Features

- Speech Circuit with Anti-clipping
- Tone-ringer Interface with DC/DC Converter
- Speaker Amplifier with Anti-distortion
- Power-supply Management (Regulated and Unregulated) and a Special Supply for Electret Microphone
- Voice Switch
- Interface for Answering Machine and Cordless Phone

Applications

- Feature Phone
- Answering Machine
- Fax Machine
- Speaker Phone
- Cordless Phone

Benefits

- No Piezoelectric Transducer Necessary for Tone Ringing
- Complete System Integration of Analog Signal Processing on One Chip
- Very Few External Components

1. Description

The programmable telephone audio processor U4091BM-R is a linear integrated circuit for use in feature phones, answering machines and fax machines. It contains the speech circuit, tone-ringer interface with DC/DC converter, sidetone equivalent and ear-protection rectifiers. The circuit is line-powered and contains all components necessary for signal amplification and adaptation to the line. The U4091BM-R can also be supplied via an external power supply. An integrated voice switch with loudspeaker amplifier enables hands-free or open-listening operation. With an anti-feedback function, acoustic feedback during open listening can be reduced significantly. The generated supply voltage is suitable for a wide range of peripheral circuits.



Programmable Telephone Audio Processor

U4091BM-R



Rev. 4872A-CORD-08/05



Figure 1-1. Block Diagram

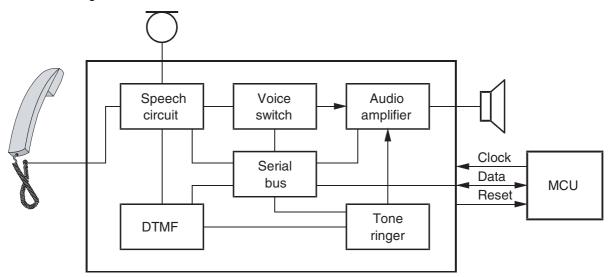
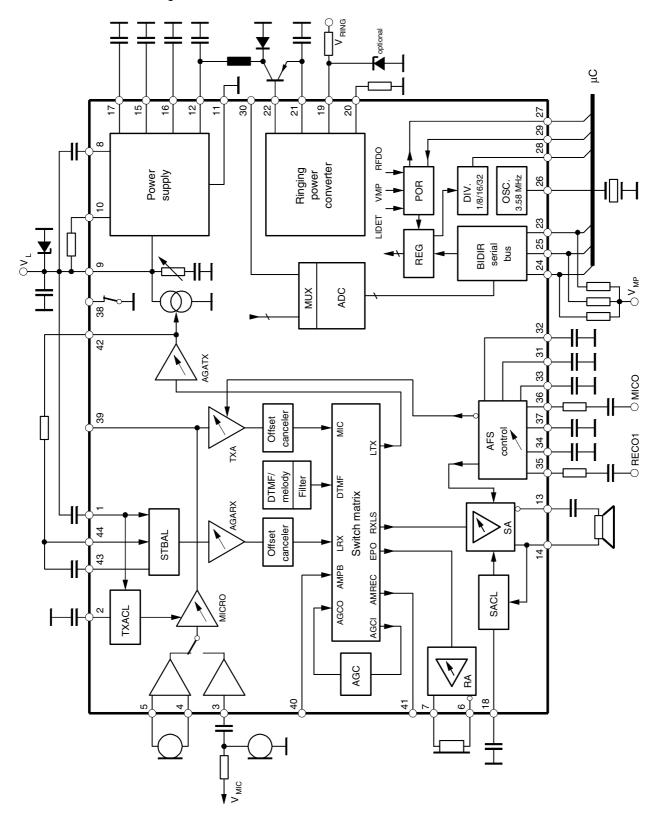


Figure 1-2. Detailed Block Diagram







2. Pin Configuration

Figure 2-1. Pinning SSO44

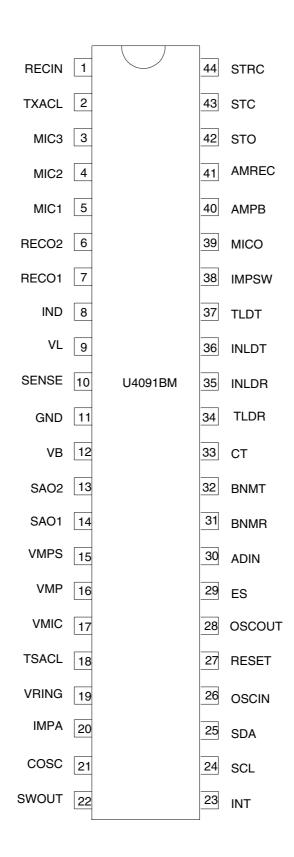


Table 2-1.Pin Description

Pin	Symbol	Function
1	RECIN	Receive amplifier input ⁽¹⁾
2	TXACL	Time constant adjustment for transmit anti-clipping
3	MIC3	Microphone input for hands-free operation
4	MIC2	Input of symmetrical microphone amplifier with high common-mode rejection ratio
5	MIC1	Input of symmetrical microphone amplifier with high common-mode rejection ratio
6	RECO2	Output of the receive amplifier
7	RECO1	Output of the receive amplifier, also used for sidetone network
8	IND	The internal equivalent inductance of the circuit is proportional to the value of the capacitor at this pin. A resistor connected to ground may be used to adjust the DC mask
9	VL	Positive supply-voltage input to the device in speech mode
10	SENSE	Input for sensing the available line current
11	GND	Ground, reference point for DC and AC signals
12	VB	Unstabilized supply voltage for speech network
13	SAO2	Negative output of speaker amplifier (push-pull only)
14	SAO1	Positive output of speaker amplifier (single-ended and push-pull operation)
15	VMPS	Unregulated supply voltage for the microcontroller (via series regulator to VMP)
16	VMP	Regulated output voltage for supplying the microcontroller (typically 3.3V/6 mA in speech mode)
17	VMIC	Reference node for microphone amplifier, supply for electret microphones
18	TSACL	Time constant for speaker amplifier anti-clipping
19	VRING	Input for ringer supply
20	IMPA	Input for adjusting the ringer input impedance
21	COSC	70-kHz oscillator for ringing power converter
22	SWOUT	Output for driving the external switch resistor
23	INT	Interrupt line for serial bus
24	SCL	Clock input for serial bus
25	SDA	Data line for serial bus
26	OSCIN	Input for 3.58-MHz oscillator
27	RESET	Reset output for the microcontroller
28	OSCOUT	Clock output for the microcontroller
29	ES	Input for external supply indication
30	ADIN	Input of A/D converter
31	BNMR	Output of background-noise monitor receive
32	BNMT	Output of background-noise monitor transmit
33	СТ	Time constant for mode switching of voice switch
34	TLDR	Time constant of receive-level detector
35	INLDR	Input of receive-level detector
36	INLDT	Input of transmit-level detector
37	TLDT	Time constant of transmit-level detector
38	IMPSW	Switch for additional line impedance
39	MICO	Microphone preamplifier output
38	IVIICO	Micropriorie preampliner output

Note: 1. The protection device at pin RECIN is disconnected.





Table 2-1. Pin Description

Pin	Symbol	Function
40	AMPB	Input for playback signal of answering machine
41	AMREC	Output for recording signal of answering machine
42	STO	Output for connecting the sidetone network
43	STC	Input for sidetone network
44	STRC	Input for sidetone network

Note: 1. The protection device at pin RECIN is disconnected.

3. DC Line Interface and Supply-voltage Generation

The DC line interface consists of an electronic inductance and a dual-port output stage which charges the capacitors at VMPS and VB. The value of the equivalent inductance is given by:

$$L = \frac{2 \times R_{SENSE} \times C_{IND} \times (R_{DC} \times R_{30})}{(R_{DC} + R_{30})}$$

The U4091BM-R contains two identical series regulators which provide a supply voltage VMP of 3.3V suitable for a microprocessor. In speech mode, both regulators are active because VMPS and VB are charged simultaneously by the DC line interface. The output current is 6 mA. The capacitor at VMPS is used to provide the microcomputer with sufficient power during long line interruptions. Thus, long flash pulses can be bridged or an LCD display can be turned on for more than 2 seconds after going on-hook. When the system is in ringing mode, VB is charged by the on-chip ringing power converter. In this mode, only one regulator is used to supply VMP with maximum 3 mA.

4. Supply Structure of the Chip

A main benefit of the U4091BM is the easy implementation of various applications due to the flexible system structure of the chip.

Possible applications:

- Group listening phone
- Hands-free phone
- Phones which feature ringing with the built-in speaker amplifier
- Answering machine with external supply

The special supply topology for the various functional blocks is shown in Figure 4-1 on page 7.

There are four major supply states:

1. Speech condition

In speech condition, the system is supplied by the line current. If the LIDET block detects a line voltage above approximately 2V, the internal signal VLON is activated. This is detected via the serial bus, all the blocks which are needed have to be switched on via the serial bus.

For line voltages below 2V, the switches remain in quiescent state as shown in the diagram.

2. Power down (pulse dialing)

When the chip is in power-down mode (bit LOMAKE), for example, during pulse dialing, all internal blocks are disabled via the serial bus. In this condition, the voltage regulators and their internal band gap are the only active blocks.

