

Telephone Line Interface and Speakerphone Circuit

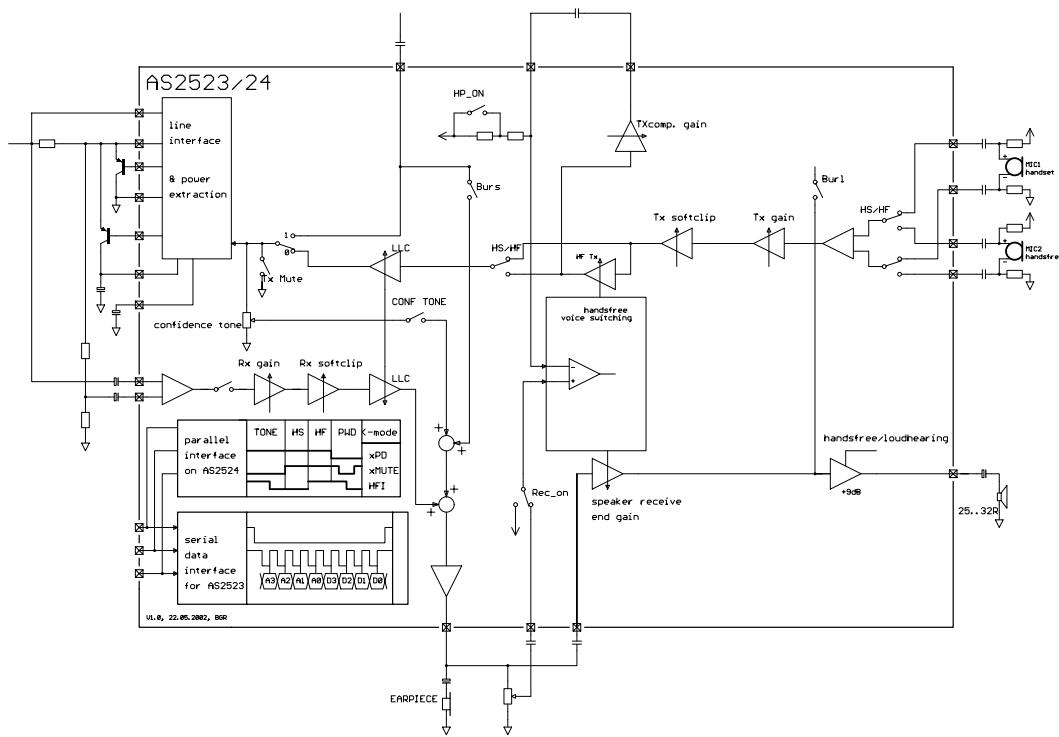
AS2523, AS2524, AS2524B

PRELIMINARY 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 loudearing 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

