

## Multi-Standard Feature Phone IC

### Description

The U3810BM multi-standard feature phone circuit is designed to be used with a microcontroller using a 2-wire serial bus. It performs all speech and line interface functions required in an electronic telephone set: the ringing function with switching regulator and melody generator, the DTMF dialling, the loudhearing with antilarsen and antidistortion systems, a power supply, a clock and a reset for the microcontroller. Transmit,

receive and loudhearing gains control / AGC range / DTMF frequencies, preemphasis and level / melody generator, and mutes are programmable through the serial bus.

Electrostatic sensitive device.  
Observe precautions for handling.



### Applications

- Feature phones
- Answering machines
- Fax machines

### Benefits

- Complete system integration of analog signal processing and digital control circuitry
- One IC for various PTT standards, e.g. programmable specification via  $\mu$ C
- Only three low-cost transducers needed (instead of four)

### Block Diagram

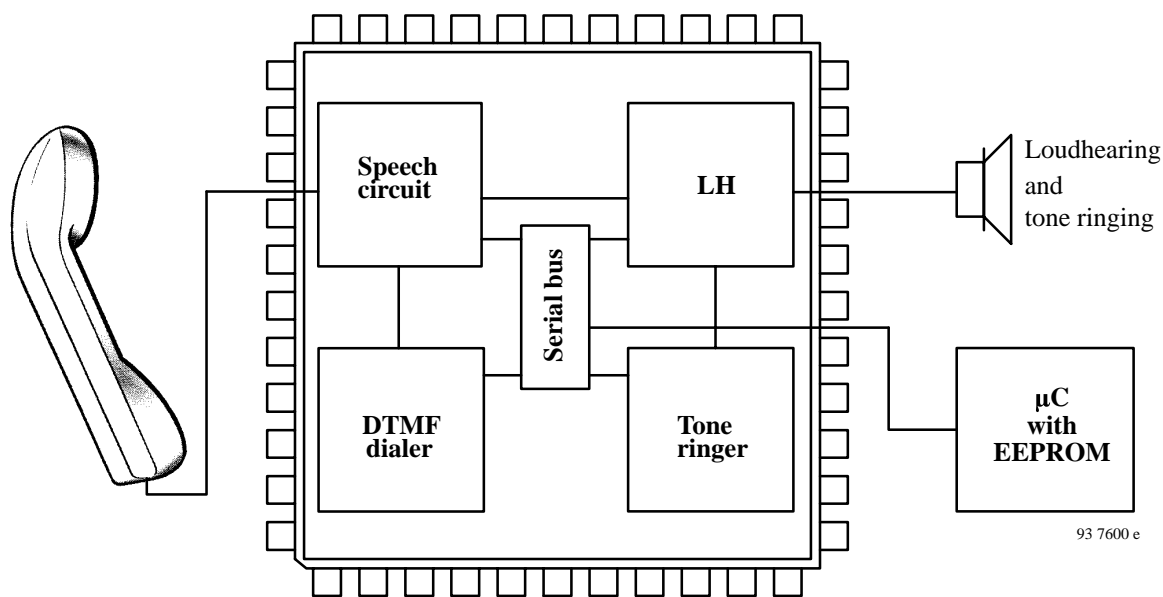


Figure 1. Block diagram

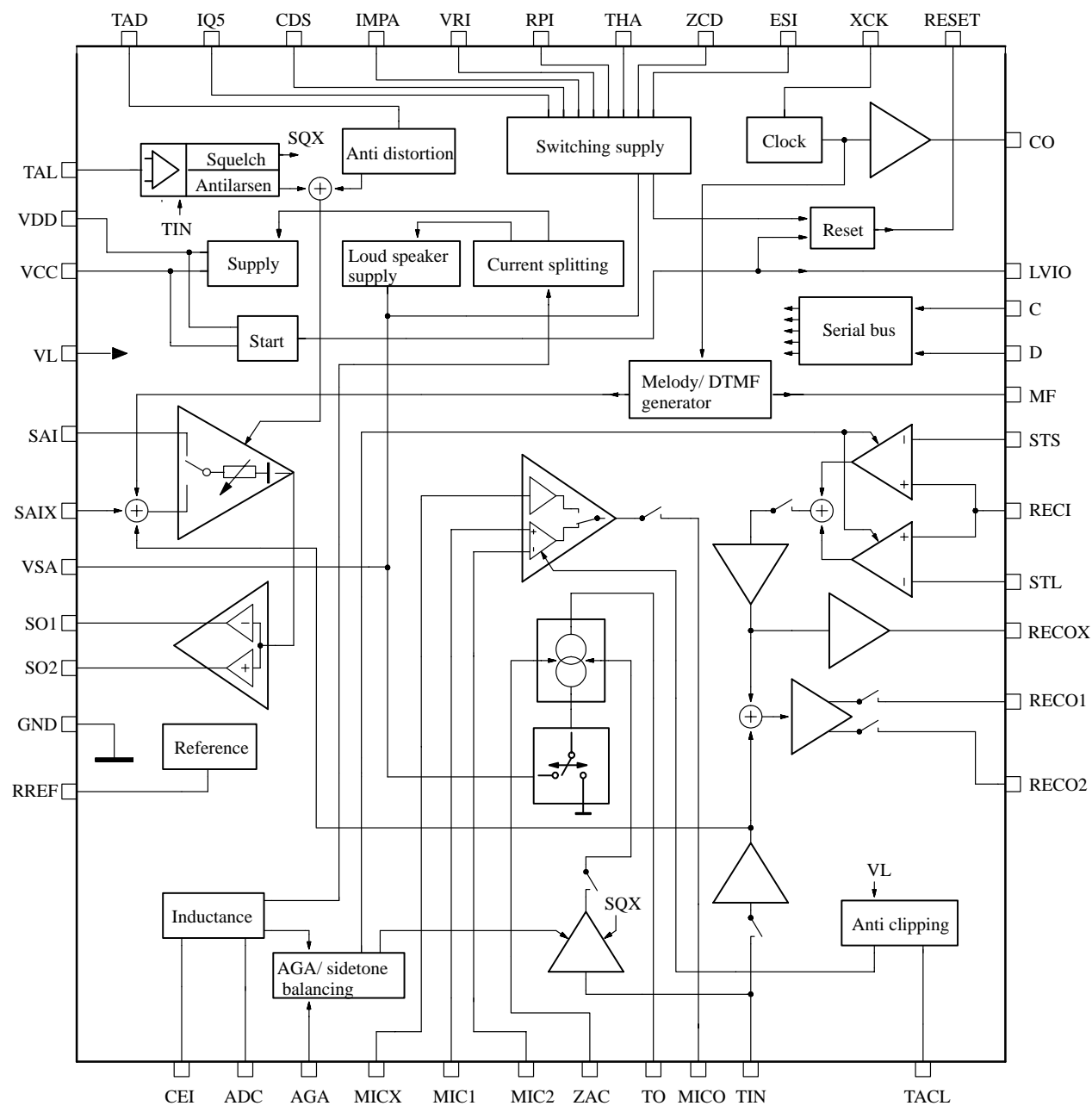
### Ordering Information

Extended Type Number	Package	Remarks
U3810BM-CP	PLCC44	
U3810BM-FN	SSO44	

## Features

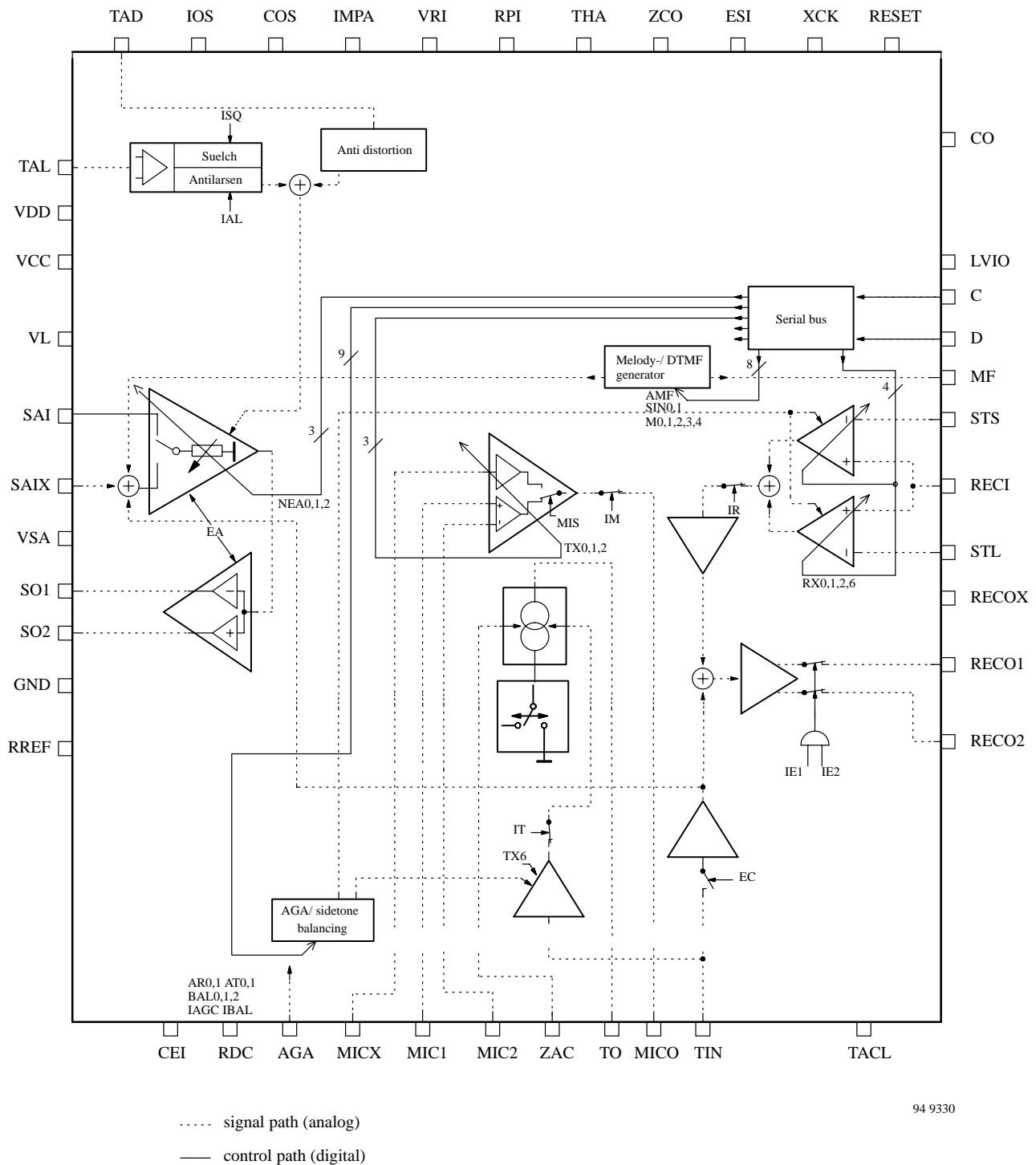
- Slope of DC characteristics adjustable by an external resistor
- Gain of transmit and receive amplifiers automatically adjusted by line-length control
- Regulation range adjustable by the serial bus
- Possibility of fixed gain (PABX)
- Sidetone balancing system adjustable with line length or by the serial bus
- Dynamic impedance adjustable by external components
- Stabilized power supply for peripherals
- Confidence level during dialing
- +6 dB possibility on second stage transmit gain
- Transmit and receive gains adjustable by serial bus
- Extra transmit input for hands-free and answering machine purpose
- +6 dB possibility on receive gain
- Receive amplifier for dynamic or piezo-electric earpieces
- Extra receive output for hands-free purpose
- High-impedance microphone inputs suitable for dynamic, magnetic, piezo-electric or electret microphone
- Distortion of line signal and sidetone prevented by dynamic range limitation in transmit direction (anti-clipping)
- Squelch system in transmission mode prevents “room noise” being transmitted, and improves antilarsen efficiency (can be inhibited).
- Loudhearing gain programmable in eight steps of 4 dB, using the serial bus, or linearly adjusted, using a potentiometer
- Antilarsen system efficiency is increased when inhibiting squelch
- Loudhearing antidistortion system by automatic gain control versus available current and voltage
- Switching regulator in ringing phase
- Input ringing detection, threshold and impedance adjustable with external resistors
- Ringing zero crossing information for external microprocessor
- Ringing programmable gain in eight steps of 4 dB using the serial bus
- Melody generator, with 30 frequencies in steps of semi tones, driven by serial bus
- Internal speed-up circuit permits a faster charge of VDD and VCC capacitor
- DTMF dialer driven by serial bus, in particular level and preemphasis adjustment
- Ability to transmit a confidence tone in speech mode using melody generator frequencies
- Four independent mutes driven by serial bus (two for transmission mute and two for reception mute)
- Standard low-cost 455-kHz ceramic / clock output for the microcontroller
- Extra loudhearing input for answering machines, hands-free and base station of cordless phones with loudhearing