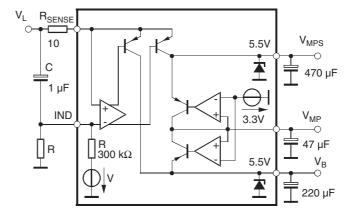
3. Ringing

During ringing, the supply for the system is fed into VB via the Ringing Power Converter (RPC). Normally, the speaker amplifier in single-ended mode is used for ringing. The frequency for the melody is generated by the DTMF/Melody generator.

4. External supply

In an answering machine, the chip is powered by an external supply via pin VB. The answering machine connections can be directly made to U4091BM-R. The answering machine is connected to the pin AMREC. For the output AMREC, an AGC function is selectable via the serial bus. The output of the answering machine will be connected to the pin AMPB, which is directly connected to the switching matrix. This enables the signal to be switched to every desired output.

Figure 4-1. Supply Generator



5. Ringing Power Converter (RPC)

The RPC transforms the input power at VRING (high voltage/low current) into an equivalent output power at VB (low voltage/high current) which is capable of driving the low-ohmic loudspeaker. The input impedance at VRING is adjustable from 3 k Ω to 12 k Ω by R_{IMPA} ($Z_{RING} = R_{IMPA} / 100$) and the efficiency of the step-down converter is approximately 65%.

6. Ringing Frequency Detector (RFD)

The U4091BM-R provides an output signal for the microcontroller. This output signal is always double the value of the input signal (ringing frequency). It is generated by a current comparator with hysteresis. The levels for the on-threshold are programmable in 16 steps, the off-level is fixed. Every change of the comparator output generates a high level at the interrupt output INT. The information can then be read out by means of a serial bus with either normal or fast read mode. The block RFD is always enabled.





Table 6-1. Threshold Level

RINGTH[0:3]	V _{RING}
0	7V
15	22V
Step	1V

7. Clock Output Divider Adjustment

The pin OSCOUT is a clock output which is derived from the crystal oscillator. It can be used to drive a microcontroller or another remote component and thereby reduces the number of crystals required. The oscillator frequency can be divided by 1, 8, 16, or 32. During power-on reset, the divider will be reset to 1 until it is changed by setting the serial bus.

Table 7-1. Clock Output

CLK[0:1]	Divider	Frequency
0	1	3.58 MHz
1	8	447 kHz
2	16	224 kHz
3	32	112 kHz

8. Serial Bus Interface

The circuit is controlled by an external microcontroller through the serial bus.

The serial bus is a bi-directional system consisting of a single-directional clock line (SCL) which is always driven by the microcontroller, and a bi-directional data-signal line. It is driven by the microcontroller as well as by the U4091BM-R (see Figure 20-1 on page 37).

The serial bus requires external pull-up resistors as only pull-down transistors (pin SDA) are integrated.

8.1 WRITE

The data is a 12-bit word:

A0-A3: address of the destination register (0 to 15)

D0-D7: content of the register

The data line must be stable when the clock is high. Data must be shifted serially. After 12 clock periods, the write indication is sent. Then, the transfer to the destination register is (internally) generated by a strobe signal transition of the data line when the clock is high.

8.2 READ

There is a normal and a fast-read cycle.

In the normal read cycle, the microcontroller sends a 4-bit address followed by the read indicator, then an 8-bit word is read out. The U4091BM-R drives the data line.

The fast read cycle is indicated by a strobe signal. With the following two clocks the U4091BM-R reads out the status bits RFDO and LIDET which indicate that a ringing signal or a line signal is present (see Figure 10-1 on page 11, Figure 10-2 on page 11 and Figure 10-3 on page 11).

DTMF Dialing

The DTMF generator sends a multi-frequency signal through the matrix to the line. The signal is the result of the sum of two frequencies and is internally filtered. The frequencies are chosen from a low and a high frequency group. The circuit conforms to the CEPT recommendation concerning DTMF option. Three different levels for the low level group and two different preemphasis (2.5 dB and 3.5 dB) can be chosen by means of the serial bus (rec. T/CF 46-03).

Attention: In high gain mode, distortion can occur if AGATX is high and DC mask is low.

10. Melody and Confidence Tone Generation

Melody and confidence tone frequencies are given in Table 10-1.

The frequencies are provided at the DTMF input of the switch matrix. A sinusoidal wave, a square wave or a pulsed wave can be selected by the serial bus. A square signal means the output is high for half of the frequency cycle, and low for the other half. A pulsed signal means high impedance phases of 1/6 of the period occur between the high and low phases.

Table 10-1. Status of Melody Generating

Decimal	DTMFM[0:2]	Status
0	000	DTMF generator OFF
1	001	Confidence tone melody on (sine)
2	010	Ringer melody (pulse)
3	011	Ringer melody (square signal)
4	100	DTMF (mid level)
5	101	DTMF (low level)
6	110	DTMF (high level)
7	111	-

Table 10-2. DTMF Frequencies

Decimal	DTMFF[0:1] in DTMF Mode	Frequency	Error (%)
0	00	697	-0.007
1	01	770	-0.156
2	10	852	0.032
3	11	941	0.316

Table 10-3. DTMF Frequencies

Decimal	DTMFF[2:3] in DTMF Mode	Frequency	Error (%)
0	00	1209	-0.110
1	01	1336	0.123
2	10	1477	-0.020
3	11	1633	-0.182





Table 10-4. DTMFF4 in DTMF Mode

Pre-emphasis Selection	Level
0	2.5 dB
1	3.5 dB

 Table 10-5.
 DTMF and Melody Frequencies

Decimal	DTMFF [0:4]	f (Hz)	Tone/ Name	Error (%)	DTMF Freq.	DTMP Freq.	Key
0	00000	440.0	A^4	-0.008	697	1209	1
1	00001	466.2	A# ⁴	-0.016	770	1209	4
2	00010	493.9	B ⁴	-0.003	852	1209	7
3	00011	523.2	C ⁴	0.014	941	1209	*
4	00100	554.4	C# ⁴	0.018	697	1336	2
5	00101	587.3	D ⁴	-0.023	770	1336	5
6	00110	622.3	D# ⁴	-0.129	852	1336	8
7	00111	659.3	E ⁴	0.106	941	1336	0
8	01000	698.5	F ⁴	-0.216	697	1477	3
9	01001	740.0	F# ⁴	-0.222	770	1477	6
10	01010	784.0	G ⁴	0.126	852	1477	9
11	01011	830.0	G# ⁴	-0.169	941	1477	#
12	01100	880.0	A^5	0.288	697	1633	Α
13	01101	932.3	A # ⁵	-0.014	770	1633	В
14	01110	987.8	B ⁵	-0.004	852	1633	С
15	01111	1046.5	C ⁵	-0.335	941	1633	D
16	10000	1108.7	C# ⁵	-0.355	697	1209	1
17	10001	1174.7	D^5	-0.023	770	1209	4
18	10010	1244.5	D# ⁵	-0.129	852	1209	7
19	10011	1318.5	E ⁵	0.106	941	1209	*
20	10100	1396.9	F ⁵	-0.214	697	1336	2
21	10101	1480.0	F# ⁵	-0.222	770	1336	5
22	10110	1568.0	G ⁵	0.126	852	1336	8
23	10111	1661.2	G# ⁵	-0.241	941	1336	0
24	11000	1760.0	A^6	-0.302	697	1477	3
25	11001	1864.6	A# ⁶	-0.014	770	1477	6
26	11010	1975.5	B ⁶	0.665	852	1477	9
27	11011	2093.0	C ₆	0.367	941	1477	#
28	11100	2217.5	C# ⁶	0.387	697	1633	Α
29	11101	2349.3	D ⁶	0.771	770	1633	В
30	11110	2663.3			852	1633	С
31	11111	2983.0			941	1633	D