Block Diagram



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

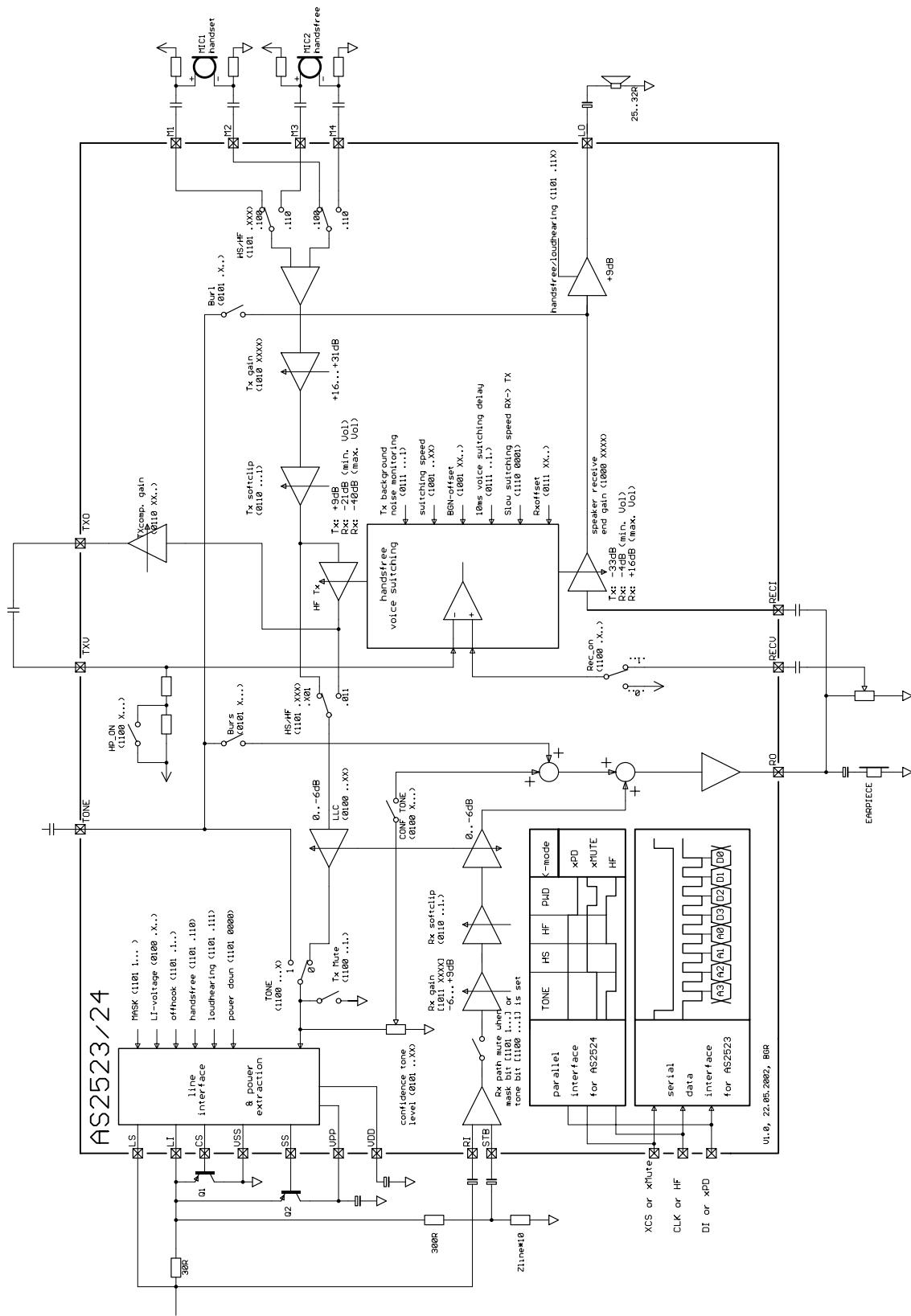
Available in 28-pin SOIC or die

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 /xMUTE | Chip Select on AS2523 Chip select input of the serial interface. Internal pull-up resistor (100kOhm) Mute transmit-, receive- path and activate tone path on AS2524 and AS2524B |
| 26 | DI /xPD | Data Input on AS2523 Data input of the serial interface. Internal pull-up resistor (100kOhm) Power down mode for AS2524 and AS2524B (On-hook) |
| 27 | CLK /HFE | Clock on AS2523 Clock input of the serial interface. Internal pull-up resistor (100kOhm) 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: $V_{LS} = V_{LI} + I_{Line} \cdot 30\Omega$. 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 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. $1.5k\Omega$ 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:

$$ZBAL/ZLINE = 10$$

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 $10k\Omega$. The soft clip circuit limits the output voltage at LS to 2Vp. There is LLC for this path.

The gain of the M3/M4 → 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 → 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 → 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.

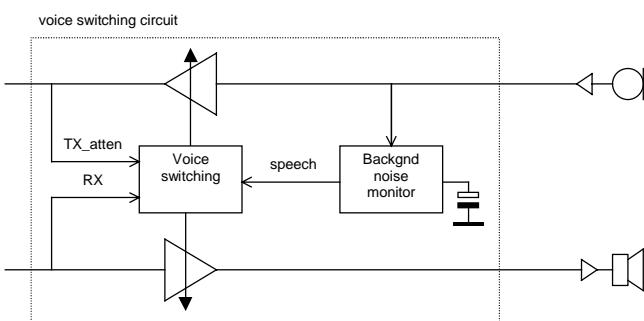
Line Loss Compensation (AS2523, AS2524B)