## Detailed Block Diagram



94 9329

Figure 2. Detailed block diagram



94 9330

Figure 3. Digital adjustment of the analog parameters by the serial-bus microprocessor interface



Figure 4. Application for feature phone

## Typical Value of External Components

Components	Min.	Typ.	Max.
RIN	0.3 MΩ	1.0 MΩ	1.5 MΩ
CSC		0.1 μF	
TSC		2N5401A	
LSC		1 mH	
DSC		SD103A	
RAD		100 kΩ	
CAD		470 nF	
CAL		470 nF	
RAL		68 kΩ	82 kΩ
CVDD		470 μF	
RSAI		20 kΩ	
CSAI		0.1 μF	
CVSA		220 μF	
CSO		47 μF	
CVCC		100 μF	
RAGA1		100 kΩ	
RAGA2		51 kΩ	
CEI		0.47 μF	
RDC		20 kΩ	
RTO		62 Ω	
(CTO		0.33 μF)	
CZAC1		0.47 μF	
RZAC		12 kΩ	

Components	Min.	Typ.	Max.
(CZAC2		470 pF)	
RDTMF1		240 kΩ	
RDMTF2		330 kΩ	
CDTMF1		4.7 nF	
CDTMF2		220 pF	
CACL		0.47 μF	
RACL		6.8 MΩ	
CEAR		4.7 μF	
Earphone		~1 kΩ	
Loudspeaker		~100 Ω	
RSTL1		6.2 kΩ	
(CSTL1		330 pF)	
RSTS1		6.2 kΩ	
(CSTS1		330 pF)	
RSTL2		43 kΩ	
RSTS2		18 kΩ	
RSTL3		100 kΩ	
RSTS3		75 kΩ	
CSTL3		10 nF	
CSTS3		10 nF	
CSTS2		470 pF	
CSTL2		1.2 nF	
X1		CSB455E (Murata)	

## Pin Description

Pin	Symbol	Function
1	RPI	Ring power information: The RC combination smooths the drive current of the loudspeaker amplifier.
2	VRI	Tone-ringer supply voltage: The rectified ringing voltage is delivered to VRI and then converted into the lower supply voltage, VSA, by the converter.
3	IMPA	External adjustment of input ringing impedance: IMPA is adjusted with a resistance between Pins 2 and 3. $ZIN = RIN/100$
4	COS	Control output switching supply: COS drives the base of the external switching transistor of the converter.
5	IOS	Current output of switching supply: This output provides a constant current which supplies the external part of converter. The magnitude of the current depends on the VRI voltage and the value of resistance RIN.
6	TAD	Adjustment of anti-distortion time constant in loudhearing with external RC combination
7	TAL	Adjustment of anti-larsen time constant in loudhearing with external RC combination
8	VDD	External logic supply
9	VCC	Power supply for peripherals: VCC and VDD are stabilized supply voltages buffered with external capacitors. They are derived internally from the same voltage source, but are separated from each other by electronic switches. VDD also supplies the digital part of the circuit and is achieved in all three modes: speech mode, ringing mode and operation with external supply. According to the application, peripheral modules are connected to VDD which, in addition to speech mode, must be supplied at least in one of the two other modes. The digital part of circuit and microprocessor must continue operating during line breaks, as they occur during pulse dialing or during flash-signal transmission. Since VDD in this time intervals is fed only from the buffer capacity, the power consumption from VDD must not cause the total voltage dump. VCC supplies no internal parts of the circuit and is supplied exclusively in speech mode. External components, which must operate only in this mode, can be connected here. The power is drawn only from the relevant buffer capacitor in the supply intervals during pulse dialing or flash-signal transmission.
10	VL	Line voltage
11	SAI	Speaker amplifier input: The signal coming from the receive part, e.g., from RECO1 or RECO2, is fed in here.
12	SAIX	Speaker amplifier input for special application, i.e., answering machine: SAIX is selected via the serial bus. SAIX has to be selected for ringing, if ringing, anti-distortion and anti-larsen are disabled.
13	VSA	Supply voltage for the loudhearing amplifier: Stabilized supply voltage buffered with an external capacitor for the loudhearing amplifier and zero crossing detector; additional connection point for an external supply. In speech mode, VSA is supplied by the analog part of the circuit. In this case, the stabilization point is adjusted to the DC line voltage $VSA = VL/1.5$ . In ringing mode, VSA is supplied directly from the converter. The stabilization point is permanently set. The logic supply, VDD, is fed by a switch from VSA. $VSA = 5.2 \text{ V}$ . In the case of an external supply, VSA serves as a point for a current from a power supply. The stabilization point of VSA and supply for the logic correspond to the operation conditions in ringing mode.

## Pin Description (continued)

Pin	Symbol	Function
14	SO1	Loudspeaker output 1
15	SO2	Loudspeaker output 2 (differential output of loudspeaker amplifier): The loudspeaker can also be connected asymmetrically to ground or to VSA via a capacitor at one of the two outputs. As a result of this connection, there is a bigger output power in reference to low line currents and loudspeakers with low impedance, while the differential connection method results in a higher power output and lower harmonic distortion with medium or high line currents and/or loudspeaker impedances.
16	GND	Ground
17	RREF	External reference resistor (connection for external reference resistor to generate the reference current): All basic currents of the circuit which must satisfy certain absolute accuracy requirements depend on this current.
18	CEI	Capacitor for electronic inductance (connection for capacitor of the electronic coil) The circuit contains a first-order RC active low-pass filter. The capacitor is connected externally between CEI and VL.
19	RDC	DC characteristic slope adjustment: A voltage across resistor RDC is proportional to the DC line voltage. This means the current flow through RDC is also proportional to the line voltage. This current drives the supply currents drawn from VL by the most important loads, and therefore defines the total current consumption of the circuit. Adjustment to the slope characteristic is realized by modification of RDC resistance.
20	AGA	Line-length adjustment (reference voltage level for AGA) The potential at this point defines the start threshold for the AGA and the automatic balancing in the receive part (both can be switched off by the serial bus). The potential is normally formed between VCC and ground using a voltage divider. When the line voltage exceeds the threshold level, the AGA or balancing becomes effective.
21	MICX	Asymmetrical microphone input for special applications: The input of the first stage of the transmit amplifier, selected by the serial bus. Anti-clipping is not effective at this input.
22	MIC1	Inverting input of microphone amplifier
23	MIC2	Non-inverting input of microphone amplifier
24	ZAC	AC impedance adjustment: Adjustment of the AC circuit impedance to the line by changing of RZAC.
25	TO	Transmit amplifier output: The transmit amplifier output modulates the current flowing into this output (typ. 4.8 mA).
26	MICO	Microphone amplifier output, output MICO; open-circuit potential = $2 \times V_{be}$
27	TIN	Transmit and DTMF input, input of the second transmit stage.
28	TACL	Adjustment of anti-clipping time constant with external RC combination Anti-clipping controls the transmitter input level to prevent clipping with high signal levels. The dynamic range of the transmit peak limiter is controlled by an internal circuit.
29	RECO2	Symmetrical output of receive amplifier
30	RECO1	Symmetrical output of receive amplifier
31	RECOX	Receive amplifier output for hands-free and answering machine applications
32	STL	Long-line sidetone network
33	RECI	Receive amplifier input: It is driven by a signal from VL.
34	STS	Short-line sidetone network



## Pin Description (continued)

Pin	Symbol	Function
35	MF	Multifrequency output. Output DTMF signal and confidence tone in pulse-density-modulated form. DTMF signal and confidence tone are generated by special generators in the digital part of the circuit. The DTMF signal consists of two weighted and superimposed pulse-density-modulated signals, while in the case of confidence tone, a pulse-density-modulated signal is superimposed on a high-frequency rectangular-pulse signal with pulse duty factor 0.5. The superimposed signals are sent out to MF for further processing.
36	D	Data input of serial bus (Pin 37): Serial data input from microprocessor for programming the circuit.
37	C	Clock line: 2-wire serial bus: This pin is used together with the data input (Pin 36) to transfer data from the microprocessor to the circuit. The last 8 bits, consisting of 5 data bits and 3 address bits, are accepted by the circuit if, during the high phase at Pin 37, a positive edge on Pin 36 takes place.
38	LVIO	Line-voltage information output State indicator of the line voltage, VL, and the supply voltage, VDD Indicates to microprocessor whether line voltage VL is present across the circuit and the supply voltage VDD is sufficiently high. If both conditions are fulfilled, LVIO is on high level.
39	CO	Clock output: Output 455-kHz clock pulse for the microprocessor. The 455-kHz clock signal generated in the circuit is amplified and sent to CO. This signal is delivered to the microprocessor as a clock signal. Various internal signals can be output to CO in test mode.
40	RESET	Reset signal for periphery: A reset signal (active low) is generated to clear all registers of the circuit. This can be tapped by peripheral modules, particularly by the microprocessor. This pin enables synchronous resetting of the circuit and peripheral modules.
41	XCK	Clock signal generator, connector for ceramic resonator (455 kHz) The clock for the digital part of the circuit is generated by a one-pin oscillator, the frequency of which is determined by the ceramic resonator.
42	ESI	Input for external supply information Input to indicate the operating state external supply. In the case of VSA supplied by a power supply unit, the supply source is connected directly to ESI. ESI is connected to VSA by an external diode in forward mode. The ESI voltage will be one forward voltage higher than the voltage on VSA. This voltage causes the circuit to be in external supply mode.
43	ZCO	Zero crossing output in ringing phase: ZCO operates when the ringer voltage, VRI, and the supply voltage, VSA, are sufficiently high. The output voltage ZCO changes state each time the rectified AC signal of THA crosses the ringing detect turn-on and turn-off thresholds, thus providing information on the frequency of the ring signal. Further analysis of the ring frequency is to be done by the microprocessor. Secondly, this pin is used as an input for switching on the test mode. Therefore, a negative voltage of approximately 1 V must be applied to Pin 43.
44	THA	Ringing detection threshold adjustment (input amplitude and frequency identification): The rectified ringing voltage is present at this pin. The circuit evaluates the amplitude of the rectified voltage at THA and the supply voltage VSA (Pin 13). If both voltages exceed certain thresholds, the signal present at THA is converted to a rectangular-pulse signal and is sent via Pin 43 to the microprocessor. If the frequency is in the required range, the microprocessor initiates transmission of the ringing signal. The start threshold can be raised with a resistor connected in series to Pin 44.