Figure 10-1. Write Cycle

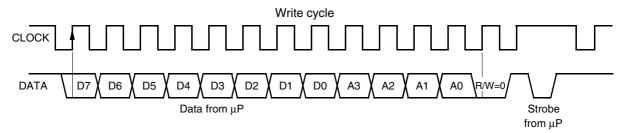


Figure 10-2. Normal Read Cycle

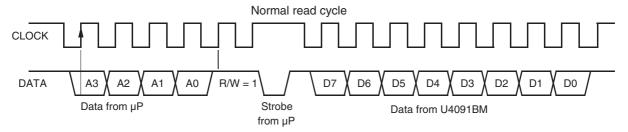


Figure 10-3. Fast Read Cycle

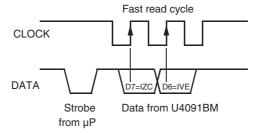




Table 10-6. Names and Functions of the Serial Registers

R0B0	
R0B2 ETX Enable transmit part R0B3 ENVM Enable VM generator R0B4 ENMIC Enable microphone R0B5 ENSTBAL Enable sidetone R0B6 MUTE Muting earpiece amplifier R0B7 ENRLT Enable POR low threshold R1B0 ENSACL Enable anti-clipping for speaker amplifier R1B1 ENSA Enable speaker amplifier and AFS R1B2 ENSAO Enable output stage speaker amplifier R1B3 ENAM Enable answering machine connections R1B4 ENAGC Enable AGC for answering machine R1B5 Reserved -	1
R0B3 ENVM Enable VM generator R0B4 ENMIC Enable microphone R0B5 ENSTBAL Enable sidetone R0B6 MUTE Muting earpiece amplifier R0B7 ENRLT Enable POR low threshold R1B0 ENSACL Enable anti-clipping for speaker amplifier R1B1 ENSA Enable speaker amplifier and AFS R1B2 ENSAO Enable output stage speaker amplifier R1B3 ENAM Enable answering machine connections R1B4 ENAGC Enable AGC for answering machine R1B5 Reserved -	0
R0B4 ENMIC Enable microphone R0B5 ENSTBAL Enable sidetone R0B6 MUTE Muting earpiece amplifier R0B7 ENRLT Enable POR low threshold R1B0 ENSACL Enable anti-clipping for speaker amplifier R1B1 ENSA Enable speaker amplifier and AFS R1B2 ENSAO Enable output stage speaker amplifier R1B3 ENAM Enable answering machine connections R1B4 ENAGC Enable AGC for answering machine R1B5 Reserved -	0
R0B4 ENMIC Enable microphone R0B5 ENSTBAL Enable sidetone R0B6 MUTE Muting earpiece amplifier R0B7 ENRLT Enable POR low threshold R1B0 ENSACL Enable anti-clipping for speaker amplifier R1B1 ENSA Enable speaker amplifier and AFS R1B2 ENSAO Enable output stage speaker amplifier R1B3 ENAM Enable answering machine connections R1B4 ENAGC Enable AGC for answering machine R1B5 Reserved -	1
R0B6 MUTE Muting earpiece amplifier R0B7 ENRLT Enable POR low threshold R1B0 ENSACL Enable anti-clipping for speaker amplifier R1B1 ENSA Enable speaker amplifier and AFS R1B2 ENSAO Enable output stage speaker amplifier R1B3 ENAM Enable answering machine connections R1B4 ENAGC Enable AGC for answering machine R1B5 Reserved -	0
R1 Enables ENRLT Enable POR low threshold ENSACL Enable anti-clipping for speaker amplifier R1B1 ENSA Enable speaker amplifier and AFS R1B2 ENSAO Enable output stage speaker amplifier R1B3 ENAM Enable answering machine connections R1B4 ENAGC Enable AGC for answering machine R1B5 Reserved -	0
R1B0 ENSACL Enable anti-clipping for speaker amplifier R1B1 ENSA Enable speaker amplifier and AFS R1B2 ENSAO Enable output stage speaker amplifier R1B3 ENAM Enable answering machine connections R1B4 ENAGC Enable AGC for answering machine R1B5 Reserved -	0
R1B1 ENSA Enable speaker amplifier and AFS R1B2 ENSAO Enable output stage speaker amplifier R1B3 ENAM Enable answering machine connections R1B4 ENAGC Enable AGC for answering machine R1B5 Reserved -	1
R1 Enables R1B2 ENSAO Enable output stage speaker amplifier R1B3 ENAM Enable answering machine connections R1B4 ENAGC Enable AGC for answering machine R1B5 Reserved -	0
R1 Enables Enables ENAM Enable answering machine connections R1B4 ENAGC Enable AGC for answering machine R1B5 Reserved -	0
R1 Enables R1B4 ENAGC Enable AGC for answering machine R1B5 Reserved -	0
R1B4 ENAGC Enable AGC for answering machine R1B5 Reserved -	0
	0
P1R6 Record	0
nibo neserveu -	0
R1B7 FOFFC Speed up offset canceller	0
R2B0 I1O1 Switch on MIC/LTX	0
R2B1 I1O2 Switch on MIC/SA	0
R2B2 I1O3 Switch on MIC/EPO	0
R2B3 I1O4 Switch on MIC/AMREC	0
R2 Matrix R2B4 I105 Switch on MIC/AGCI	0
R2B5 I2O1 Switch on DTMF/LTX	0
R2B6 I2O2 Switch on DTMF/SA	0
R2B7 I2O3 Switch on DTMF/EPO	0
R3B0 I2O4 Switch on DTMF/AMREC	0
R3B1 I2O5 Switch on DTMF/AGCI	0
R3B2 I3O1 Switch on LRX/LTX	0
R3B3 I3O2 Switch on LRX/SA	0
R3 Matrix R3B4 I3O3 Switch on LRX/EPO	0
R3B5 I3O4 Switch on LRX/AMREC	0
R3B6 I3O5 Switch on LRX/AGCI	0
R3B7 I4O1 Switch on AMPB/LTX	0

 Table 10-6.
 Names and Functions of the Serial Registers (Continued)

Register	Group	No.	Name	Description	Status
		R4B0	1402	Switch on AMPB/SA	0
R4	Matrix	R4B1	1403	Switch on AMPB/EPO	0
		R4B2	1404	Switch on AMPB/AMREC	0
		R4B3	I4O5	Switch on AMPB/AGCI	0
		R4B4	I5O1	Switch on AGCO/LTX	0
		R4B5	1502	Switch on AGCO/SA	0
		R4B6	1503	Switch on AGCO/EPO	0
		R4B7	1504	Switch on AGCO/AMREC	0
		R5B0	EAFS	Enable AFS block	0
R5	R4B2 I4O4 Switch on AMPB/AMREC R4B3 I4O5 Switch on AMPB/AGCI R4B4 I5O1 Switch on AGCO/LTX R4B5 I5O2 Switch on AGCO/SA R4B6 I5O3 Switch on AGCO/EPO R4B7 I5O4 Switch on AGCO/AMREC	0			
	MICLIM	R5B2	AGATX1	Gain transmit AGA	0
	Matrix	AGATX2	Gain transmit AGA MSB	0	
		R5B4	I4O2 Switch on AMPB/SA I4O3 Switch on AMPB/EPO I4O4 Switch on AMPB/AMREC I4O5 Switch on AMPB/AMREC I5O1 Switch on AGCO/LTX I5O2 Switch on AGCO/SA I5O3 Switch on AGCO/EPO I5O4 Switch on AGCO/AMREC EAFS Enable AFS block AGATX0 Gain transmit AGA LSB AGATX1 Gain transmit AGA MSB MICHF Select RF-microphone input DBM5 Maximum transmit level for anti-clipping MICO Gain microphone amplifier LSB MIC1 Gain microphone amplifier MSB SD Shut down Reserved - SL0 Slope adjustment for sidetone LSB SL1 Slope adjustment for sidetone MSB LF0 Low frequency adjustment for sidetone LS LF1 Low frequency adjustment for sidetone LF2 Low frequency adjustment for sidetone LF3 Low frequency adjustment for sidetone LF3 Low frequency adjustment for sidetone PO Pole adjustment for sidetone P1 Pole adjustment for sidetone P2 Pole adjustment for sidetone P3 Pole adjustment for sidetone P4 Pole adjustment for sidetone P5 AGARX0 Gain receive AGC LSB AGARX1 Gain receive AGC MSB	Select RF-microphone input	0
		R4B0 I4O2 Switch on AMPB/SA R4B1 I4O3 Switch on AMPB/EP R4B2 I4O4 Switch on AMPB/EP R4B3 I4O5 Switch on AMPB/AM R4B3 I4O5 Switch on AMPB/AM R4B4 I5O1 Switch on AGCO/LT R4B5 I5O2 Switch on AGCO/SA R4B6 I5O3 Switch on AGCO/SA R4B6 I5O3 Switch on AGCO/AM R5B0 EAFS Enable AFS block R5B1 AGATX0 Gain transmit AGA L R5B2 AGATX1 Gain transmit AGA L R5B3 AGATX2 Gain transmit AGA M R5B4 MICHF Select RF-microphon R5B5 DBM5 Maximum transmit Ie R5B6 MICO Gain microphone am R5B7 MIC1 Gain microphone am R6B0 SD Shut down R6B1 Reserved - R6B2 SL0 Slope adjustment for R6B3 SL1 Slope adjustment for R6B4 LF0 Low frequency adjus R6B5 LF1 Low frequency adjus R6B6 LF2 Low frequency adjus R6B7 LF3 Low frequency adjus R7B0 P0 Pole adjustment for s R7B1 P1 Pole adjustment for s R7B3 P3 Pole adjustment for s R7B4 P4 Pole adjustment for s R7B5 AGARX0 Gain receive AGC L R7B7 AGARX2 Gain receive AGC L R7B7 AGARX2 Gain receive AGC L R7B8 EA2 Gain earpiece amplit R8B9 EA0 Gain earpiece amplit R8B1 EA1 Gain earpiece amplit R8B3 EA3 Gain earpiece amplit R8B4 EA4 Gain earpiece amplit R8B5 IMPH Line impedance sele R8B6 LOMAKE Short circuit during p	Maximum transmit level for anti-clipping	0	
		R5B6	MIC0	Gain microphone amplifier LSB	0
		R5B7	MIC1	Gain microphone amplifier MSB	0
		R6B0	SD	Shut down	0
		R6B1	Reserved	-	0
		R6B2	SL0	Slope adjustment for sidetone LSB	0
	Shut down	R6B3	SL1	Slope adjustment for sidetone MSB	0
R6	Sidetone	R6B4	LF0	Low frequency adjustment for sidetone LSB	0
		R6B5	LF1	Low frequency adjustment for sidetone	0
		R6B6	LF2	Low frequency adjustment for sidetone	0
		R6B7	LF3	Low frequency adjustment for sidetone MSB	0
		R7B0	P0	Pole adjustment for sidetone LSB	0
		R7B1	P1	Pole adjustment for sidetone	0
R5		R7B2	P2	Pole adjustment for sidetone	0
	Sidetone	R7B3	P3	Pole adjustment for sidetone	0
R6	AGARX	R7B4	P4	Pole adjustment for sidetone MSB	0
		R7B5	AGARX0	-	0
		R7B6	AGARX1	Gain receive AGC	0
					0
		R8B0	EA0	Gain earpiece amplifier LSB	0
					0
					0
					0
R8					0
	impedance				0
					0
				7.7	0