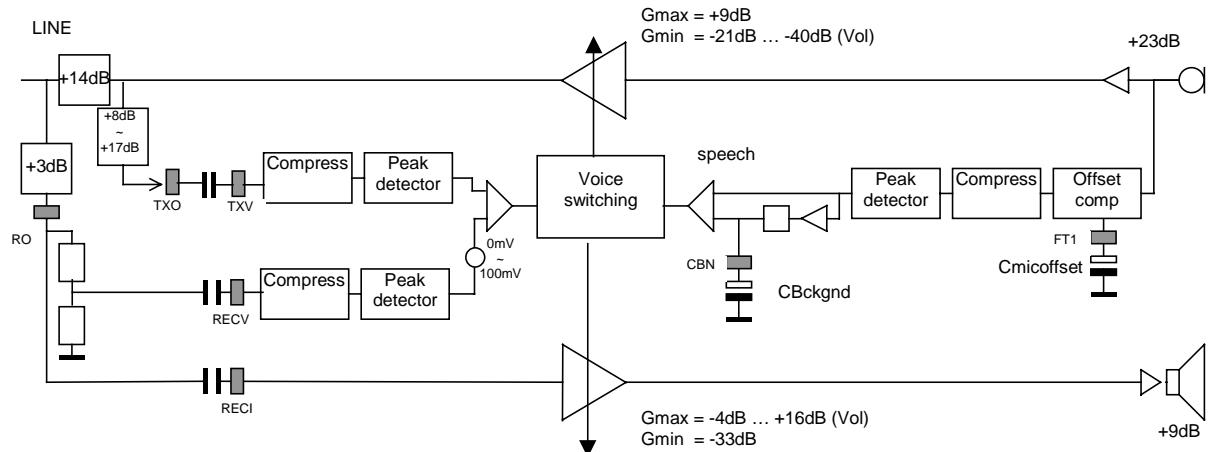
- programmable on AS2523
- not activated at AS2524
- permanently activated at AS2524B

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.

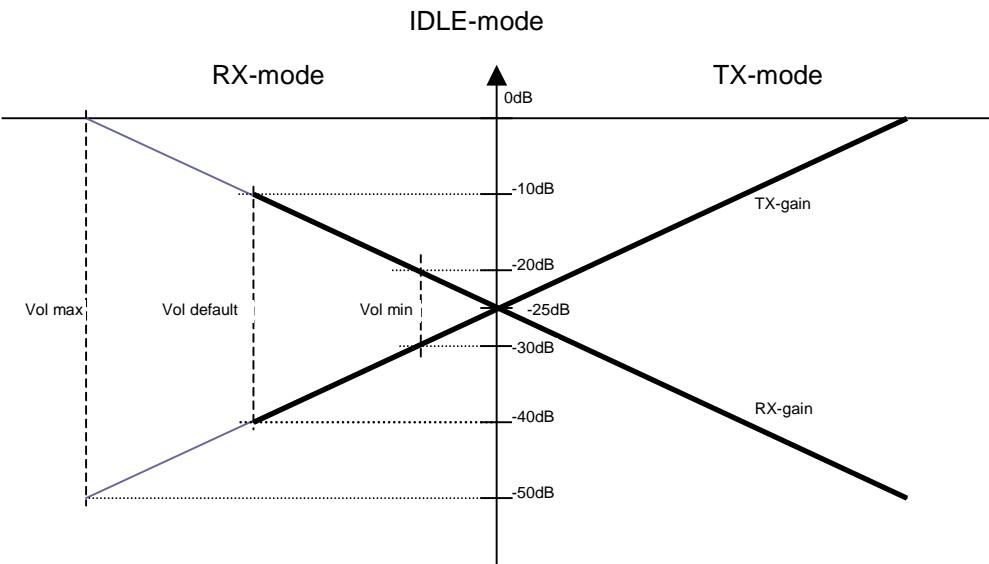
Handsfree

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-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.