## Absolute Maximum Ratings

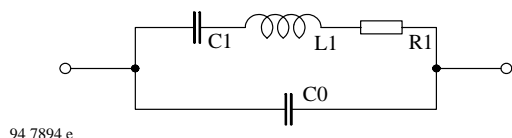
Parameters	Symbol	Value	Unit
DC calling voltage (Pin 2)	VRI	35	V
DC calling current (Pin 2)	IR	30	mA
Conversation line voltage (Pin 10) Voltage pulse for $t \leq 20$ ms	V <sub>L</sub>	15 17	V V
Conversation line current	I <sub>L</sub>	150	mA
Total power dissipation (assembly on PC board $\geq 24$ cm <sup>2</sup> )	P <sub>tot</sub>	1	W
Operating temperature range	T <sub>amb</sub>	-25 to +55	°C
Storage temperature range	T <sub>stg</sub>	-55 to +150	°C
Junction temperature	T <sub>j</sub>	125	°C

## Thermal Resistance

Parameters	Symbol	Value	Unit
Junction ambient (for assembly on PC board $\geq 24$ cm <sup>2</sup> )	R <sub>thJA</sub>	70	K/W

## Electrical Characteristics

I<sub>L</sub> = 28 mA, T<sub>amb</sub> = 25°C, f = 1 kHz, R<sub>DC</sub> = 20 kΩ, all internal registers cleared, unless otherwise specified



- 455-kHz ceramic resonator: MURATA or equivalent
- Refer to the test circuits

Figure 5. Resonator equivalent circuit

Resonance factor  $Q_m = 3100$ ,  $L1 = 6.1$  mH,  $C1 = 21$  pF,  $CO = 268.5$  pF,  $R1 = 5.5$  Ω (Schematic above)

All resistances are specified at 1%, all capacitances at 2%.

Parameters	Test Conditions	Min.	Typ.	Max.	Unit	Fig.
Line voltage	I <sub>L</sub> = 15 mA I <sub>L</sub> = 28 mA I <sub>L</sub> = 60 mA	4.2 6.7 12.8	4.75 7.2 13.45	5.2 7.5 14.1	V	6
VDD, VCC stabilized power supply	I <sub>L</sub> = 8 mA, -I <sub>dd</sub> (ICC) = 0.6 mA I <sub>L</sub> = 28 mA, -I <sub>dd</sub> (ICC) = 2.3 mA	2.5 3.2	2.65 3.45	3.6	V	6
IDD at VDD = 3.5 V Internal operating supply current	S4 on 3		180	210	μA	9
Leakage current				100	nA	
Speed-up off threshold VSOFF	See figure 11 I <sub>Lmax</sub> = 80 mA V <sub>L</sub> = 4 V	2.45	2.65	2.8	V	7
Speed-up on line-current ISON	See figure 14 I <sub>L</sub> decreasing V <sub>DD</sub> = 2.8 V	5.0	5.9	7.5	mA	
Speed-up current	V <sub>L</sub> = 4 V	40	70		mA	7

## Electrical Characteristics – Transmission

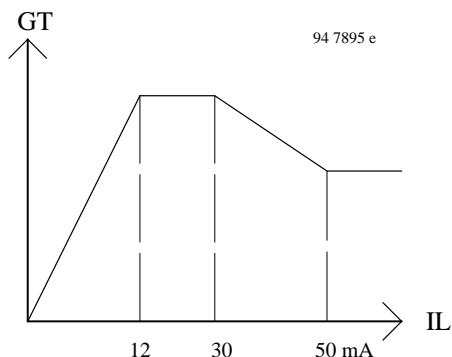
Parameters	Test Conditions	Min.	Typ.	Max.	Unit	Fig.
Transmit gain (note 1)	$V_{MIC} = 3 \text{ mV}_{rms}$					
on MIC1 / MIC2	MIS TX2 TX1 TX0 max. gain 0 1 1 1 min. gain 0 0 0 0	47.0 39.8	48.0 41.0	49.0 42.2	dB dB	6
on MICX	$V_{MICX} = 5 \text{ mV}_{rms}$ MIS TX2 TX1 TX0 max. gain 1 1 1 1 min. gain 1 0 0 0	43.0 35.8	44.0 37.0	45.0 38.2	dB dB	6
Gain adjustment of microphone amplifier	Gain change between two steps (both on MIC1/MIC2 and on MICX)	0.8	1.0	1.2	dB	6
Transmit gain without AGC $G_T$ at 28 mA	$V_{MIC} = 3 \text{ mV}_{rms}$ , Bit AGC=1					
$\Delta G_T$ at $I_L = 20$ to 28 mA	TX2 TX1 TX0 0 1 1	42.6	44.0	45.5	dB	6
$\Delta G_T$ at $I_L = 28$ to 60 mA	0 1 1 0 1 1	-0.5 -0.5	0.0 0.0	0.5 0.5	dB	6
Gain change between 28 and 60 mA on MIC1 / MIC2	$V_{MIC} = 3 \text{ mV}_{rms}$ : AT1 AT0 (MIS = 0) 0 0 0 1	3.6 4.9	4.1 5.35	4.6 5.9	dB	6
on MICX	or $V_{MICX} = 5 \text{ mV}_{rms}$ 1 0 (MIS = 1) 1 1	6.3 7.7	6.9 8.2	7.4 8.7	dB	6
+6 dB delta transmit gain	$I_L = 28 \text{ mA}$ and 60 mA $V_{MIC} = 1.5 \text{ mV}_{rms}$ , Bit TX6 = 1	5.4	6.0	6.6	dB	6
Noise at line psophometrically weighted	GT = max. gain $V_{MIC} = 0$ (MIS = 0) ISQ = 0 $V_{MIC} = 0$ (MIS = 0) ISQ = 1 $V_{MICX} = 0$ (MIS = 1) ISQ = 1		-79 -73	-75 -70 -64	dBmp	6
Muted gain	$V_{MIC} = 3 \text{ mV}_{rms}$ (MIS = 0) or $V_{MIC} = 5 \text{ mV}_{rms}$ (MIS = 1) (at max. and min. gain) *)					
on MIC1 / MIC2	Bit IM = 1 Bit IT = 1	65 65			dB dB	6
on MICX	$V_{MIC} = 3 \text{ mV}_{rms}$ (MIS = 1) or $V_{MIC} = 5 \text{ mV}_{rms}$ (MIS = 0)	60 60			dB dB	6
Microphone input impedance						
on MIC1/MIC 2	$V_{MIC} = 3 \text{ mV}_{rms}$ (MIS = 0)	70	110		k $\Omega$	6
on MICX	$V_{MICX} = 5 \text{ mV}_{rms}$ (MIS = 1)	35	55		k $\Omega$	6
CMRR common mode rejection ratio	$G_T =$ at maximum gain *)		65		dB	6
Voltage step on Pin 26 when going from transmission to mute mode	$I_L = 28 \text{ mA}$ and 60 mA $V_{MIC} = 0$ (MIS = 0) or $V_{MICX} = 0$ (MIS = 1)	-75		+75	mV	6
Bit IM from 0 to 1	At maximum gain *)					

## Transmission (continued)

Parameters	Test Conditions	Min.	Typ.	Max.	Unit	Fig.
<b>Dynamic limiter (anti-clipping) <math>CACL = 470 \text{ nF}</math>, <math>RACL = 6.8 \text{ M}\Omega</math>, operational only on MIC1/MIC2</b>						
Output voltage swing (peak-to-peak value)	$V_{MIC} = 4 \text{ mV}_{rms} + 10 \text{ dB}$ TX2 TX1 TX0 1 1 1	3.3		4.4	$V_{pp}$	6
Delta output voltage swing	$V_{MIC} = 9 \text{ mV}_{rms} + 10 \text{ dB}$ TX2 TX1 TX0 0 0 0	-200	0	200	$\text{mV}_{pp}$	6
	$I_L = 60 \text{ mA}$ , $AT1 = AT0 = 1$ $V_{MIC} = 22 \text{ mV}_{rms} + 10 \text{ dB}$ TX2 TX1 TX0 0 0 0	-200	0	200	$\text{mV}_{pp}$	6
Line distortion (on $600 \Omega$ )	$V_{MIC} = 9 \text{ mV}_{rms}$ $V_{MIC} = 9 \text{ mV}_{rms} + 26 \text{ dB}$ TX6 TX2 TX1 TX0 0 0 0 0			2 3	% %	6
	$V_{MIC} = 2 \text{ mV}_{rms}$ $V_{MIC} = 2 \text{ mV}_{rms} + 26 \text{ dB}$ $I_L = 60 \text{ mA}$ , $AT0 = AT1 = 1$ TX6 TX2 TX1 TX0 1 1 1 1			2 3.5	% %	6
	$V_{MIC} = 22 \text{ mV}_{rms}$ $V_{MIC} = 22 \text{ mV}_{rms} + 24 \text{ dB}$ TX6 TX2 TX1 TX0 0 0 0 0			3 3	% %	6
<b>Squelch function <math>CAL = 470 \text{ nF}</math>, <math>RAL = 68 \text{ k}\Omega</math></b>						
Dynamic range attenuation	$I_L = 28 \text{ mA}$ and $60 \text{ mA}$ (note 2)	8.3	9.3	10.3	dB	6
Squelch inhibition (tested on GT)	$\Delta G_T = G_{T1}(ISQ = 1) - G_{T2}(ISQ = 0)$ $G_{T1} (V_{MIC} = 160 \mu V_{rms})$ , $G_{T2} (V_{MIC} = 3 \text{ mV}_{rms})$ at maximum gain *)	-0.3	0	0.3	dB	6

\*) Max. gain is without +6 dB function which is especially devoted for DTMF

## Notes



- Note 1: transmit gain:  $G_T = V_L/V_{MIC}$  on MIC1/MIC2  
 $G_T = V_L/V_{MIC}$  on MICX with the above values of RAG1 and RAG2
- Note 2: Squelch dynamic range:  $\Delta G_T = G_{T0} - G_{T1}$   
 $G_{T0}$  measured at  $V_{MIC} = 1 \text{ mV}_{rms}$   
 $G_{T1}$  measured at  $V_{MIC} = 160 \mu V_{rms}$