 Table 10-6.
 Names and Functions of the Serial Registers (Continued)

Register	Group	No.	Name	Description	Status
		R9B0	AFS0	AFS gain adjustment LSB	0
		R9B1	AFS1	AFS gain adjustment	0
		R9B2	AFS2	AFS gain adjustment	0
DO	AFC	R9B3	AFS3	AFS gain adjustment	0
R9	AFS	R9B4	AFS4	AFS gain adjustment	0
		R9B5	AFS5	AFS gain adjustment MSB	0
		R9B6	AFS4PS	Enable 4-point sensing	0
		R9B7	Reserved	-	0
		R10B0	SA0	Gain speaker amplifier LSB	0
		R10B1	SA1	Gain speaker amplifier	0
		R10B2	SA2	Gain speaker amplifier	0
R10	SA	R10B3	SA3	AFS1 AFS gain adjustment 0 AFS2 AFS gain adjustment 0 AFS3 AFS gain adjustment 0 AFS4 AFS gain adjustment MSB 0 AFS5 AFS gain adjustment MSB 0 AFS4PS Enable 4-point sensing 0 Reserved - 0 SA0 Gain speaker amplifier LSB 0 SA1 Gain speaker amplifier 0 SA2 Gain speaker amplifier 0 SA3 Gain speaker amplifier MSB 0 SE Speaker amplifier single-ended mode 0 LSCUR0 Speaker amplifier charge-current adjustment LSB 0 LSCUR1 Speaker amplifier charge-current adjustment MSB 0 LSCUR1 Speaker amplifier charge-current adjustment	0
HIU	SA	R10B4	SA4	Gain speaker amplifier MSB	0
		R10B5	SE	Speaker amplifier single-ended mode	0
		R10B6	LSCUR0	Speaker amplifier charge-current adjustment LSB	0
		R10B7	LSCUR1	Speaker amplifier charge-current adjustment MSB	0
		R11B0	ADC0	Input selection ADC	0
		R11B1	ADC1	Input selection ADC	0
		R11B2	ADC2	Input selection ADC	0
R11	ADC	R11B3	ADC3	Input selection ADC	0
חוו	ADC	R11B4	NWT	Network tuning	0
		R11B5	SOC	Start of ADC conversion	0
		R11B6	ADCR	Selection of ADC range	0
		R11B7	MSKIT	Mask for interrupt bits	0
		R12B0	DTMFF0	DTMF frequency selection	0
		R12B1	DTMFF1	DTMF frequency selection	0
		R12B2	DTMFF2	DTMF frequency selection	0
R12	DTMF	R12B3	DTMFF3	DTMF frequency selection	0
1112	BTIVII	R12B4	DTMFF4	DTMF frequency selection	0
		R12B5	DTMFM0	Generator mode selection	0
		R12B6	DTMFM1	Generator mode selection	0
		R12B7	DTMFM2	Generator mode selection	0
		R13B0	CLK0	Selection clock frequency for microcontroller	0
		R13B1	CLK1	Selection clock frequency for microcontroller	0
	OL IX	R13B2	RTH0	Ringer threshold adjustment LSB	0
R13	CLK RTH	R13B3	RTH1	Ringer threshold adjustment	0
1110	TM	R13B4	RTH2	Ringer threshold adjustment	0
		R13B5	RTH3	Ringer threshold adjustment MSB	0
		R13B6	TME0	Test mode enable (low active)	0
		R13B7	TME1	Test mode enable (high active)	0

Table 10-6. Names and Functions of the Serial Registers (Continued)

Register	Group	No.	Name	Description	Status
		R14B0	TME2	Test mode enable (high active)	0
		R14B1	TME3	Test mode enable (low active)	0
		R14B2	Reserved	-	0
R14	TM	R14B3	CLOR0	Adjustment for calculated receive log amp LSB	0
N14	CLOR	R14B4	CLOR1	Adjustment for calculated receive log amp	0
		R14B5	CLOR2	Adjustment for calculated receive log amp	0
		R14B6	CLOR3	Adjustment for calculated receive log amp	0
		R14B7	CLOR4	Adjustment for calculated receive log amp MSB	0
		R15B0	Reserved	-	0
		R15B1	Reserved	-	0
		R15B2	Reserved	-	0
D1E	CLOT	R15B3	CLOT0	Adjustment for calculated transmit log amp LSB	0
R15	CLOT	R15B4	CLOT1	Adjustment for calculated transmit log amp	0
		R15B5	CLOT2	Adjustment for calculated transmit log amp	0
		R15B6	CLOT3	Adjustment for calculated transmit log amp	0
		R15B7	CLOT4	Adjustment for calculated transmit log amp MSB	0

10.1 Power-on Reset

To avoid undefined states of the system when it is powered on, an internal reset clears the internal registers.

The system (U4091BM-R + microcontroller) is woken up by any of the following conditions:

- VMP > 2.75V and VB > 2.95V
- and line voltage (VL)
- or ringer (VRING)
- or external supply (ES)

The power-down of the circuit is caused by a shut-down sent by the serial bus (SD = 1), low-voltage reset, or by the watchdog function (see Figure 12-2 on page 17, Figure 12-3 on page 17 and Figure 12-4 on page 17).

11. Watchdog Function

To avoid the system operating the microcontroller in a fault state, the circuit provides a watchdog function. The watchdog has to be retriggered every second by triggering the serial bus (sending information to the IC or other remote components at the serial bus). If there has been no bus transmission for more than one second, the watchdog initiates a reset.

The watchdog provides a reset for the external microcontroller, but does not change the U4091BM-R's registers.





12. Acoustic Feedback Suppression

Acoustical feedback from the loudspeaker to the hands-free microphone may cause instability of the system. The U4091BM-R has a very efficient feedback-suppression circuit which offers a 4-point or (alternatively) a 2-point signal-sensing topology (see Figure 12-1).

Two attenuators (TXA and SAI) reduce the critical loop gain via the serial bus either in the transmit or in the receive path. The overall loop gain remains constant under all operating conditions.

The LOGs produce a logarithmically-compressed signal of the TX- and RX-envelope curve. The AFSCON block determines whether the TX or the RX signal has to be attenuated.

The voice-switch topology can be selected by the serial bus. In 2-point-sensing mode, AFSCON is controlled directly by the LOG outputs.

Figure 12-1. Basic System Configurations

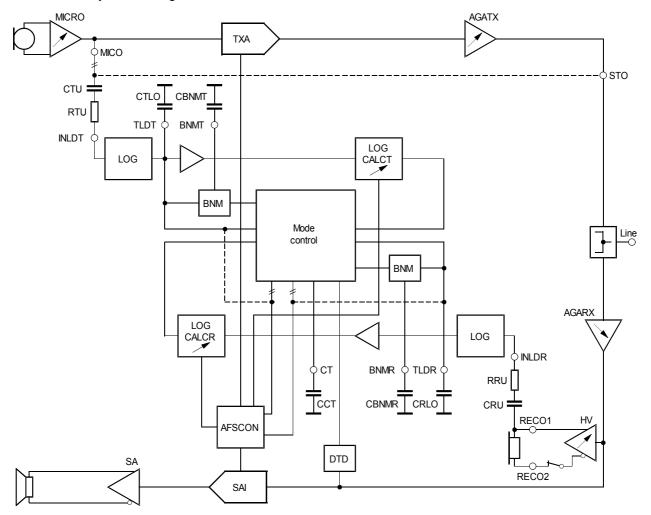


Figure 12-2. Power-on Reset (Line)

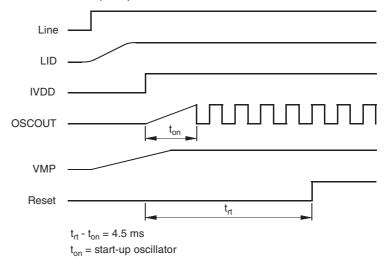


Figure 12-3. Power-on Reset (Ringing)

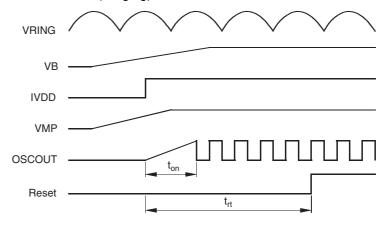
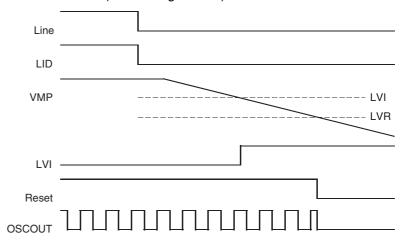


Figure 12-4. Power-on Reset (Low Voltage Reset)





12.1 Dial-tone Detector

The dial-tone detector is a comparator with one side connected to the speaker amplifier input and the other to VM with a 35-mV offset (see Figure 12-5 on page 21). If the circuit is in idle mode, and the incoming signal is greater than 35 mV (25 mV $_{rms}$), the comparator's output will change thus disabling the receive idle mode. This circuit prevents the dial tone (which would be considered as continuous noise) from fading away as the circuit would have the tendency to switch to idle mode. By disabling the receive idle mode, the dial tone remains at the normally expected full level.

