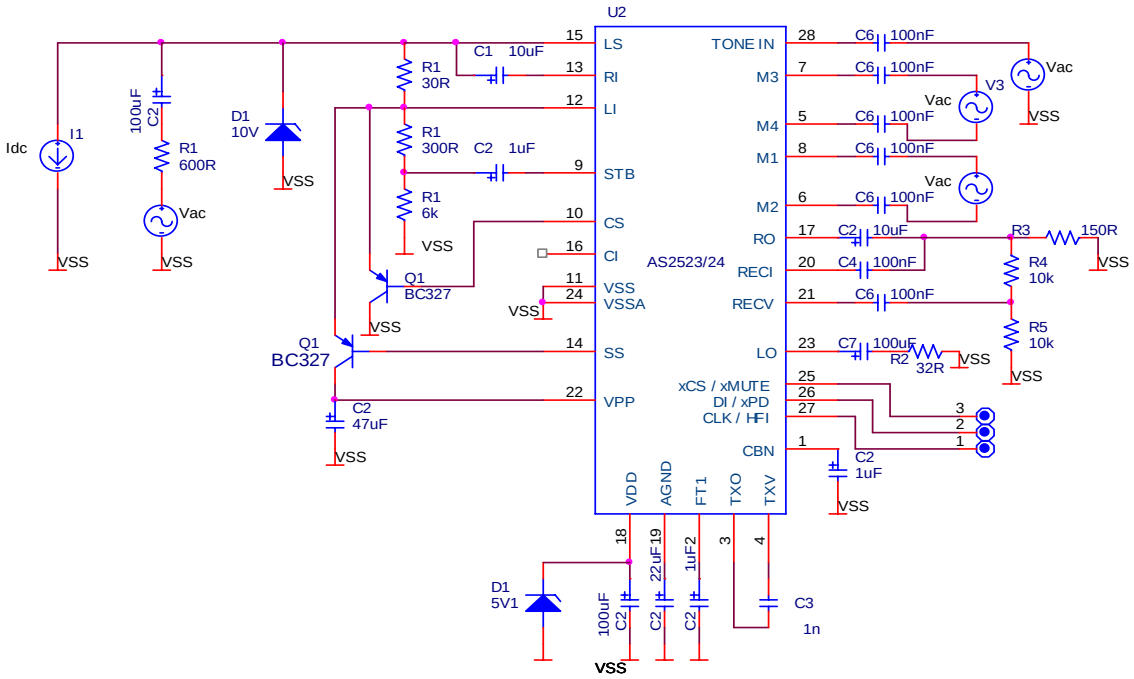
Receive characteristics

I_{Line}=15mA f=800Hz, default settings unless other specified

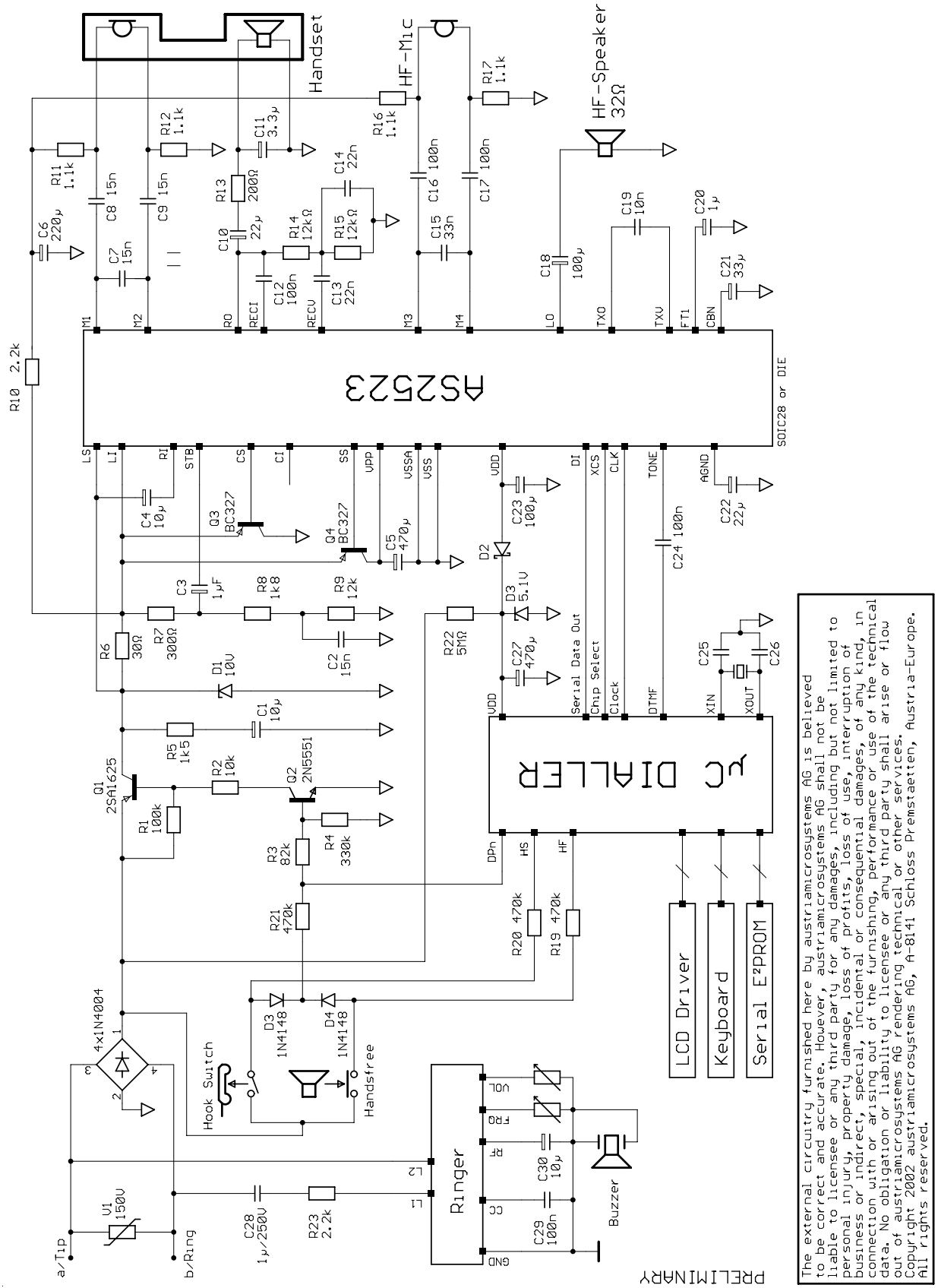
Symbol	Parameter	Conditions	Min	Type	Max	Units	Test
A_{RO}	Receive Gain LS → RO	$Z_{AC(syn)}$=1000Ω Vol default	6	+7.5	9	dB	Y
A_{LO12}	Receive Gain LS → LO	$Z_{AC(syn)}$=1000Ω Vol maximum	+32.0	+34	+36.0	dB	Y
ΔA_{RX}	Variation with frequency	f=500Hz to 3.4kHz		±0.8		dB	N
A_{VRRO}	Control range LS → RO	16 steps		-7/+8		dBr	N
A_{VRLS}	Control range LS → LO	16 steps		-7/+8		dBr	N
THD_{LS}	Distortion LO	V_{LS}=0.25VRMS			5	%	Y
THD_{RO}	Distortion RO	V_{LS} =0.25VRMS			2	%	Y
V_{UFC}	Unwanted Freq. Cmp.	f = 200 ... 20kHz			-60	dBm	Y
V_{AGCRO}	Soft Clip Level RO			1		VP	N
V_{AGCLS}	Soft Clip Level LO			1.1		VP	N
A_{SCLS}	Soft Clip Overdrive LO			10		dB	N
T_{attack}	Attack time			70		us/6dB	N
T_{decay}	Decay time			100		ms/6dB	N

Symbol	Parameter	Conditions	Min	Type	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 ΔZAC/°C	Return Loss Temp. Variation	ZAC(syn)=1000Ω	18	0.5		dB Ω/°C	Y