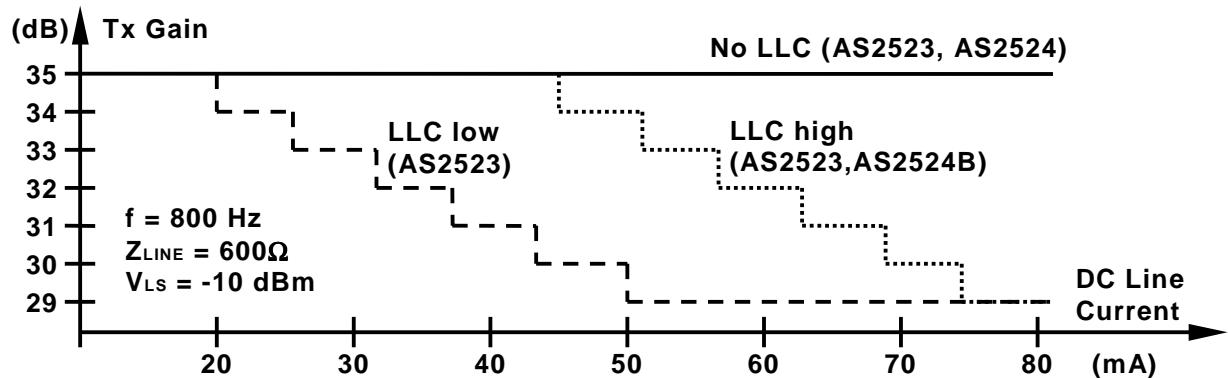


Figure 1 Typical Tx Gain Characteristics Line Loss Compensation

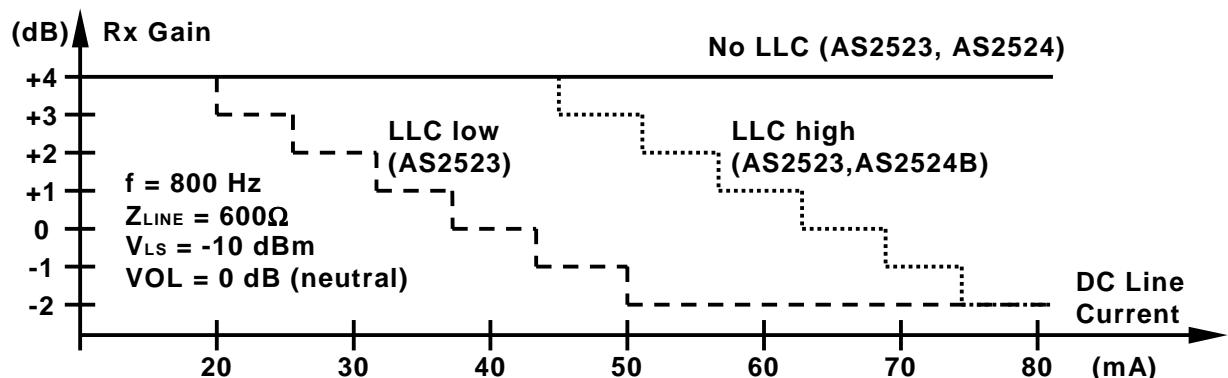


Figure 2 Typical Rx Gain Characteristics Line Loss Compensation

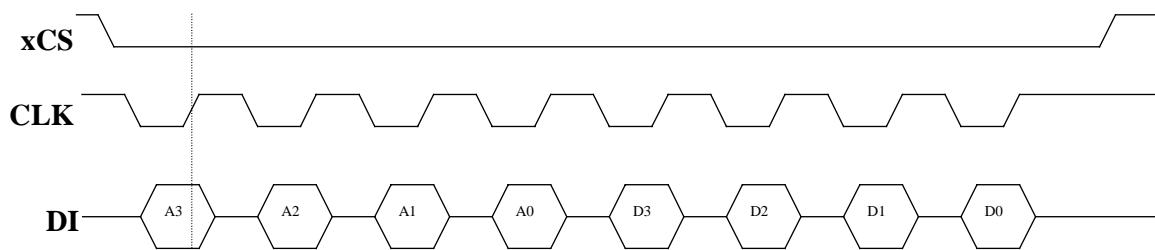
Serial Interface on AS2523

Registers

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

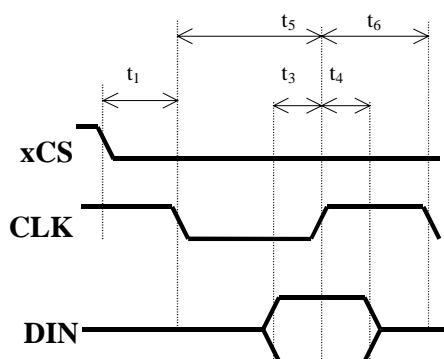
Timing

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 |
|--------------------|---|--|---------------------------|
| A A A A 3 2 1 0 | D D D D 3 2 1 0 | | for AS2523, AS2524HF |
| 0 0 0 0 | Nop | | na |
| 0 0 0 1 | | | na |
| 0 0 1 0 | | | na |
| 0 0 1 1 | | | na |
| 0 1 0 0 | x x x x x | Conf-Tone: 0... No Conf Tone 1... Conf Tone | 0111 |
| | Conf Liv Lls1 Lls0 | 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 |
| 0 1 0 1 | | BURS: Analog tone at RO 0...OFF, 1...ON BURL: Analog tone at LS 0...OFF, 1...ON CT1 CTO: @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 |
| 0 1 1 0 | x 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 |
| 0 1 1 1 | Rxoffset Bit4 Rxoffset Bit0 Delay_10ms Bron | 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 Bmon 0... BGN off 1... BGN on | 0101 |
| 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 |
| 1 0 0 1 | x 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 |
| 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 |
| 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 |
| 1 1 0 0 | HP_on Rec_on Mute Tone | 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 |
| 1 1 0 1 | x x x x x | Mask, Lh, Hf, Hook Switch Mask: 0... No mask 1... Mask activated 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 |
| 1 1 1 0 | Mask His Hf Lh | slow_rec: fast change to TX, slow change to RX, no IDLE Krat5,Test3,Test2: For factory test only ! | 0000 |