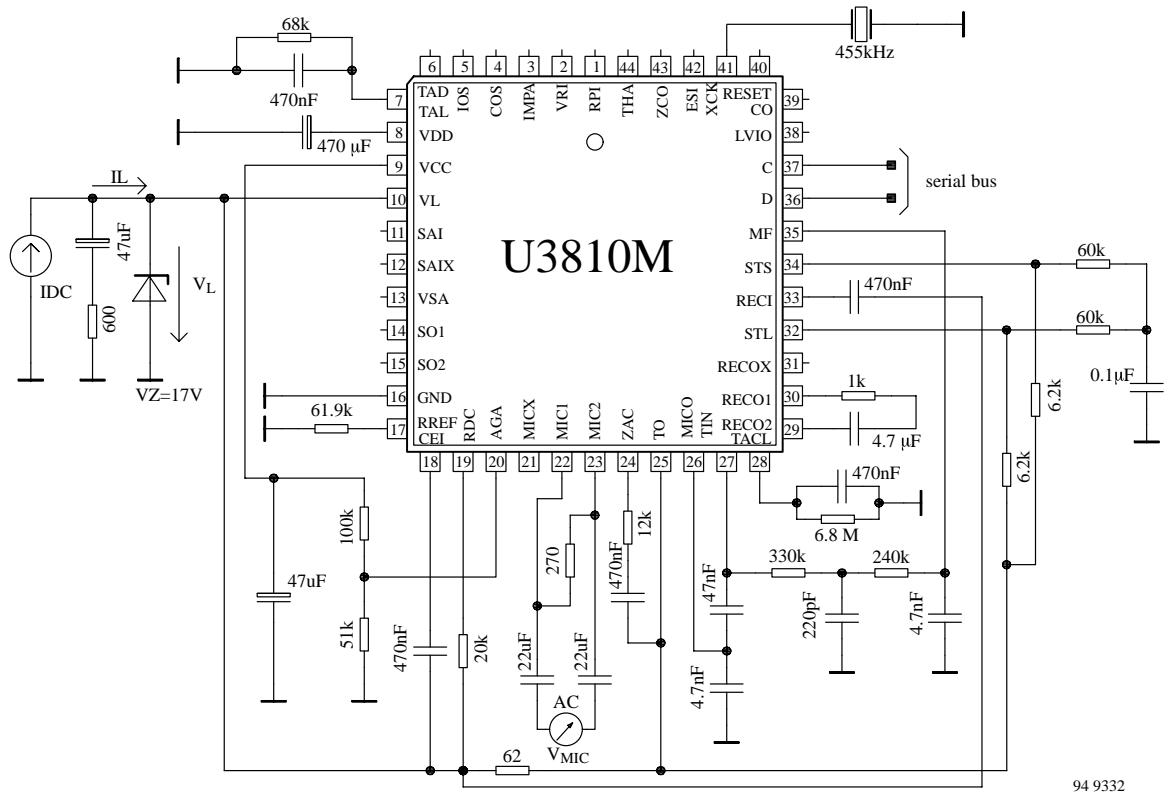


Figure 6. Test circuit for transmission

## Electrical Characteristics – Receive

Parameters	Test Conditions	Min.	Typ.	Max.	Unit	Fig.
$G_R$ receiving gain $G_R = V_R/V_L$ (for normal output)	$V_{GEN} = 0.3 V_{rms}$ RX2 RX1 RX0 maximum gain 1 1 1 minimum gain 0 0 0	3.5 -3.7	4.5 -2.5	5.5 -1.3	dB dB	7
$\Delta G_R = V_R/V_{RECOX}$	$V_{GEN} = 0.3 V_{rms}$	14.0	15.0	16.0	dB	7
Gain adjustment at earphone	$I_L = 28 \text{ mA}$ and 60 mA Attenuation between two steps (both on RECO1/RECO2 and on RECOX)	0.8	1.0	1.3	dB	7
Receiving gain without AGC $G_R = V_R/V_L$ $G_R$ at 28 mA $\Delta G_R$ at 20 mA < $I_L$ < 28 mA $\Delta G_R$ at 28 mA < $I_L$ < 60 mA	$V_{GEN} = 0.3 V_{rms}$ Bit: AGC = 1, IBAL = 1 RX2 RX1 RX0 1 1 1 1 1 1 1 1 1	-3 -0.5 -0.5	-1.5 0 0	0 0.5 0.5	dB dB dB	7
$\Delta G_R$ receiving gain between 28 and 60 mA on RECO1/RECO2 and on RECOX	$V_{GEN} = 0.3 V_{rms}$ AR1 AR0 0 0 0 1 1 0 1 1	3.6 4.9 6.3 7.5	4.1 5.5 6.9 8.3	4.6 5.9 7.4 8.7	dB dB dB dB	7

\*  $V_R = V_{RECO1} - V_{RECO2}$

## Receive (continued)

Parameters	Test Conditions	Min.	Typ.	Max.	Unit	Fig.
+6 dB delta receiving gain on RECO1/RECO2 and on RECOX	$V_{GEN}=0.3 V_{rms}$ , $I_L = 28$ mA and 60 mA (at maximum and minimum gain) Bit RX6 = 1	5.6	6.1	6.7	dB	7
Muted gain	$V_{GEN} = 0.3 V_{rms}$ Mute on RECO1/RECO2 Bit IR=1 Bit IE1 = IE2 = 1 RECOX Bit IR=1	65 60 36			dB dB dB	7
Noise at earpiece psophometric weighted	At maximum gain $V_{GEN} = 0$ V		150	220	$\mu V_p$	7
Receiving distortion	$I_L = 28$ mA and 60 mA max gain, RX6 = 1 $VR = 5 V_{pp}$ min. gain, RX6 = 0 $VR = 2 V_{pp}$		1	3	%	7
Receiver output impedance on RECO1/RECO2 (Pins 29 – 30)	$V_R = 50$ mV <sub>rms</sub>	40	65	85	$\Omega$	7
Receiver output impedance on RECOX (Pin 31)	$V_{RECOX} = 50$ mV <sub>rms</sub>	800	950	1100	$\Omega$	7
Receiver output impedance on RECOX (Pin 31)	$V_{RECOX} = 50$ mV <sub>rms</sub>	800	950	1100	$\Omega$	7
Receiver output offset (Pins 29 – 30)	$I_L = 28$ mA and 60 mA Maximum gain RX6 = 1	–800		+600	mV	7
Automatic sidetone balancing ( $V_R^*/V_{MIC}$ )	At maximum gain $V_{MIC} = 4$ mV <sub>rms</sub> $I_L = 28$ mA $I_L = 60$ mA		24 16		dB dB	6
Digitally balanced sidetone tested on receiving gain $V_R/V_L$	$V_{GEN} = 0.3 V_{rms}$ maximum gain close switch S1 IBAL = 1 BAL2 BAL1 BAL0 0 0 0 0 0 1 0 1 0 0 1 1 1 0 1 1 1 0 1 1 1	18.0 16.9 15.5   4.2	19.0 17.9 16.5 14.5 12.3 9.2 5.7	20.0 18.9 17.5   7.2	dB dB dB dB dB dB dB	7
Confidence level attenuation	$V_{MIC} = 2$ mV <sub>rms</sub> $I_R = 1$	60			dB	6
Confidence level gain $V_R/V_{MIC}$	$V_{MIC} = 2$ mV <sub>rms</sub> $I_R = 1$ EC = 1	19.5	22.0	23.5	dB	6
Z line match. impedance	$V_{GEN} = 0.3 V_{rms}$ $I_L = 28$ and 60 mA	520	570	620	$\Omega$	7

\*  $VR = V_{RECO1} - V_{RECO2}$

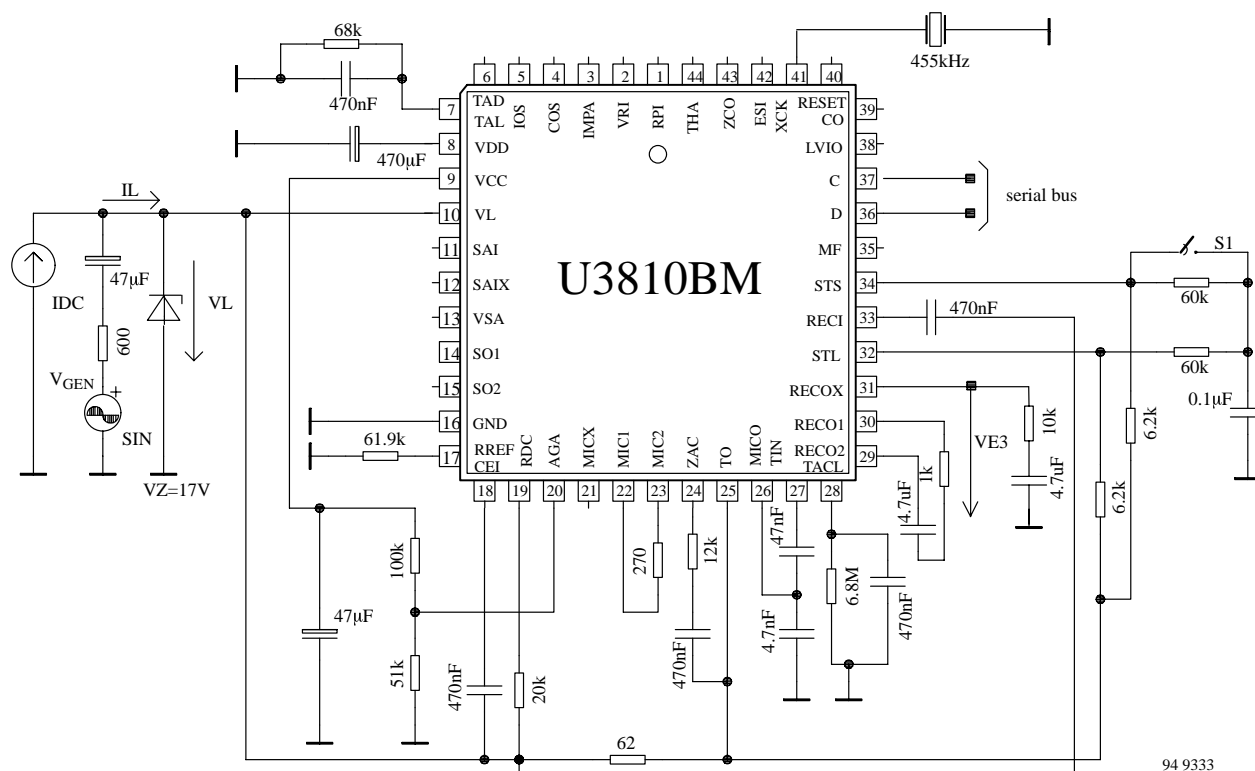


Figure 7. Test circuit for receive

## Electrical Characteristics – Speaker Amplifier

Parameters	Test Conditions	Min.	Typ.	Max.	Unit	Fig.
VSA shunt regulator power supply (Pin 13) in transmission mode	Speaker amplifier without signal EA = 1 IL = 28 mA IL = 60 mA	4.0 7.85	4.3 8.3	4.6 8.7	V V	8
Loudhearing gain (note 3) from SAI/SAIX to SO1, SO2	$\text{GSA1} = \frac{\text{VSO1} - \text{VSO2}}{\text{VSAI}} \quad \text{EA} = 1$ $\text{GSA2} = \frac{\text{VSO1} - \text{VSO2}}{\text{VSAIX}}$ NEA2 NEA1 NEA0 VSAI=VSAIX 1 1 1 VSAI=3.5 mV <sub>rms</sub> 0 0 0 VSAI=88 mV <sub>rms</sub>	33 5	34 6	35 7	dB dB	8
Loudhearing gain between 28 and 60 mA GSA1 and GSA2 (note 3)	EA = 1	−0.5	0	0.5	dB	8
Gain adjustment of speaker amplifier	EA = 1 gain change between two steps	3.8	4.0	4.2	dB	8