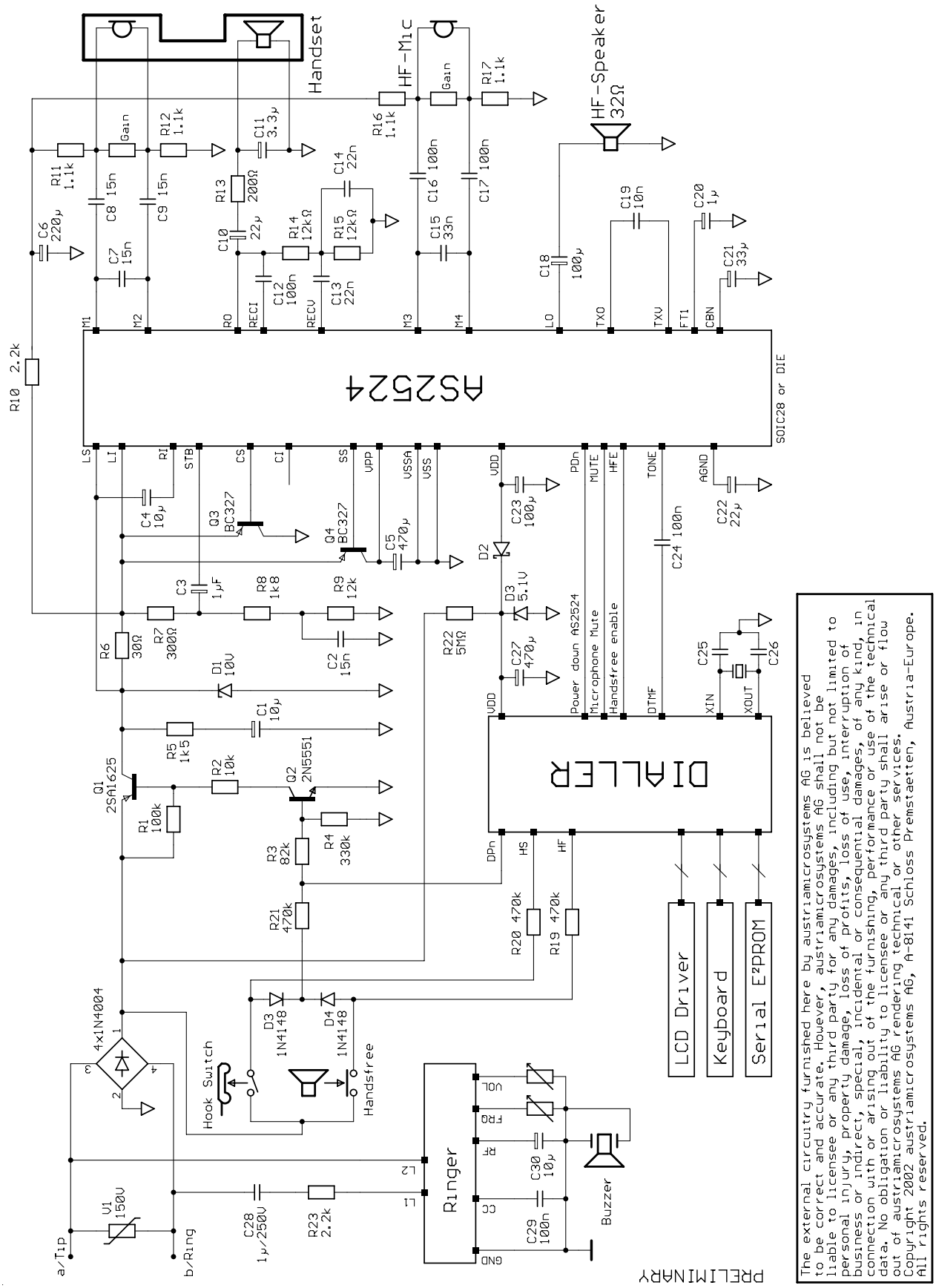
Test circuit



Typical application AS2523



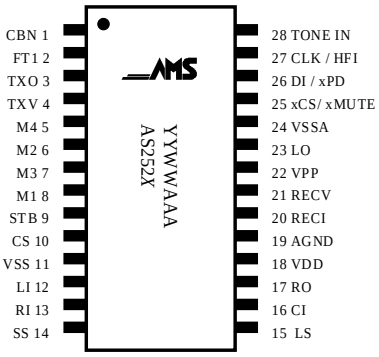
Typical application AS2524 and AS2524B



Packaging

28-pin plastic SOIC
For exact mechanical package dimensions please see austriamicrosystems$_{AG}$ packaging information.

Pin-out, Marking



YY	year of production
WW	calendar week of production
AAA	austriamicrosystems$_{AG}$ assembly ID

X	3 ... AS2523 Version
	4 ... AS2524 Version
	4B .. AS2524B Version

Ordering Information

Number	Package	Description
AS2523 T	SOIC28	plastic surface mounted package – 28 leads
AS2524 T	SOIC28	Plastic surface mounted package – 28 leads
AS2524B T	SOIC28	Plastic surface mounted package – 28 leads
AS2523 F	DOF	Dice-on-Foil
AS2524 F	DOF	Dice-on-Foil
AS2524B F	DOF	Dice-on-Foil

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