| 1 1 1 1 | x x x x x | Reset to defaults | 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 <= 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 1.5kΩ-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

$I_{LINE}=15\text{mA}$ w/o operation of any additional external circuitry, unless other specified

| Symbol | Parameter | Conditions | Min | Type | Max | Units | Test |
|-----------|--|---|------------|------------|------------|--------|--------|
| I_{DDS} | Operating Current | Speech Mode | | 3 | 6 | mA | Y |
| I_{DDH} | Operating Current | Handsfree Mode | | 7 | 10 | mA | Y |
| V_{LI} | Line Voltage L1 Volt=0 Line Voltage L1 Volt=1 | $15\text{mA} \leq I_{LINE} \leq 100\text{mA}$ | 3.2 4.2 | 3.5 4.5 | 3.8 4.8 | V V | Y Y |
| I_{OL} | Output Current, Sink CS,SS | $V_{OL}=0.4\text{V}$ | | 1 | | mA | N |

Transmit characteristics

$I_{LINE}=15\text{mA}$ $f=800\text{Hz}$, default settings unless other specified

| Symbol | Parameter | Conditions | Min | Type | Max | Units | Test |
|-----------------|---|--|----------|----------------|----------|----------|--------|
| AM1/2TX | Transmit Gain M1/M2 → LS AS2524 AS2524B | $Z_{AC(\text{syn})}=1000\Omega$ | 35 39 | +36.5 +42.5 | 38 44 | dB dB | Y Y |
| AM3/4TX | Transmit Gain M3/M4 → LS AS2524 AS2524B | $Z_{AC(\text{syn})}=1000\Omega$ | 43 49 | +45.0 +51 | 47 53 | dB dB | Y Y |
| A tone TX | Transmit Gain Tone → LS | $Z_{AC(\text{syn})}=1000\Omega$ | 12 | +14dB | 16 | dB | Y |
| ΔA_{TX} | Variation with frequency | $f = 500\text{Hz} \dots 3.4\text{kHz}$ | | ± 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 | $V_{LS}=0.25\text{VRMS}$ | | | 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 |
|----------|------------------------------------|--|-----|------|-----|-------|------|
| 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 |
| VNO | Noise Output Voltage LS | TAMP=25°C Handset mode Gain = 36.5dB | | | -72 | dBmp | Y |
| AMUTE | Mute Attenuation | Mute activated | 60 | | | dB | Y |