12.2 Background Noise Monitors

This circuit distinguishes speech (which consists of bursts) from background noise (a relatively constant signal level). There are two background-noise monitors, one for the receive path and the other for the transmit path. The receive background-noise monitor is operated on by the receive level detector, while the transmit background noise monitor is operated on by the transmit level detector (see Figure 12-6 on page 21). They monitor the background noise by storing a DC voltage representative of the respective noise levels in capacitors at CBNMR and CBNMT. The voltages at these pins have slow rise times (determined by the internal current source and an external capacitor), but fast decay times. If the signal at TLDR (or TLDT) changes slowly, the voltage at BNMR (or BNMT) will remain more positive than the voltage at the non-inverting input of the monitor's output comparator. When speech is present, the voltage at the non-inverting input of the comparator will rise more quickly than the voltage at the inverting input (due to the burst characteristic of speech), causing its output to change. This output is sensed by the mode-control block.

12.3 4-point Sensing

In 4-point-sensing mode, the receive- and the transmit-sensing paths include additional CLOGs (calculated logarithmic amplifiers). The block MODECON compares the detector output signals and decides whether receive, transmit or idle mode has to be activated. Depending on the mode decision, MODECON generates a differential voltage to control AFSCON.

The MODECON block has seven inputs:

- The output of the transmit log (LOGT) the comparison of LOGT, CLOGR
- The output of the receive clog (CLOGR) designated I1
- The output of the transmit clog (CLOGT) the comparison of CLOGT, LOGR
- The output of the receive log (LOGR) designated I2
- The output of the transmit background-noise monitor (BNMT) designated I3
- The output of the receive background-noise monitor (BNMR) designated I4
- The output of the dial-tone detector

The differential output (AFST, AFSR) of the block MODECON controls AFSCON. The effect of I1-I4 in Table 12-1 on page 19.

Table 12-1. Mode Decision for Signal Sensing

	Input					
l1	12	13	14	Mode		
Т	Т	S	X	Transmit		
Т	R	Υ	Υ	Change mode		
R	Т	Υ	Υ	Change mode		
R	R	X	S	Receive		
Т	Т	N	X	Idle		
Т	R	N	N	Idle		
R	Т	N	N	Idle		
R	R	X	N	Idle		

Note: X = don't care; Y = I3 and I4 are not both noise.

I1 = TLOGT > CLOGR LOGT < CLOGR I1 = RLOGR < CLOGT I2 = TLOGR > CLOGT I2 = RBNMT detects speech 13 = SBNMT detects noise I3 = NBNMR detects speech 14 = SI4 = NBNMR detects noise

12.4 Term Definitions

- 1. Transmit means the transmit attenuator is fully on, and the receive attenuator is at maximum attenuation.
- 2. Receive means the receive attenuator is fully on, and the transmit attenuator is at maximum attenuation.
- 3. In idle mode, the transmit and receive attenuator are at half of their maximum attenuation.
 - Change mode means both the transmit and receive speech are present in approximately equal levels. The attenuators are quickly switched (30 ms) to the opposite mode until one speech level dominates the other.
 - Idle means speech has ceased in both transmit and receive paths. The attenuators are then slowly switched (1.5s) to idle mode.
- 4. Switching to full transmit or receive modes from the idle mode is done at a fast rate (30 ms).





12.5 Summary of Truth Table

- 1. The circuit will switch to transmit mode if
 - Both transmit level detectors sense higher signal levels than the respective receive level detectors, and
 - The transmit background-noise monitor indicates the presence of speech
- 2. The circuit will switch to receive mode if
 - Both receive level detectors sense higher signal levels than the respective transmit level detectors, and
 - The receive background-noise monitor indicates the presence of speech
- 3. The circuit will switch to the reverse mode if
 - The level detectors disagree on the relative strengths of the signal levels, and
 - At least one of the background-noise monitors indicates speech
- 4. The circuit will switch to idle mode when
 - Both speakers are quiet (no speech present), or
 - When one speaker speech level is continuously overridden by noise at the other speaker's location

The time required to switch the circuit between transmit, receive and idle is determined by internal current sources and the capacitor at pin CT. A diagram of the CT circuitry is shown in Figure 12-7 on page 21. It operates as follows:

- CCT is typically 4.7 μF.
- To switch to transmit mode, ITX is turned on (IRX is off), charging the external capacitor to -240 mV below VM. (An internal clamp prevents further charging of the capacitor.)
- To switch to receive mode, IRX is turned on (ITX is off), increasing the voltage on the capacitor to +240 mV with respect to VM.
- To switch to reverse mode, the current sources ITX, IRX are turned off, and the current source IFI is switched on, discharging the capacitor to VM.
- To switch to idle mode, the current sources ITX, IRX, IFI are turned off, and the current source ISI charges the capacitor to VM.

Figure 12-5. Dial Tone Detector

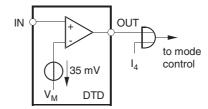


Figure 12-6. Background Noise Monitor

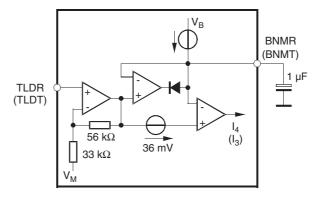


Figure 12-7. Generation of Control Voltage (CT) for Mode Switching

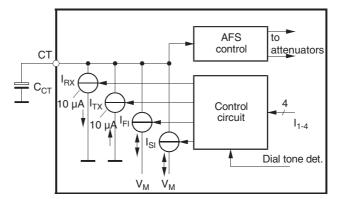




Figure 12-8. Block Diagram Hands-free Mode U4091BM-R 2-point Signal Sensing

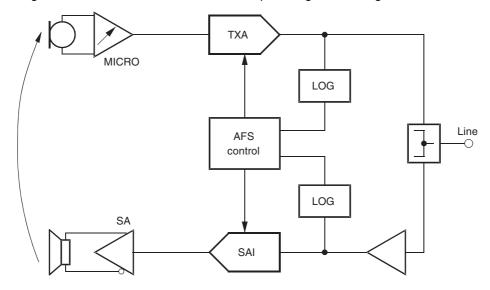
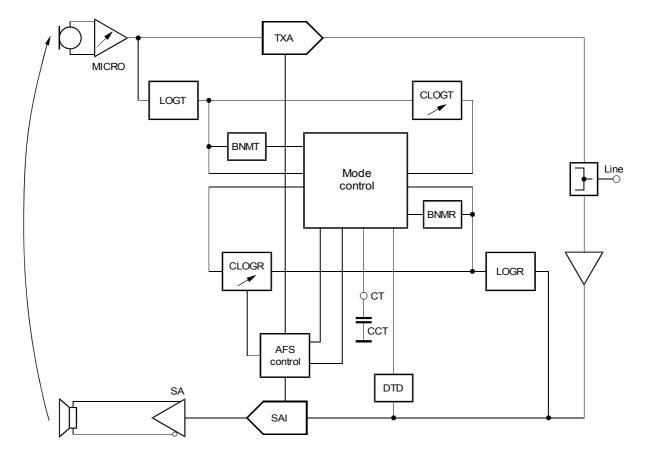


Figure 12-9. Block Diagram Hands-free Mode U4091BM-R 4-point Signal Sensing



13. Analog-to-Digital Converter (ADC)

This circuit is a 7-bit successive-approximation analog-to-digital converter in switched capacitor technique. An internal band gap circuit generates a 1.25-V reference voltage which is the equivalent of 1 MSB (1 LSB = 19.5 mV). The possible input voltage at ADIN is 0V to 2.48V.

The ADC needs an SOC (Start Of Conversion) signal. In the High phase of the SOC signal, the ADC is reset. Then, $50~\mu s$ after the beginning of the Low phase of the SOC signal, the ADC generates an EOC (End Of Conversion) signal which indicates that the conversion is finished. The rising edge of EOC generates an interrupt at the INT output. The result can be read out by the serial bus.

Voltages higher than 2.45V have to be divided. The signal connected to the ADC is determined by 4 bits: ADC0, ADC1, ADC2 and ADC3. TLDR/TLDT measuring is possible relative to a preceding reference measurement. The current range of IL can be doubled by ADCR. If ADCR is High, S has the value 0.5, otherwise S = 1.

The source impedance at ADIN must be lower than 250 $k\Omega$

Accuracy: 1 LSB + 3%

Figure 13-1. Timing of ADC

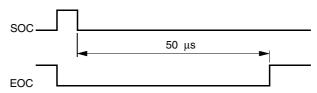
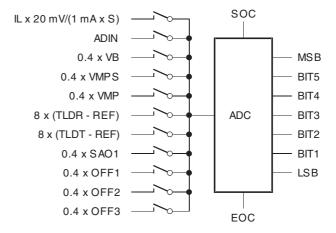


Figure 13-2. ADC Input Selection





Input Selection ADC Table 13-1.