## Speaker Amplifier (continued)

Parameters	Test Conditions	Min.	Typ.	Max.	Unit	Fig.
Distortion (measured on 100-Ω load)	EA = 1 NEA2 NEA1 NEA0 VSAI 1 1 1 VSAI=12mV <sub>rms</sub> VSAI=30mV <sub>rms</sub> 0 0 0 VSAI=250mV <sub>rms</sub>  EA = 1, I <sub>L</sub> = 60 mA NEA2 NEA1 NEA0 1 1 1 VSAI=80mV <sub>rms</sub> 0 0 0 VSAI=250mV <sub>rms</sub>			2.0 4.0 1.5  2.0 2.0	% % %  % %	8
Input impedance SAI SAIX	EA = 1	4 4	7 7	10 10	kΩ	8
Confidence gain in loudhearing	V <sub>MIC</sub> = 4 mV <u>V<sub>SO</sub></u> EA = 1 V <sub>MIC</sub> EC = 1, ITAL = 1	36		40	dB	8
Loudhearing input cross talk attenuation	NEA = 7 EA = 1 VSAI = 12 mV <sub>rms</sub> LIS = 1 VSAIX = 12 mV <sub>rms</sub> LIS = 0	60 60			dB dB	8
Output power (note 4, 5)	EA = 1 I <sub>L</sub> = 20 mA 50 Ω load asym. I <sub>L</sub> = 28 mA 100 Ω load sym. I <sub>L</sub> = 60 mA 50 Ω load asym. 100 Ω load sym.	9.0 19.0 140	12.5 11.5 24 30 120 170		mW mW mW mW mW mW	8
Output offset (Pins 14 – 15)	LIS = 0, 1, EA = 1, max. and min. gain	–200		+200	mV	8
Leakage current (Pin 6)	EA = 1			140	nA	8
Offset (Pin 7)	RAL = 68 kΩ, I <sub>L</sub> = 28 mA and 60 mA EA = 1			140	mV	6
<b>Anti-larsen system CAL = 470 nF, RAL = 68 kΩ</b>						
Dynamic range attenuation	IL = 28 mA and 60 mA EA = 1 VSAI = 6 mV <sub>rms</sub> Bit ISQ = 0 BIT ISQ = 1 (note 5)	11 23	12 26	13 29	dB dB	8

- Note 3: GSA1 measured with S2 on 11  
GSA2 measured with S2 on 12 and Bit LIS = 1
- Note 4: Anti-larsen dynamic range:;  
ΔGLS = GLSA – GLSB  
GLSA measured at V<sub>MIC</sub> = 160 μV<sub>rms</sub>  
GLSB measured at V<sub>MIC</sub> = 1 mV<sub>rms</sub>
- Note 5: The available output current of the speaker amplifier can be increased by reduction of the quiescent current of the receiver output stage (bits IE1, IE2, see “contents of internal registers”).





Figure 8. Test circuit for speaker amplifier

## DTMF Dialing

The output pin MF provides the multi-frequency signal to be transmitted on line. This signal is the result of the sum of two frequency pulse modulations and requires an external filter to compose a dual sine wave. The frequencies are chosen in a low group and a high group.

The circuit conforms to the T/CS 46-02 CEPT recommendation concerning DTMF option 1 (-9/-11 dBm) and option 2 (-6/-8 dBm) transmit level 5 (example: in figure 6, option 2 can be fulfilled with 3.5 dB preemphasis and 1.5 Vpp low frequency level Pin 35).

Two different low levels (with 3 dB difference) and two different preemphases (2.5 and 3.5 dB) can be chosen through the serial bus.

## Melody – Confidence Tone

Melody/confidence tone frequencies are given in table 2.

In the state  $SIN1 = 1$ ,  $SIN0 = 0$ , the IC delivers a single pulse-density-modulated frequency at pin MF (the same behavior as DTMF), denoted as a confidence tone. The confidence tone is sent either to the line or on the earpiece. In the state  $SIN1 = 0$ ,  $SIN0 = 1$ , a square wave is sent to the loudhearing input for ringing melodies.

Table 1. Frequency tolerance of the output tones for DTMF signalling tone output frequency when using 455 kHz

Standard Frequency Hz	Tone Output Frequency Hz	Frequency Deviation	
		%	Hz
Low Group			
697	697.85	0.12	+0.85
770	771.18	0.15	+1.18
852	852.06	0.01	+0.06
941	940.08	-0.10	-0.92
High Group			
1209	1210.1	0.09	+1.1
1336	1338.2	0.17	+2.2
1477	1477.3	0.02	+0.3
1633	1636.7	0.22	+3.7

Note: Frequency can be directly measured on CO when S3 is closed (figure 9)

Table 2. Melody generator mode

AMF	SIN0	SIN1	CO
1	1	1	DTMF : HF
0	1	1	DTMF : LF
0	1	0	MELODY
0	0	1	CONFIDENCE TONE

Table 3. DTMF two tone matrix

Upper frequency/ Hz	1209	1336	1477	1633
Lower frequency/ Hz	Corresponding telephone buttons			
697	1	2	3	A
770	4	5	6	B
852	7	8	9	C
941	*	0	#	D

Table 4. Frequency tolerance of the output tone – tone output frequency when using 455 kHz

Standard Frequency Hz	Tone Output Frequency Hz	Frequency Deviation ‰
440.00	440.04	0.09
466.16	466.19	0.06
493.88	493.49	-0.78
523.25	522.99	-0.50
554.36	554.88	0.92
587.33	587.86	0.89
622.25	621.58	-1.07
659.25	659.42	0.26
698.46	697.85	-0.87
740.00	741.04	1.41
784.00	784.48	0.62
830.00	830.29	-0.37
880.00	878.38	-1.84
932.30	932.38	0.08
987.77	989.13	1.38
1046.50	1048.39	1.80
1108.73	1109.76	0.93
1174.66	1172.68	-1.69
1244.50	1243.17	-1.07
1318.50	1315.03	-2.63
1396.90	1395.71	-0.86
1480.00	1477.27	-1.84
1568.00	1568.97	0.62
1661.20	1660.58	-0.37
1760.00	1763.57	2.03
1864.65	1864.75	0.06
1975.50	1978.26	1.40
2093.00	2087.16	-2.79
2217.46	2208.74	-3.93
2349.30	2345.36	-1.68

## Electrical Characteristics – DTMF Generation (Pin MF)

Parameters	Test Conditions	Min.	Typ.	Max.	Unit	Fig.
Tone frequency accuracy	See table 1					
Low-group tone level without attenuation	Note 7 BFOA (preemphasis A)	1.35		1.65	V	9
	S3 closed BFOB (preemphasis B)	1.25	1.50 1.40	1.55	V	
High-group tone level without attenuation	Note 7 HFOA (preemphasis A) S3	1.80	2.00	2.20	V	9
	closed HFOB (preemphasis B)	1.90	2.10	2.35	V	
Preemphasis A without attenuation	Note 7 PROA S3 closed	2.04	2.54	3.04	dB	9
Preemphasis B without attenuation	Note 7 PROB S3 closed	3.02	3.52	4.02	dB	9
Low-group tone level with attenuation	Note 7 BF1A (preemphasis A)	0.95	1.05	1.15	V	9
	S3 closed BF1B (preemphasis B)	0.90	1.00	1.10	V	
High-group tone level with attenuation	Note 7 HF1A (preemphasis A)	1.25	1.40	1.55	V	9
	S3 closed HF1B (preemphasis B)	1.35	1.50	1.65	V	

## DTMF Generation (Pin MF) (continued)

Parameters	Test Conditions	Min.	Typ.	Max.	Unit	Fig.
Preemphasis A with attenuation	Note 7 PR1A S3 closed	2.04	2.54	3.04	dB	9
Preemphasis B with attenuation	Note 7 PR1B S3 closed	3.02	3.52	4.02	dB	9
Leakage	S4 = 2 Bits SIN1 SIN0 S3 closed 0 0	−100		100	nA	9
Distortion at line	IL = 28 mA M = 8 Key = «3» TX6 = 1 IM = 1		1	3	%	6
Low-group tone level at line	IL = 28 mA and 60 mA M = 8 Key = «3» TX6 = 1 IM = 1	−10	−8	−6	dBm	6
<b>Melody generation</b>						
Tone frequency accuracy	See table 4					
Confidence tone level	Note: 7 CTL	2.10	2.33	2.60	V	9

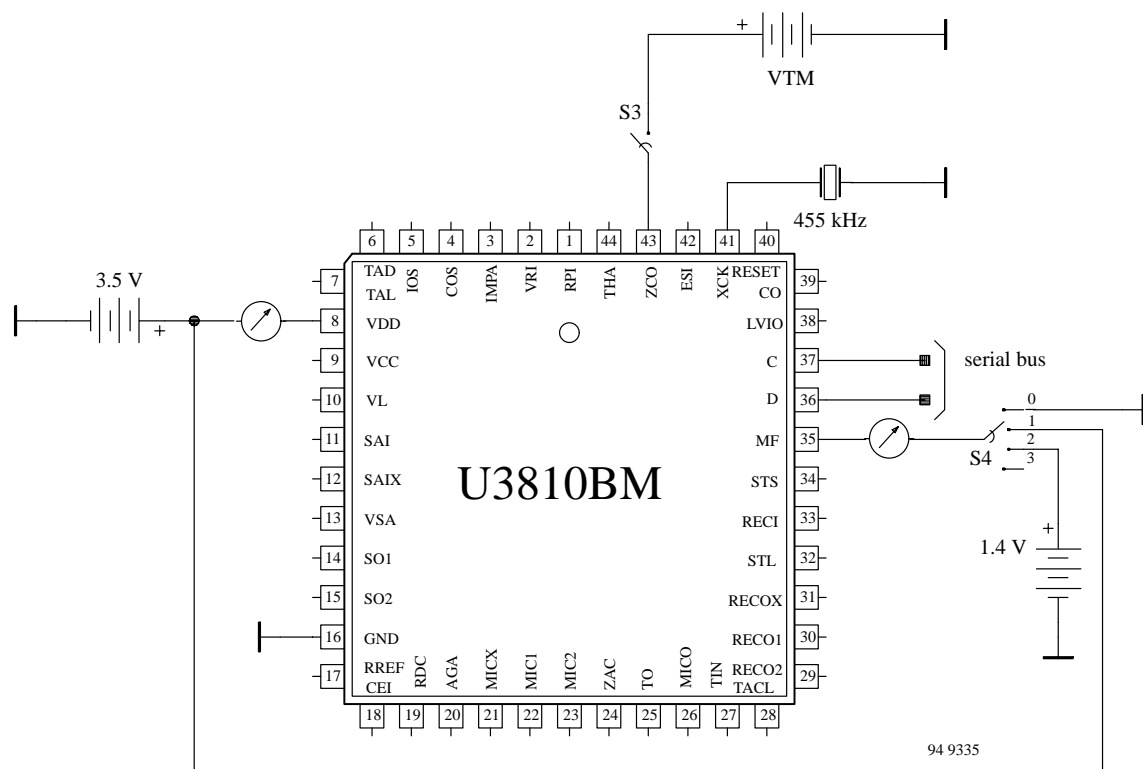


Figure 9. Test circuit for DTMF generation

Table 5. DTMF setup for indirect (by current) level measurement

	Internal Registers Set by Serial Bus				Internal Signal Set by Clock Count			
S4 (Figure 9)	AMF	SIN1	SIN0	M4	FB	FH	F227	I measured
0	0	1	1	0	1	0	X	I1
0	0	1	1	0	0	1	X	I2
0	0	1	1	1	0	1	X	I3
0	0	1	0	0	0	X	1	I4
0	1	1	1	X	0	0	1	I5
1	0	1	1	1	0	1	X	I6
1	0	1	1	0	1	0	X	I7
1	0	1	1	1	1	0	X	I8
1	0	1	0	X	1	X	0	I9
1	1	1	1	X	1	1	0	I10