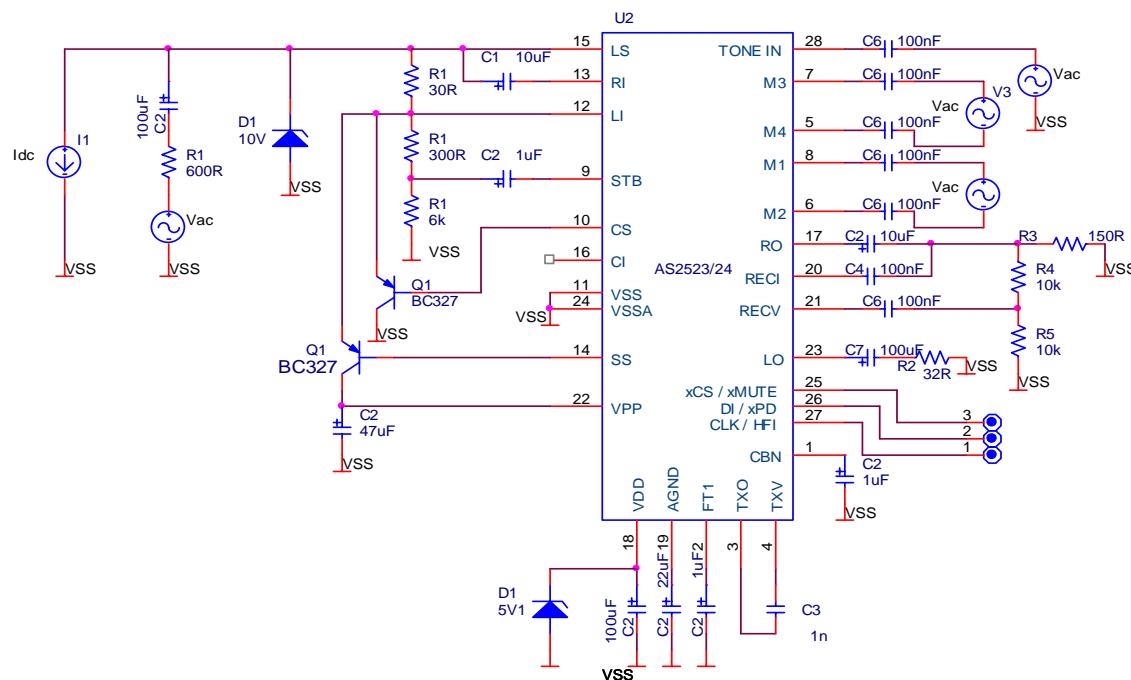
Receive characteristics

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

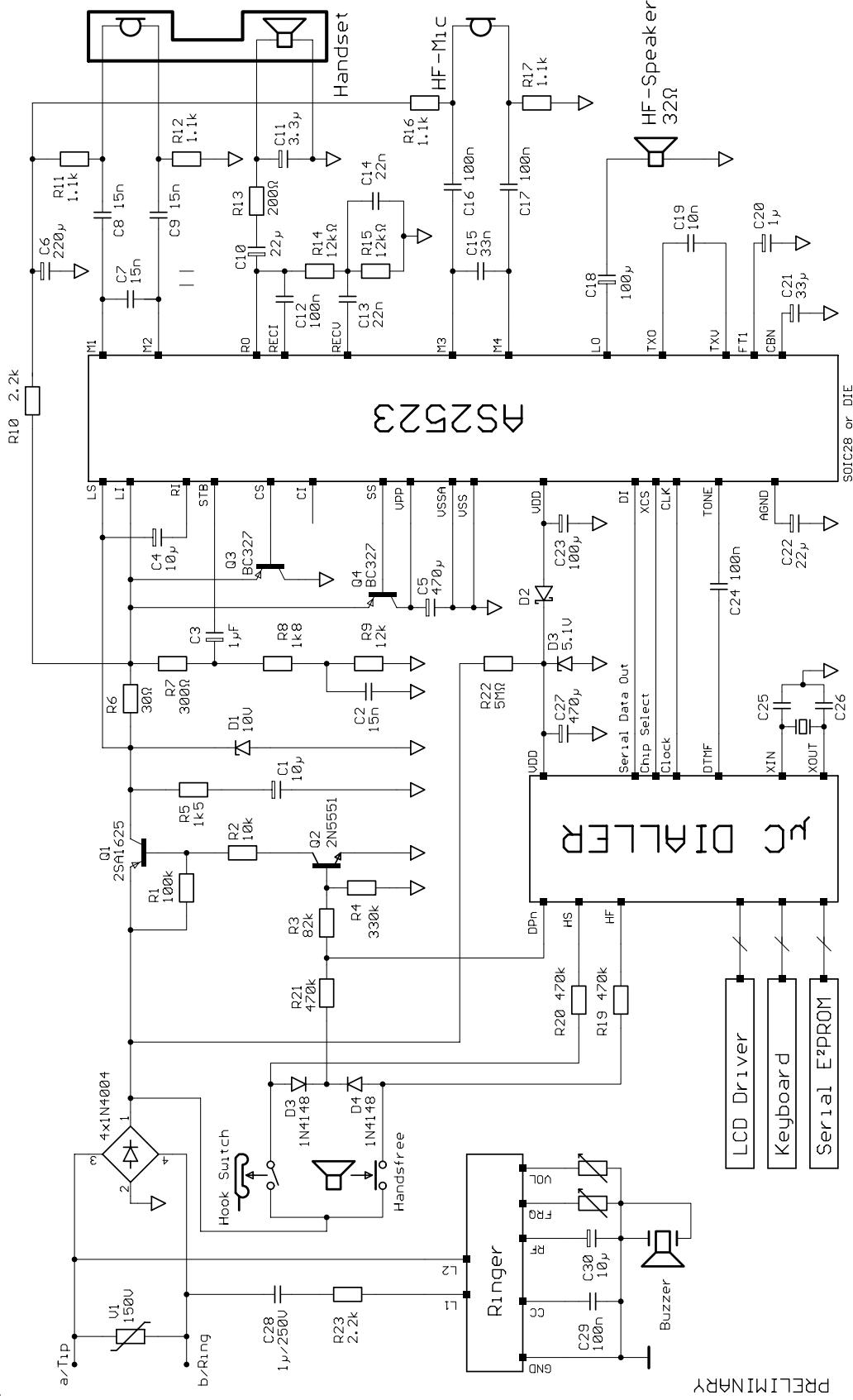
| Symbol | Parameter | Conditions | Min | Type | Max | Units | Test |
|---------|---------------------------|-------------------------------|-------|-------|-------|--------|------|
| ARO | Receive Gain LS → RO | ZAC(syn)=1000Ω Vol default | 6 | +7.5 | 9 | dB | Y |
| AL012 | Receive Gain LS → LO | ZAC(syn)=1000Ω Vol maximum | +32.0 | +34 | +36.0 | dB | Y |
| ΔARX | Variation with frequency | f=500Hz to 3.4kHz | | ±0.8 | | dB | N |
| AVRRO | Control range LS → RO | 16 steps | | -7/+8 | | dBr | N |
| AVRLS | Control range LS → LO | 16 steps | | -7/+8 | | dBr | N |
| THDLS | Distortion LO | V _{LS} =0.25VRMS | | | 5 | % | Y |
| THDRO | Distortion RO | V _{LS} =0.25VRMS | | | 2 | % | Y |
| VUFC | Unwanted Freq. Cmp. | f = 200 ... 20kHz | | | -60 | dBm | Y |
| VAGCRO | Soft Clip Level RO | | | 1 | | VP | N |
| VAGCLS | Soft Clip Level LO | | | 1.1 | | VP | N |
| ASCLS | Soft Clip Overdrive LO | | | 10 | | dB | N |
| Tattack | Attack time | | | 70 | | us/6dB | N |
| Tdecay | Decay time | | | 100 | | ms/6dB | N |

| Symbol | Parameter | Conditions | Min | Type | Max | Units | Test |
|-------------------|---|---------------------------|-----|----------|-----|------------|------|
| VNO RO | 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