Decimal	ADC[1:4]	Symbol	Value
0	0000	OFF	-
1	0001	IL	I1 = S × 127 mA × D / 128
2	0010	ADIN extern	V2 = 2.5V × D / 128 (maximum 2.5V)
3	0011	VB	V3 = (2.5V / 0.4) × D / 128
4	0100	VMPS	V4 = (2.5V / 0.4) × D / 128
5	0101	VMP	V5 = (2.5V / 0.4) × D / 128
6	0110	TLDR	$V6 = 8 \times (V_p - Ref) \times D / 128$
7	0111	TLDT	$V7 = 8 \times (V_p - Ref) \times D / 128$
8	1000	Not used	-
9	1001	SAO1	V4 = (2.5V / 0.4) × D / 128
10	1010	Offcan1	Atmel's internal use
11	1011	Offcan2	-
12	1100	Offcan3	-
13	1101	Not used	-
14	1110	Not used	-
15	1111	Not used	-

D = measured digital word ($0 \le D \le 127$) Note:

S = programmable gain 0.5 or 1 $V_p = peak value of the measured signal$

14. Switch Matrix

The switch matrix has 5 inputs and 5 outputs. Every pair of I/Os except AGCO and AGCIN can be connected. The inputs and outputs used must be enabled. If 2 or more inputs are switched to an output, the sum of the inputs is available at the output.

The inputs MIC and LRX have offset cancellers with a 3-dB corner frequency of 270Hz. AMPB has a $60\text{-k}\Omega$ input impedance. The TXO output has a digitally-programmable gain stage with a gain of 2 dB to 9 dB (in 1 dB steps) depending on AGATX0 (LSB), AGATX1, AGATX2 (MSB), and a first order low-pass filter with 0.5 dB damping at 3300Hz and 3 dB damping at 9450Hz. The outputs RXLS, EPO and AMREC have a gain of 0 dB. The offset at the outputs of the matrix is less than 30 mV. If a switch is open, the path has a damping of more than 60 dB.

Figure 14-1. Switch Matrix Diagram

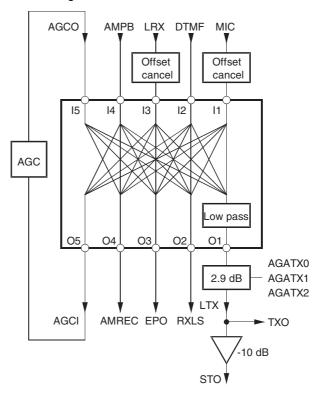


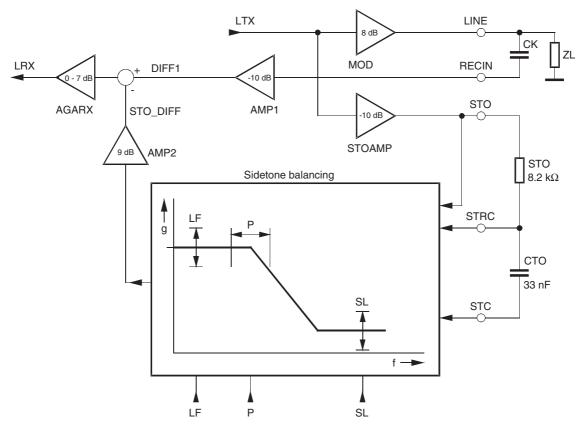


Table 14-1. Bits and Corresponding Switches

Register	No.	Name	Description
	R2B0	I1O1	Switch on MIC/LTX
	R2B1	I1O2	Switch on MIC/RXLS
	R2B2	I1O3	Switch on MIC/EPO
R2	R2B3	I1O4	Switch on MIC/AMREC
nz	R2B4	I1O5	Switch on MIC/AGCI
	R2B5	I2O1	Switch on DTMF/LTX
	R2B6	1202	Switch on DTMF/RXLS
	R2B7	1203	Switch on DTMF/EPO
	R3B0	1204	Switch on DTMF/AMREC
	R3B1	1205	Switch on DTMF/AGCI
	R3B2	I3O1	Switch on LRX/LTX
R3	R3B3	1302	Switch on LRX/RXLS
no	R3B4	13O3	Switch on LRX/EPO
	R3B5	1304	Switch on LRX/AMREC
	R3B6	1305	Switch on LRX/AGCI
	R3B7	I4O1	Switch on AMPB/LTX
	R4B0	1402	Switch on AMPB/RXLS
	R4B1	1403	Switch on AMPB/EPO
	R4B2	1404	Switch on AMPB/AMREC
R4	R4B3	I4O5	Switch on AMPB/AGCI
N4	R4B4	I5O1	Switch on AGCO/LTX
	R4B5	1502	Switch on AGCO/RXLS
	R4B6	15O3	Switch on AGCO/EPO
	R4B7	15O4	Switch on AGCO/AMREC

15. Sidetone System

Figure 15-1. Principle Circuit of Sidetone Balancing



The Sidetone Balancing (STB) has the task of reducing the cross-talk from LTX (microphone) to LRX (earpiece) in the frequency range of 0.3 kHz to 3.4 kHz. The LTX signal is converted into a current in the MOD block. This current is transformed into a voltage signal (LINE) by the line impedance ZL. The LINE signal is fed into the summing amplifier DIFF1 via capacitor CK and attenuator AMP1.

On the other hand the LTX buffered by STOAMP drives an external low-pass filter (RST, CST). The external low-pass filter and the internal STB have the transfer function drawn in the STB box. The amplified STB output signal drives the negative input of the summing block. If both signals at the DIFF1 block are equal in level and phase, we have good suppression of the LTX signal. In this condition, the frequency and phase response of the STB block will represent the frequency curve on line.

In real life, the line impedance ZL varies strongly for different users. To obtain good suppression with one application for all different line impedances, the STB function is programmable.



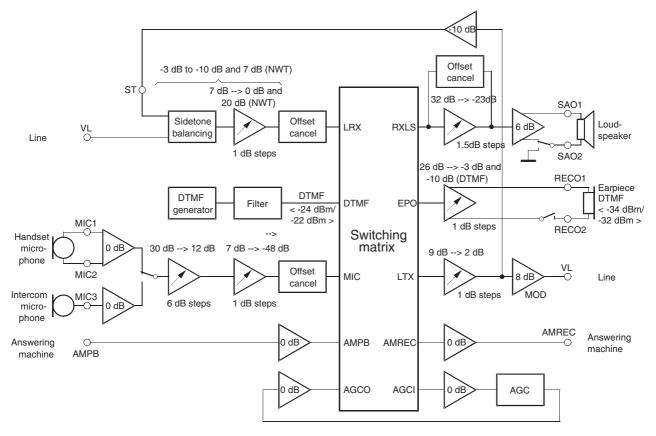
The 3 programmable parameters are

- LF (gain at low frequency)
 LF has 15 programming steps of 0.5 dB
 LF(0) provides -2 dB gain, LF(15) provides 5.5 dB gain
 STO_DIFF(LF) = (-10 dB 2 dB + 0.5 dB × LF + 9 dB) × LTX
- P (the pole position of the low-pass)
 The P adjustment has 31 steps. P(0) means the lowpass determined by the external application (RST, CST). The internally processed low-pass frequency is fixed by the following equation.

$$f(P) = \frac{1}{2 \times \pi \times CST \times RST} \times 1.122^{P}$$

3. SL (sidetone slope; the pole frequency of the high-pass)
The SL has 3 steps. SL(0) is a lower frequency of the high-pass. SL(3) is a higher frequency of the high-pass. SL can be used to influence the suppression at high frequencies.

Figure 15-2. Audio Frequency Signal Management U4091BM-R



16. Absolute Maximum Ratings

Stresses beyond those listed under "Absolute Maximum Ratings" may cause permanent damage to the device. This is a stress rating only and functional operation of the device at these or any other conditions beyond those indicated in the operational sections of this specification is not implied. Exposure to absolute maximum rating conditions for extended periods may affect device reliability.

Parameters	Symbol	Value	Unit
Line current	ال	140	mA
DC line voltage	V _L	12	V
Maximum input current	I _{RING}	15	mA
Junction temperature	T _j	125	°C
Ambient temperature	T _{amb}	−25 to +75	°C
Storage temperature	T _{stg}	-55 to +150	°C
Total power dissipation, T _{amb} = 60°C	P _{tot}	0.9	W

17. Thermal Resistance

Parameters	Symbol	Value	Unit	
Junction ambient SSO44	R _{thJA}	70	K/W	

18. Electrical Characteristics

f = 1 kHz, 0 dBm = 775 mV_{rms}, IVMIC = 0.3 mA, IMP = 3 mA, R_{DC} = 1.3 M Ω , T_{amb} = 25°C, Z_{ear} = 68 nF + 100 Ω , RLS = 50 Ω , Z_M = 68 nF, resonator: f = 3.58 MHz, all bits in reset condition, unless otherwise specified.