X: either 1 or 0

## Note 7: DTMF calculations

BF level without attenuation with pre-emphasis A:

$$\text{BFOA} = \left( \frac{I1}{I1 + I7} + \frac{I6}{I6 + I2} \right) \frac{VDD}{2}$$

BF level without attenuation with pre-emphasis B:

$$\text{BFOB} = \left( \frac{I1}{I1 + I8} + \frac{I6}{I6 + I3} \right) \frac{VDD}{2}$$

HF level without attenuation with pre-emphasis A:

$$\text{HFOA} = \left( \frac{I2}{I2 + I6} + \frac{I7}{I7 + I1} \right) \frac{VDD}{2}$$

HF level without attenuation with pre-emphasis B:

$$\text{HFOB} = \left( \frac{I3}{I3 + I6} + \frac{I8}{I8 + I1} \right) \frac{VDD}{2}$$

Pre-emphasis A without attenuation:

$$\text{PROA} = 20 \log \left( \frac{\text{HFOA}}{\text{BFOA}} \right)$$

Pre-emphasis B without attenuation:

$$\text{PROB} = 20 \log \left( \frac{\text{HFOB}}{\text{BFOB}} \right)$$

BF level with attenuation with pre-emphasis A:

$$\text{BF1A} = \left( \frac{I1}{I1 + I7 + I10} + \frac{I6}{I6 + I2 + I5} \right) \frac{VDD}{2}$$

BF level with attenuation with pre-emphasis B:

$$\text{BF1B} = \left( \frac{I1}{I1 + I8 + I10} + \frac{I6}{I6 + I3 + I5} \right) \frac{VDD}{2}$$

HF level with attenuation with pre-emphasis A:

$$\text{HF1A} = \left( \frac{I2}{I2 + I6 + I10} + \frac{I7}{I7 + I1 + I5} \right) \frac{VDD}{2}$$

HF level with attenuation with pre-emphasis B:

$$\text{HF1B} = \left( \frac{I3}{I3 + I6 + I10} + \frac{I8}{I8 + I1 + I5} \right) \frac{VDD}{2}$$

Pre-emphasis A without attenuation:

$$\text{PR1A} = 20 \log \left( \frac{\text{HF1A}}{\text{BF1A}} \right)$$

Pre-emphasis B without attenuation:

$$\text{PR1B} = 20 \log \left( \frac{\text{HF1B}}{\text{BF1B}} \right)$$

Confidence tone level:

$$\text{CTL} = \left( \frac{I1}{I1 + I9} + \frac{I6}{I6 + I4} \right) \frac{VDD}{2}$$

## Electrical Characteristics – Ringer

Parameters	Test Conditions	Min.	Typ.	Max.	Unit	Fig.
THA threshold voltage THTV	$V_S = 5\text{ V}$ S5 on 1	8.30	8.75	9.20	V	10
THA hysteresis $\Delta TH$	$V_S = 5\text{ V}$ S5 on 1	435	465	495	mV	10
VSA threshold voltage VSAON (ring detector enabled)	$V_{THA} = 12\text{ V}$ S5 on 1	3.0	3.2	3.4	V	10
VSA threshold voltage VSAOFF (ring detector disabled)	$V_{THA} = 12\text{ V}$ S5 on 1	2.45	2.5	2.65	V	10
Switching supply output current	$V_S = 5\text{ V}$ $V_{IN} = 30\text{ V}$ S7 on 1 S5 on 1 $R_{IN} = 300\text{ k}\Omega$	33	37		mA	10
Input impedance $V_{IN}/I_{IN}$	S5 on 1 $V_S = 5\text{ V}$ S7 on 1 Off state $V_{IN} = 5\text{ V}$ $R_{IN} = 300\text{ k}\Omega$ On state $V_{IN} = 30\text{ V}$ $R_{IN} = 300\text{ k}\Omega$ On state $V_{IN} = 30\text{ V}$ $R_{IN} = 1500\text{ k}\Omega$	50.0 2.78 13.4	2.90 14.1	3.05 15.0	k $\Omega$ k $\Omega$ k $\Omega$	10
RPI ringing power information	$R_{IN} = 300\text{ k}\Omega$ S7 on 2 S5 on 2 $V_{IN} = 30\text{ V}$ $V_{ESI} = 0\text{ V}$ S5 on 1 $V_{IN} = 0\text{ V}$ $V_{ESI} = 5\text{ V}$	1.48 1.57	1.55 1.61	1.64 1.64	V V	10
VSA/VDD switch off VSA- OFF1 (measured on VSA)	S8 closed S5 on 1 $I_{VDD} = -1\text{ mA}$	5.55	5.8	6.0	V	10
VSA shunt regulator VSAL VSAH	S5 on 3 $I_{SA} = 2\text{ mA}$ $I_{SA} = 45\text{ mA}$	4.75 5.0	5.0 5.3	5.15 5.7	V V	10
Difference between max. VSA voltage and cut off voltage	$VSADIFF = VSAOFF1 - VSAH$	250	500		mV	10
ZCO Zero crossing information	$V_S = 5\text{ V}$ S5 on 1 S6 closed $I_{ZCO} = 100\text{ }\mu\text{A}$ $V_{THA} = 7.5\text{ V}$ $I_{ZCO} = -100\text{ }\mu\text{A}$ $V_{THA} = 12.0\text{ V}$	4.4		0.5	V V	10
Ringer output power (on 100- $\Omega$ load)	$V_{IN} = 30\text{ V}$ $R_{IN} = 300\text{ k}\Omega$ S7 on 2 S5 on 2 SIN1 SIN0 LIS EA 0 1 1 1 NEA – maximum gain	70	105	130	mW	10
Extra ringing attenuation	$V_{IN} = 30\text{ V}$ $R_{IN} = 300\text{ k}\Omega$ S7 on 2 S5 on 2 SIN1 SIN0 LIS AMF EA 0 1 1 1 1 NEA – maximum gain	-12.8	-12.2	-11.6	dB	10



Figure 10. Test circuit for ringer



Figure 11.

## Electrical Characteristics of Logical Part

$f_{\text{xck}} = 455 \text{ kHz}$

$V_{\text{DD}} = 3.5 \text{ V}$

Parameter	Min.	Typ.	Max.	Unit
INPUTS : C, D Low-voltage input $V_{\text{il}}$ High-voltage input $V_{\text{ih}}$	2.8		0.7	V V
Input leakage current $I_{\text{i}}$ ( $0 < V_{\text{I}} < V_{\text{DD}}$ )	-1		1	$\mu\text{A}$
Output: RESET, CO, LVIO Low-voltage output ( $I_{\text{ol}} = 100 \mu\text{A}$ ) $V_{\text{ol}}$ High-voltage output ( $I_{\text{oh}} = -100 \mu\text{A}$ ) $V_{\text{oh}}$	3.1		0.35	V V
CLOCK: CO (figure 16) Using reference ceramic resonator period: $t_{\text{cyc}}$ High pulse width: $t_{\text{wch}}$	2.19 1.10	2.20	2.21 1.45	$\mu\text{s}$ $\mu\text{s}$
SERIAL BUS (figure 20) Data setup time $t_{\text{sud}}$ Data hold time $t_{\text{hd}}$ Clock low time $t_{\text{cl}}$ Clock high time $t_{\text{ch}}$ Hold time before transfer condition $t_{\text{eon}}$ Data low pulse on transfer condition $t_{\text{eh}}$ Data high pulse on transfer condition $t_{\text{eoff}}$	0.1 0 2 2 0.1 0.2 0.2			$\mu\text{s}$ $\mu\text{s}$ $\mu\text{s}$ $\mu\text{s}$ $\mu\text{s}$ $\mu\text{s}$ $\mu\text{s}$
RESET TIMING (figures 12, 13 and 14) Clock start-up time $t_{\text{on}}$ Clock inhibition time $t_{\text{off}}$ Reset time (without $t_{\text{on}}$ ) $t_{\text{r}}$		3  31.6	5 150 32	ms $\mu\text{s}$ ms



## Power-on Reset and Reset Pin

The system (U3810BM + microcontroller) is woken up by an initial condition:

- line voltage (VL)
- ringer (THA)
- external supply (ESI)

To avoid undefined states of the system when it is powered on, an internal reset clears the internal registers, and maintains pin RESET low during trt.