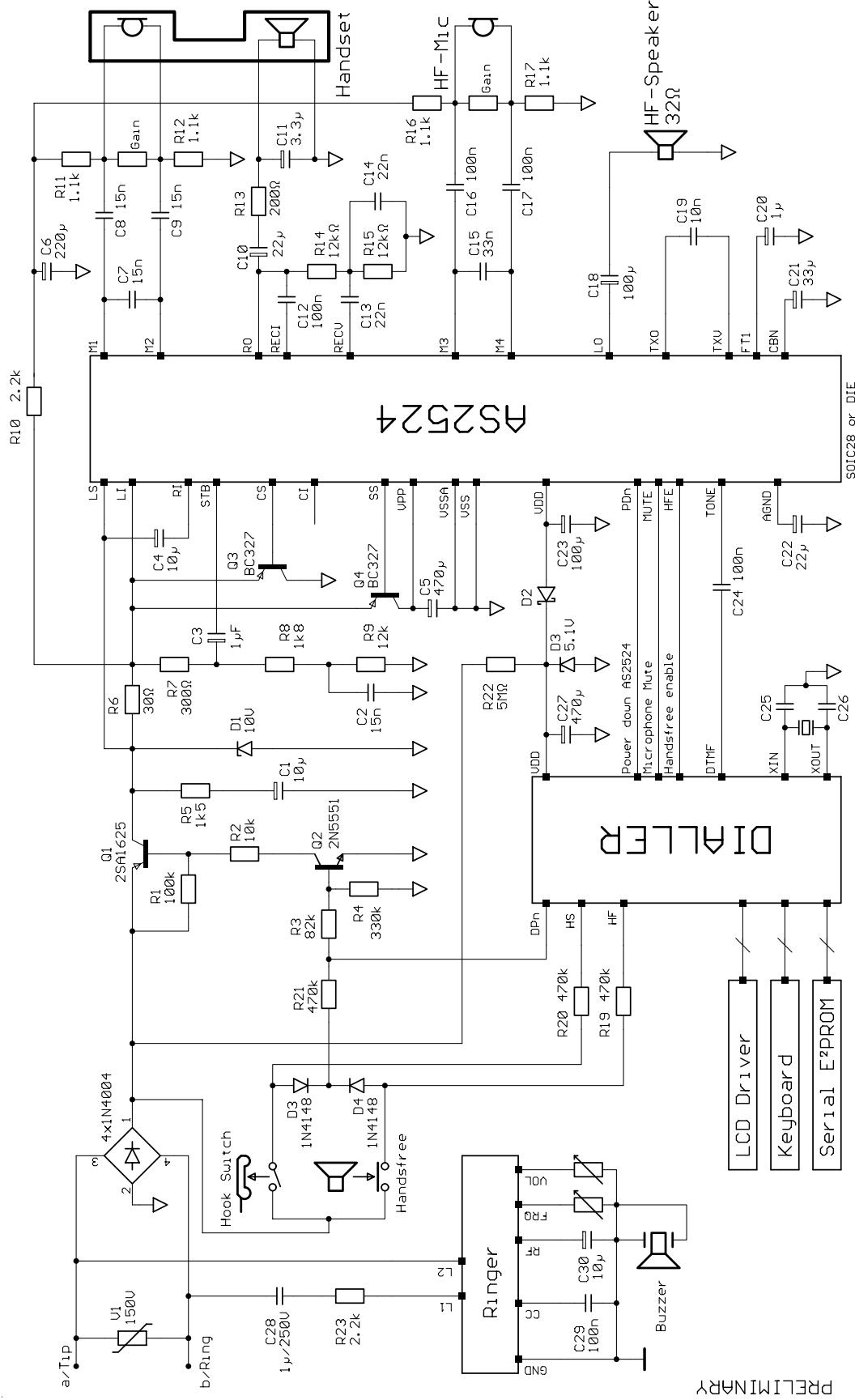


Typical application AS2523



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Typical application AS2524 and AS2524B



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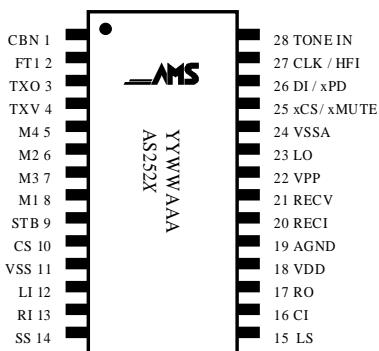
Austria-Europe.

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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