Parameters	Test Conditions	Symbol	Min.	Тур.	Max.	Unit
DC Characteristics						
DC voltage drop over circuit	I _L = 2 mA I _L = 14 mA I _L = 60 mA I _L = 100 mA	V _L	4.4 8.6	1.6 4.8 7.2 9.2	5.2 9.8	V V V
Transmission Amplifier, $I_L = 14 \text{ mA}$ ERX = ETX = ENMIC = ENSTBAL =	, V _{MIC} = 2 mV, MICG[0:1] = 2, AGATX[0: I1O1 = I3O3 = 1, (G _T = 48 dB)	:2] = 7	•			
Transmit amplification	MICG[0:1] = 2 AGATX[0:2] = 7	G _T	45.3	46.5	47.7	dB
Frequency response due to internal filters	$I_L \ge 14 \text{ mA},$ f = 1 kHz to 3.4 kHz	ΔG_T	-1		0	dB
Gain change with current	I _L = 14 mA to 100 mA	ΔG_T			±0.5	dB
Gain deviation	$T_{amb} = -10^{\circ}C \text{ to } +60^{\circ}C$	ΔG_{T}			±0.5	dB
CMRR of microphone amplifier		CMRR	60	80		dB
Input resistance of MIC amplifier		R _i		50		kΩ
Input resistance of MIC3 amplifier	MICHF = 1	R _i	75	150	300	kΩ
Gain difference between MIC1/MIC2 to MIC3	MICHF = 1	ΔG_T			±0.4	dB
Distortion at line	$I_L \ge 14 \text{ mA}, V_L = 700 \text{ mV}_{rms}$	d _t			2	%
	$I_L \ge 19 \text{ mA}, d < 5\%, V_{MIC} = 10 \text{ mV}$ CTXA = 1 μ F, DBM5 = 0	V _{Lmax}	1.8	3.0	4.2	dBm
Maximum output voltage	DBM5 = 1	V_{Lmax}	4.8	6.0	6.6	dBm
	V _{MIC} = 20 mV, MICG[0:1] = 3	V _{MICOmax}		-4.2		dBm





f = 1 kHz, 0 dBm = 775 mV_{rms}, IVMIC = 0.3 mA, IMP = 3 mA, R_{DC} = 1.3 MΩ, T_{amb} = 25°C, Z_{ear} = 68 nF + 100Ω, RLS = 50Ω, Z_{M} = 68 nF, resonator: f = 3.58 MHz, all bits in reset condition, unless otherwise specified.

Parameters	Test Conditions	Symbol	Min.	Тур.	Max.	Unit
Noise at line psophometrically weighted	I _L ≥ 14 mA, MICG[0:1] = 2 AGATX[0:2] = 7	No		- 73	-70	dBmp
Anti-clipping attack time release time	CTXA = 1 µF each 3 dB overdrive	t _a		2 80		ms ms
Gain at low operating current	$\label{eq:loss_loss} \begin{split} I_L &= 8 \text{ mA}, \ I_{MP} = 1 \text{ mA} \\ V_{MIC} &= 0.5 \text{ mV} \\ I_{VMIC} &= 300 \mu\text{A} \end{split}$	G_{T}	45		48	dB
Distortion at low operating current	$I_L = 8$ mA, $I_{MP} = 1$ mA $V_{MIC} = 5$ mV $I_{VMIC} = 300$ μ A	d _t			5	%
Receiving Amplifier $I_L = 14 \text{ mA}, V_{GEN} = 300 \text{ mV},$ ERX = ETX = ENMIC = ENSTBAL = I^{T}	IO1 = I3O3 = 1, SL[0:1] = 0, LF[0:3] :	= 1, P[0:4] = 31,	AFS[0:5] =	: 54, AGAF	RX[0:2] = 0)
Adjustment range of receiving gain	Single ended, $I_L \ge 14$ mA, Mute = 1, $EA[0:4] = 2$ to 31 $AGARX[0:2] = 0$ to 7	G _R	-19		+17	dB
Receiving amplification	Differential AGARX[0:2] = 0 EA[0:4] = 15 EA[0:4] = 31	G _R	-1 14.7	0 15.7	1 16.7	dB dB
Frequency response	$I_L \ge 14$ mA, f = 1 kHz to 3.4 kHz	ΔG_{RF}	-1		0	dB
Gain change with current	I _L = 14 mA to 100 mA	ΔG_{R}			±0.5	dB
Gain deviation	$T_{amb} = -10^{\circ}C \text{ to } +60^{\circ}C$	ΔG_R			±0.5	dB
Ear protection differential	$I_L \ge 14 \text{ mA},$ $V_{GEN} = 11 \text{ V}_{rms}$ EA[0:4] = 15	EP			3	V _{rms}
MUTE suppression (earpiece disconnect from matrix)	I _L = 14 mA, I303 = 0	ΔG_R	60			dB
Output voltage d < 2% differential	$I_L = 14 \text{ mA}$ $Z_{ear} = 68 \text{ nF} + 100\Omega$ $EA[0:4] = 11$		0.775			V_{rms}
Maximum output current d < 2%	$Z_{ear} = 100\Omega$ EA[0:4] = 31	I _{out}	4			mA _p
Receiving noise psophometrically weighted	$I_L = 14 \text{ mA}$ $Z_{ear} = 68 \text{ nF} + 100\Omega$ $EA[0:4] = 15$			– 79	-76	dBmp
Sidetone suppression	$Z = 600\Omega$		20			dB
Output resistance	Each output against GND	R _o			10	Ω
Gain at low operating current (receive only)	$I_L = 6.5 \text{ mA}, I_{MP} = 1 \text{ mA}$ $I_M = 300 \text{ mA}$ $V_{GEN} = 200 \text{ mV}$ EA[0:4] = 21, ENMIC = ETX = I101 = 0	G _R	-2	0	2	dB

Max.

Unit

Тур.

18. Electrical Characteristics (Continued)

Parameters

f = 1 kHz, 0 dBm = 775 mV_{rms}, IVMIC = 0.3 mA, IMP = 3 mA, R_{DC} = 1.3 MΩ, T_{amb} = 25°C, Z_{ear} = 68 nF + 100Ω, RLS = 50Ω, Z_{M} = 68 nF, resonator: f = 3.58 MHz, all bits in reset condition, unless otherwise specified.

Test Conditions

Distortion at low operating current	$I_L = 6.5$ mA, $I_{MP} = 1$ mA $I_M = 300$ μ A, EA[0:4] = 15, ENMIC = ETX = I101 = 0	dR			5	%
Adjustment step: earpiece amplifier	DEA[0:4] = 1 for EA[0:4] = 2 to 31		0.8	1	1.2	dB
Adjustment step: AGARX	DAGARX[0:2] = 1		0.8	1	1.2	dB
Gain for DTMF signal	AMPB → RECO1/2 EA[0:4] = 1			-10		dB
AC impedance	IMPH = 0 IMPH = 1	Z _{impl} Z _{imph}	595 980	625 1030	655 1080	Ω Ω
DTMF, I _L = 14 mA, ETX = I201 = 1, A	GATX[0:2] = 7, DTMFM[0:2] = 4, DTMF	F[0:4] = 0	•		•	•
DTMF level at line (mid gain)	Sum level, 600Ω , DTMFM[0:2] = 4		-5.1	-3.6	-2.1	dBm
DTMF level at line (low gain)	Sum level, 600Ω , DTMFM[0:2] = 5		-7.6	-6.1	-4.6	dBm
DTMF level at line (high gain)	Sum level, 600Ω, DTMFM[0:2] = 6 AGATX[0:2] = 1		-5.2	-3.7	-2.2	dBm
Pre-emphasis	600 W, DTMFF4 = 0 DTMFF4 = 1		2	2.5 3.5	3 4	dBm dBm
ERX = ETX = ENMIC = ENSTBAL = I Minimum line current for operation	ENAM = I4O2 = 0 SE = 0, I3O2 = 1	I _{Lmin}			11	mA
	IMP 1 mA, $V_{GEN} = 300 \text{ mV}$ $V_{AMPB} = 3 \text{ mV}, I_L = 15 \text{ mA},$				11	mA
Gain from AMPB to SAO	SA[0:4] = 31 SA[0:4] = 0	G _{SA}	36	37 –5.5	38	dB
Adjustment step speaker amplifier	DSA[0:4] = -1		1.15	1.35	1.55	dB
Output power single ended	Load resistance: $R_{LS} = 50\Omega$ $V_{AMPB} = 40 \text{ mV, SE} = 1$ $I_{L} = 15 \text{ mA}$ $I_{L} = 20 \text{ mA}$	P _{SA} P _{SA}	3	7 20		mW mW
Maximum output power differential	Load resistance: $R_L = 50\Omega$ $V_{AMPB} = 60 \text{ mV, SE} = 0$ $V_B = 5 \text{V}$	P _{SA}		150		mW
Output noise (input AMPB open) psophometrically weighted	I _L > 15 mA	n _{SA}			240	mV_{psoph}
		· ·SA				psoph





f = 1 kHz, 0 dBm = 775 mV_{rms}, IVMIC = 0.3 mA, IMP = 3 mA, R_{DC} = 1.3 MΩ, T_{amb} = 25°C, Z_{ear} = 68 nF + 100Ω, RLS = 50Ω, Z_{M} = 68 nF, resonator: f = 3.58 MHz, all bits in reset condition, unless otherwise specified.