### 1.) Power-on in speed-up condition ( $V_{DD} < V_{SON}$ )

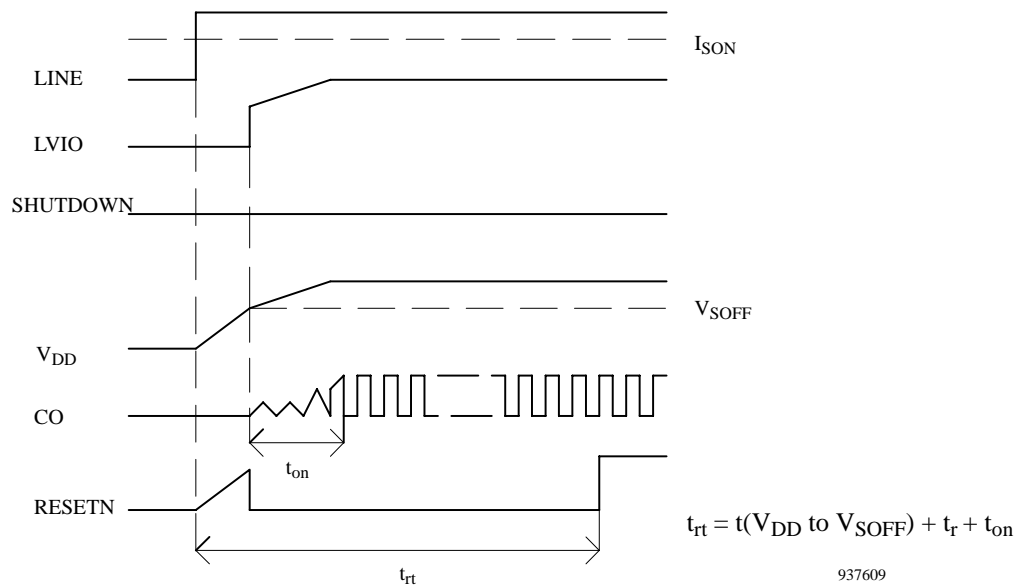


Figure 12. Power on in speed-up condition ( $V_{DD} < V_{SON}$ )

### 2.) Power-on without speed-up ( $V_{DD} > V_{SON}$ )

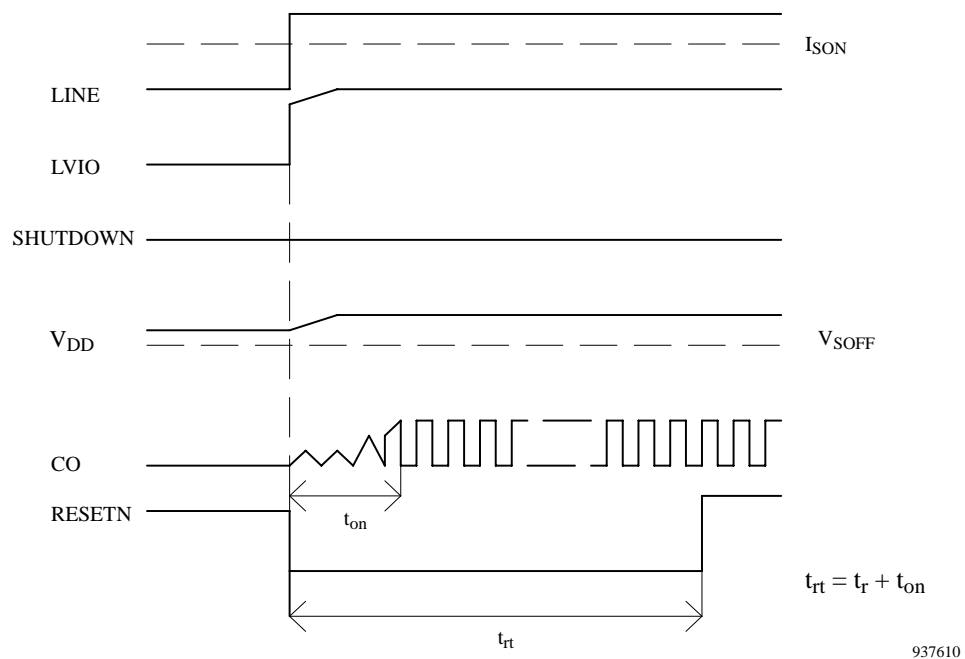


Figure 13. Power on without speed-up ( $V_{DD} > V_{SON}$ )

## 3.) Line break

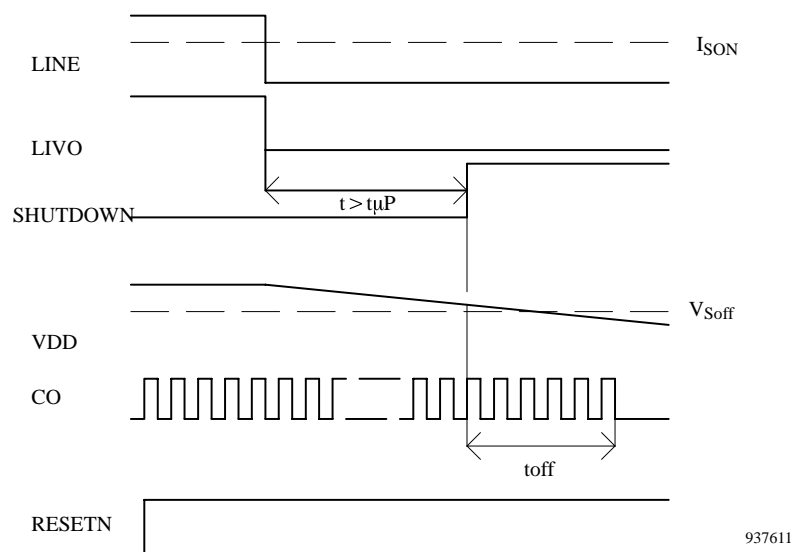


Figure 14. Line break

When the microprocessor detects LIVO low during  $t > t_{\mu C}$  (internal microprocessor timing special for line breaks), it forces high the shutdown bit through the serial bus, thus leading the IC, after  $t_{off}$ , to go into standby mode (oscillator stop). Pin RESET remains high.

When the line break is shorter than  $t_{\mu C}$ , nothing appears.

## 4.) Line current fall

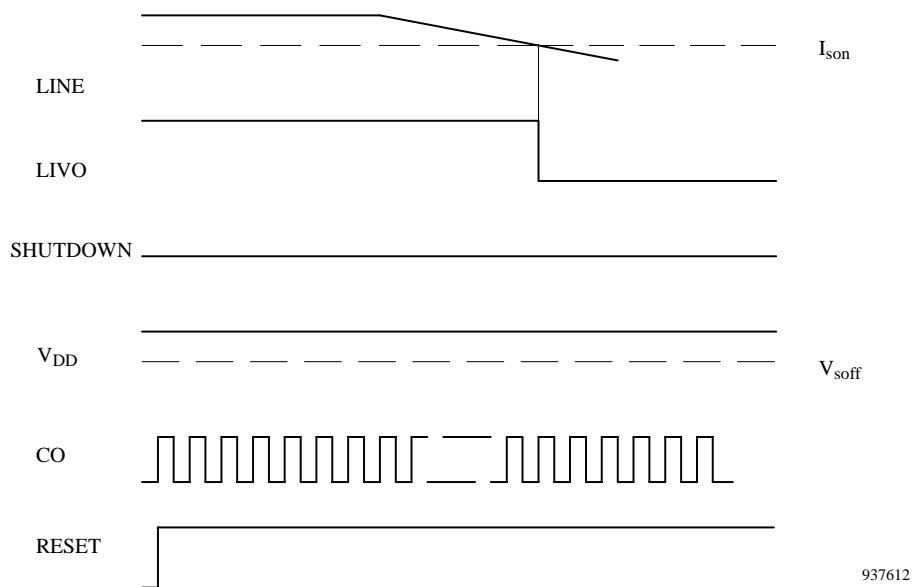


Figure 15. Line current fall

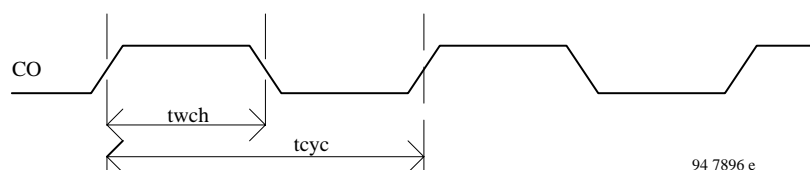


Figure 16. Clock output (CO)

## Serial Bus

The circuit is remoted by an external microcontroller through the serial bus:

The data is an 8-bit word:

B7 – B6 – B5: address of the destination register (0 to 7)

B4 – B0: contents of register

The data line must be stable when the clock is high and data must be serially shifted.

After 8 clock periods, the transfer to the destination register is (internally) generated by a low-to-high transition of the data line when the clock is high.

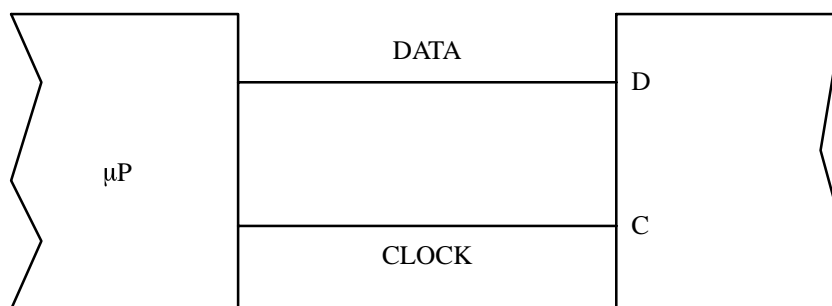


Figure 17. Principle of the serial bus

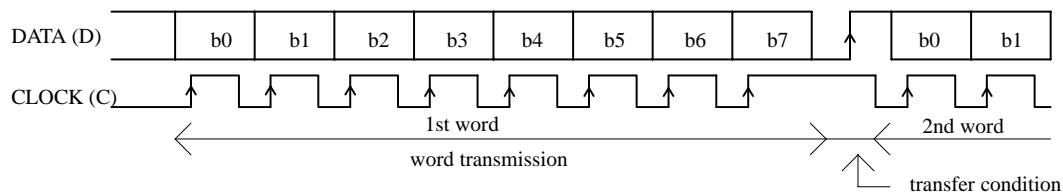


Figure 18.

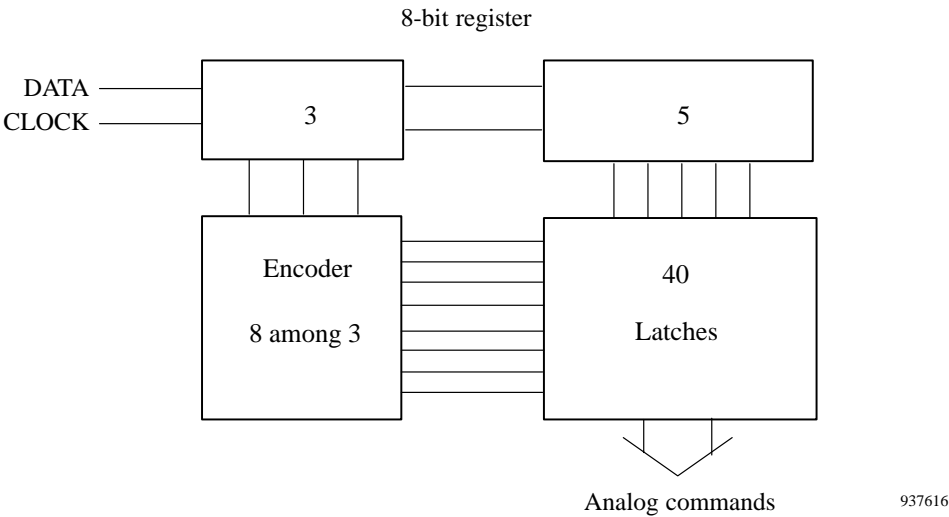


Figure 19. Serial bus interface

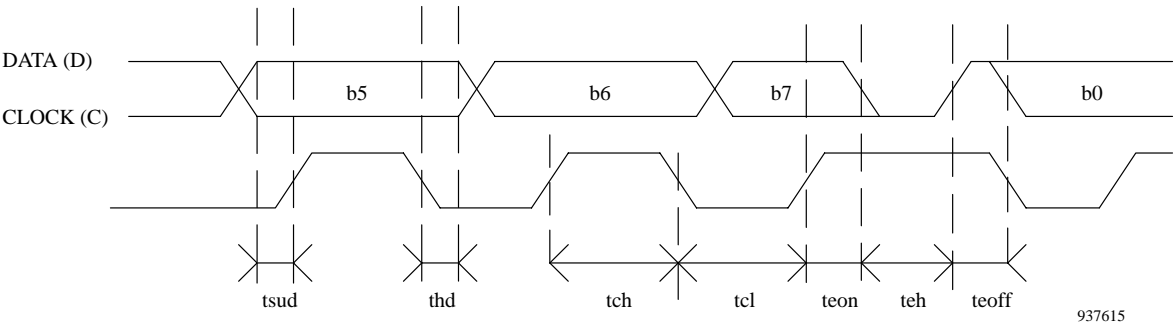


Figure 20. Serial bus timing

Content of Internal Registers

0: Transmission mode

MIS	TX6	TX2	TX1	TX0	MIS: Microphone input switching TX6: +6dB TX: Transmission gain adjustment
-----	-----	-----	-----	-----	--

1: Reception mode

LIS	RX6	RX2	RX1	RX0	LIS: Loudhearing input switching RX6: +6dB RX: Reception gain adjustment
-----	-----	-----	-----	-----	--

2: Loudhearing mode

IAL	EA	NEA2	NEA1	NEA0	IAL: Anti-larsen inhibition EA: Loudhearing enable NEA: Loudhearing gain adjustment
-----	----	------	------	------	---

3: AGC mode

IAGC	AR1	AR0	AT1	AT0	IAGC: AGC inhibition AT: Transmission AGC adjustment AR: Reception AGC adjustment
------	-----	-----	-----	-----	---

4: Sidetone mode

IE2	IBAL	BAL2	BAL1	BAL0	IE2: Reception output amplifier current adjustment IBAL: Inhibition of automatic sidetone balance
-----	------	------	------	------	--

5: Sidetone balancing

BAL	Z
0	1 STS
1	5/6 STS + 1/6 STL
2	2/3 STS + 1/3 STL
3	1/2 STS + 1/2 STL
4	1/2 STS + 1/2 STL
5	1/3 STS + 2/3 STL
6	1/6 STS + 5/6 STL
7	1 STL

6: Internal inhibitions

IR	EC	IT	IM	IE1	IR: Reception inhibition EC: Confidence enable IT: Transmit inhibition IM: Microphone inhibition IE1: Reception output amplifier current adjustment
----	----	----	----	-----	---

7: Earpiece current

IE1	IE2	IREC	
0	0	3 mA	
0	1	2 mA	
1	0	1 mA	
1	1	0 mA	Earpiece inhibition

8: Melody / DTMF choice

M4	M3	M2	M1	M0	
----	----	----	----	----	--

## 9: Melody / DTMF selection

M	Melody or Confidence Tone Output		DTMF mode	
			Key	HF/LF
00	A3	440.0	«1»	2.5 dB
01	A#3	466.2	«4»	”
02	B3	493.5	«7»	”
03	C4	523.0	«*»	”
04	C#4	554.9	«2»	”
05	D4	587.8	«5»	”
06	D#4	621.6	«8»	”
07	E4	659.4	«0»	”
08	F4	697.8	«3»	”
09	F#4	741.0	«6»	”
0A	G4	784.5	«9»	”
0B	G#4	830.3	«#»	”
0C	A4	878.4	«A»	”
0D	A#4	932.4	«B»	”
0E	B4	989.1	«C»	”
0F	C5	1048.4	«D»	”
10	C#	1109.7	«1»	3.5 dB
11	D5	1172.7	«4»	”
12	D#5	1243.2	«7»	”
13	E5	1315.0	«*»	”
14	F5	1395.7	«2»	”
15	F#5	1477.3	«5»	”
16	G5	1569.0	«8»	”
17	G#5	1660.6	«0»	”
18	A5	1763.6	«3»	”
19	A#5	1864.7	«6»	”
1A	B5	1978.3	«9»	”
1B	C6	2087.2	«#»	”
1C	C#6	2208.7	«A»	”
1D	D6	2345.4	«B»	”
1E			«C»	”
1F			«D»	”

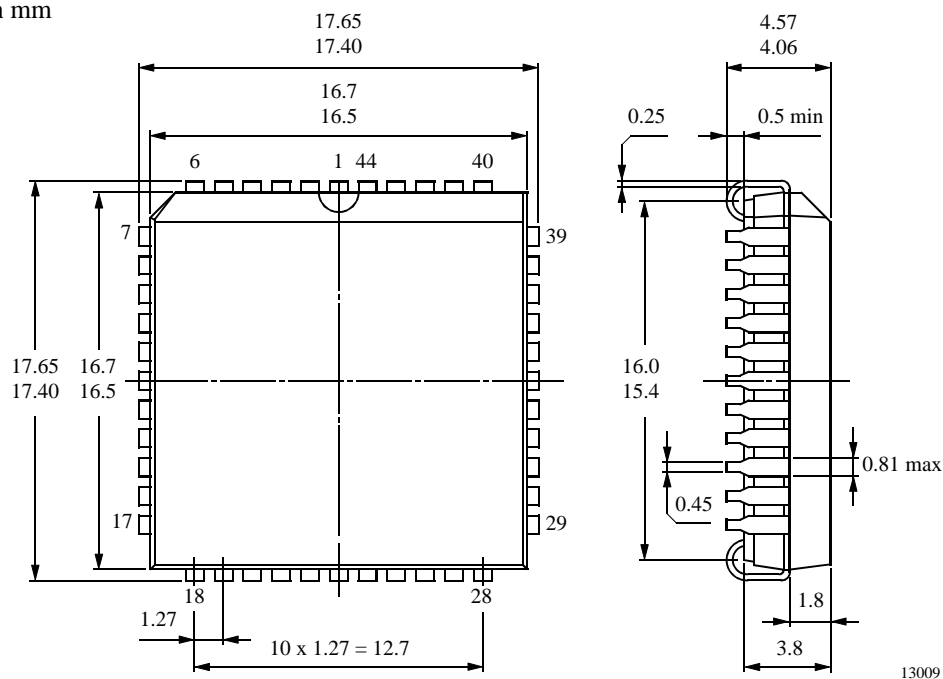
## 10: Control register

AMF	ISQ	SD	SIN1	SIN0	AMF: MF output attenuation –3 dB extra ringing attenuation (12 dB) ISQ: Squelch inhibition SD: Shutdown SIN: Generator mode 0: OFF 1: Melody (Ringer) 2: Confidence tone 3: DTMF
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## Package Information

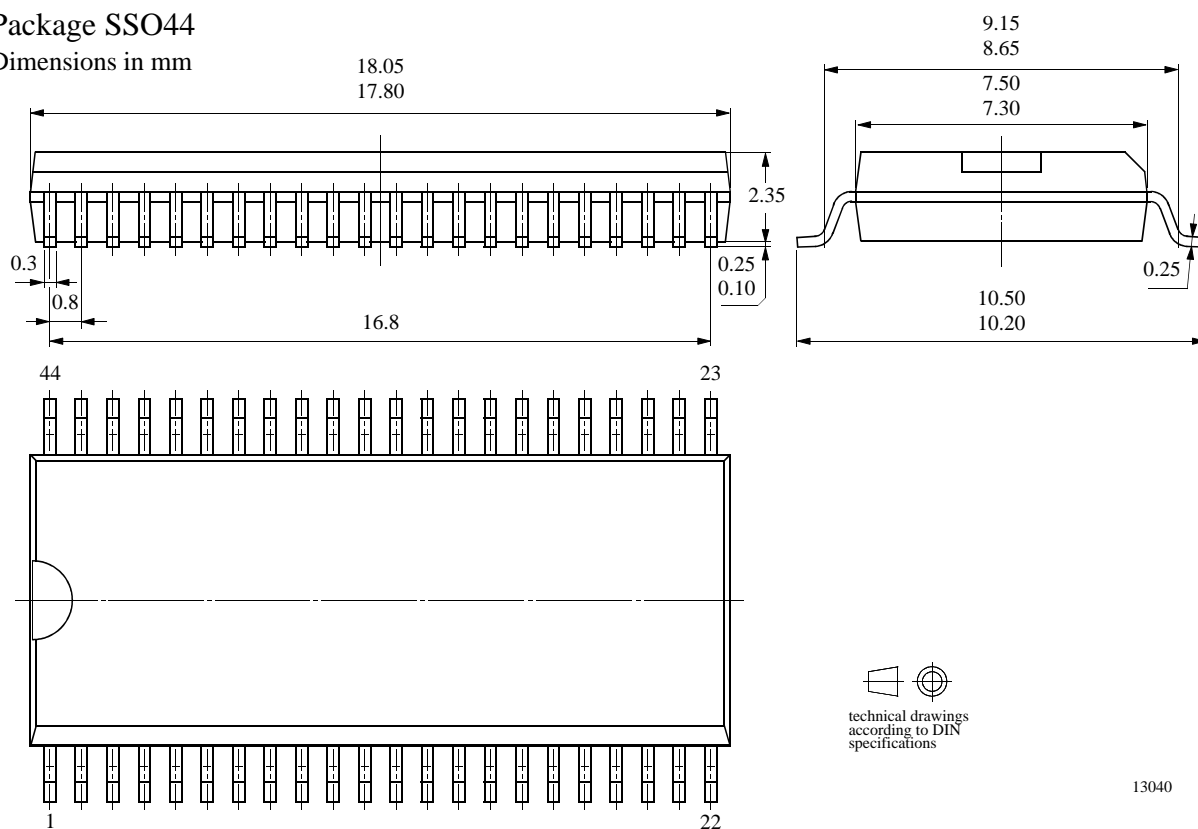
### Package PLCC44

Dimensions in mm



### Package SSO44

Dimensions in mm



## Ozone Depleting Substances Policy Statement

It is the policy of **TEMIC Semiconductor GmbH** to

1. Meet all present and future national and international statutory requirements.
2. Regularly and continuously improve the performance of our products, processes, distribution and operating systems with respect to their impact on the health and safety of our employees and the public, as well as their impact on the environment.

It is particular concern to control or eliminate releases of those substances into the atmosphere which are known as ozone depleting substances (ODSs).

The Montreal Protocol (1987) and its London Amendments (1990) intend to severely restrict the use of ODSs and forbid their use within the next ten years. Various national and international initiatives are pressing for an earlier ban on these substances.

**TEMIC Semiconductor GmbH** has been able to use its policy of continuous improvements to eliminate the use of ODSs listed in the following documents.

1. Annex A, B and list of transitional substances of the Montreal Protocol and the London Amendments respectively
2. Class I and II ozone depleting substances in the Clean Air Act Amendments of 1990 by the Environmental Protection Agency (EPA) in the USA
3. Council Decision 88/540/EEC and 91/690/EEC Annex A, B and C (transitional substances) respectively.

**TEMIC Semiconductor GmbH** can certify that our semiconductors are not manufactured with ozone depleting substances and do not contain such substances.

**We reserve the right to make changes to improve technical design and may do so without further notice.**

Parameters can vary in different applications. All operating parameters must be validated for each customer application by the customer. Should the buyer use TEMIC products for any unintended or unauthorized application, the buyer shall indemnify TEMIC against all claims, costs, damages, and expenses, arising out of, directly or indirectly, any claim of personal damage, injury or death associated with such unintended or unauthorized use.

TEMIC Semiconductor GmbH, P.O.B. 3535, D-74025 Heilbronn, Germany  
Telephone: 49 (0)7131 67 2831, Fax number: 49 (0)7131 67 2423



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