Parameters	Test Conditions	Symbol	Min.	Тур.	Max.	Unit
Mute suppression	$I_L = 15 \text{ mA}, V_L = 0 \text{ dBm},$ $V_{AMPB} = 4 \text{ mV}$ I4O2 = 0	VSAO			-56	dBm
Gain change with current	I _L = 15 mA to 100 mA	ΔG_SA			1	dB
Gain change with frequency	I _L = 15 mA f = 1 kHz to 3.4 kHz	ΔG_SA	-1		0	dB
Attack time of anti-clipping	20 dB overdrive	t _r		2		ms
Release time of anti-clipping		t _f		170		ms
Adjustment step of charge current	ENSAO = 0, SE = 1 DLSCUR[0:1] = 1		-480	-400	-320	μΑ
Adjustment step of discharge current	ENSAO = 0, SE = 0 DLSCUR[0:1] = 1		320	400	480	μΑ
Charge current Pin SAO2	ENSAO = 0, SE = 1 LSCUR[0:1] = 3	I _{CHA}	-1.45	-1.2	-0.95	mA
Discharge current pin SAO2	ENSAO = 0, SE = 0 LSCUR[0:1] = 3	I _{DIS}	0.95	1.2	1.45	mA
Microphone Amplifier, $V_B = 5V$, $V_{MIC} = 2$ mV, $V_{MIC3} = 2$ mV, I	ENMIC = ENAM = I104 = 1, MICHF	= 0				
	MICG[0:1] = 0		17.4	18.1	18.8	dB
Gain MIC amp.:	MICG[0:1] = 1		23.2	23.7	24.6	dB
MIC1/2 →AMREC	MICG[0:1] = 2		29.1	29.8	30.5	dB
	MICG[0:1] = 3		35.0	35.7	36.4	dB
MIC3 to AMREC	MICHF = 1, MICG[0:1] = 3		35.0	35.7	36.5	dB
Input suppression: MIC3 to MIC1/2	MICG[0:1] = 0, MICHF = 0		60			dB
MIC1/2 to MIC3	MICHF = 1		60			dB
Settling time offset cancellers	5 τ, FOFFC = 0			9	12	ms
Settling time offset cancellers in speed-up mode	5 τ, FOFFC = 1			1.8	2.4	ms
AGC for Answering Machine, AMPI ENAM = ENAGC = I4O5 = I5O4 = 1	B to AMREC,	·				
Nominal gain	V _{AMPB} = 5 mV		23.5	25.5	27.5	dB
Maximum output level	$V_{AMPB} = 50 \text{ mV}, d < 5\%$		240	300	360	mVp
Attack time	20 dB overdrive			1		ms
Release time				45		ms
Switching Matrix, VL = 0, VB = 5V, ENAM = I4O4 = 1, V	$V_{AMPB} = 0.6 V_{rms}$					
Input impedance AMPB			50	60	70	kΩ
Gain AMPB to AMREC			-0.7	-0.3	+0.1	dB
Maximum input level AMPB	I4O5 = I5O4 = 1, I4O4 = 0				600	mV
Maximum output level AMREC	1404 = 1				VB – 600 mV	V _{PP}

f = 1 kHz, 0 dBm = 775 mV_{rms}, IVMIC = 0.3 mA, IMP = 3 mA, R_{DC} = 1.3 MΩ, T_{amb} = 25°C, Z_{ear} = 68 nF + 100Ω, RLS = 50Ω, Z_{M} = 68 nF, resonator: f = 3.58 MHz, all bits in reset condition, unless otherwise specified.

Parameters	Test Conditions	Symbol	Min.	Тур.	Max.	Unit
Offset	I4O4: 1 to 0	ΔV_{AMREC}			±30	mV
Mute switching matrix	14O4 = 0		60			dB
Power-on Reset $VL = 0$, $V_{MP} = 3.3V$, $V_{B} = 5V$, U4091 i	n Power-down Mode					
Power-on reset by ES VB high, VMP threshold	VB = 4V, ES = 4V, raise VMP until RESET goes to low	VMP _{on}	2.65	2.75	2.85	V
Power-on reset by ES VMP high, VB threshold	VMP = 3V, ES = 4V, raise VB until RESET goes to low	VB _{on}		3.2		V
Low-voltage Interrupt VL = 0, V _{MP} = 3.3V, V _B = 0V						
VMP decreasing	Decrease VMP until INT returns to high	VLVI	2.5	2.6	2.7	V
Power-off Reset $VL = 0$, $V_{MP} = 3.3V$, $V_{B} = 0V$						
Low-voltage reset	Decrease VMP until RESET returns to low	VLVR	2.35	2.45	2.55	V
Difference voltage between low- voltage interrupt and reset	VLVI – VLVR		100	150		mV
Logical Part V _{MP} = 3.3V, V _B = 5V						
Output impedance at OSCOUT			0.6	0.9	1.2	kΩ
Pins SCL, SDA (input mode) Input leakage current	Low level High level 0 < V _i < V _{MP}		0.8 × V _{MP} -1		0.2 × V _{MP} +1	V V μA
Pins INT, SDA (output mode)	Output low (resistance to GND)		150	230	350	Ω
Switch for Additional Impedance (F $V_{MP} = 3.3V$, $V_B = 3V$	Pin IMPSW)					
Switch-off leakage current	$0 < V_i < V_{MP}$ $IMPSW = 0$		-0.5		5	μΑ
Resistance to GND	IMPSW = 1			50	80	Ω
Maximum current	IMPSW = 1		-5		5	mA
AFS (Acoustic Feedback Suppress ERX = ETX = ENMIC = ENSTBAL =	sion), I _L = 14 mA, V _{GEN} = 300 mV, I1O1 = I3O3 = 1, SL[0:1] = 0, LF[0:3] = 1	, P[0:4] = 31,	AGARX[0:	2] = 0		
Adjustment range of attenuation	I _L ≥ 15 mA		0		50	dB
Attenuation of transmit gain	$I_L \ge 15$ mA, $I_{INLDT} = 0$ μ A $I_{INLDR} = 10$ μ A	ΔG_T	47	50	53	dB
Attenuation of speaker amplifier	$I_L \ge 15$ mA, $I_{INLDT} = 10$ μ A $I_{INLDR} = 0$ μ A	G _{SA}	47	50	53	dB





f = 1 kHz, 0 dBm = 775 mV_{rms}, IVMIC = 0.3 mA, IMP = 3 mA, R_{DC} = 1.3 MΩ, T_{amb} = 25°C, Z_{ear} = 68 nF + 100Ω, RLS = 50Ω, Z_{M} = 68 nF, resonator: f = 3.58 MHz, all bits in reset condition, unless otherwise specified.

Parameters	Test Conditions	Symbol	Min.	Тур.	Max.	Unit
Supply Voltages, V _{MIC} = 25 mV,	$T_{amb} = -10^{\circ}C \text{ to } + 60^{\circ}C$			ч	1	
V _{MP}	I_L = 14 mA, R_{DC} = 680 k Ω I_{MP} = 3 mA	V _{MP}	3.1	3.3	3.5	V
V _{MPS}	$I_L = 100 \text{ mA}, R_{DC} = \text{inf.},$ $I_{MP} = 0 \text{ mA}$	V _{MPS}			5.5	V
V _{MIC}	I_L = 14 mA, R_{DC} = 1.3 M Ω I_M = 700 A	V _{MIC}	1.5		4	V
V_B	$I_B = +20 \text{ mA}, I_L = 0 \text{ mA}$	V _B		5.5	6.3	V
Ringing Power Converter, IMP	= 1 mA, IM = 0, R_{IMPA} = 500 $k\Omega$					
Maximum output power	$V_{RING} = 20.6V$ ENSA = ENSAO = SE = 1	P _{SA}		15		mW
	V _{RING} : high to low				7.4	V
Threshold	Low to high, RINGTH [0:3] = 0		6.0	6.7	7.4	V
	Low to high, RINGTH [0:3] = 15		19	21	23	V
Adjustment steps threshold	DRINGTH = 1		0.8	1	1.2	V
Input impedance	$V_{RING} = 30V$		4.6	5.8	7.0	kΩ
Maximum input voltage		V _{RINGmax}	30			V
Serial Bus SCL, SDA, AS, VMP	= 3.3V, RSDA = RSCL = RINT = 12 $k\Omega$					
Input voltage HIGH LOW	SDA, SCL, INT	V _{iBUS}	3.0		V _{DD} 1.5	V
Output voltage Acknowledge LOW	SDA I _{SDA} = 3 mA	V _O			0.4	V
Clock frequency	SCL	f _{SCL}			100	kHz
Rise time SDA, SCL		t _r			1	μs
Fall time SDA, SCL		t _f			300	ns
Period of SCL HIGH LOW	HIGH LOW	t _H	4.0 4.7			μs μs
Setup Time						
Start condition Data Stop condition Time space ⁽¹⁾		t _{sSTA} t _{sDAT} t _{sSTOP} t _{wSTA}	4.7 250 4.7 4.7			μs ns μs μs
Hold Time		·				
Start condition DATA		t _{hSTA} t _{hDAT}	4.0 0			μs μs

19. Test Circuits

Figure 19-1. Basic Test Circuit

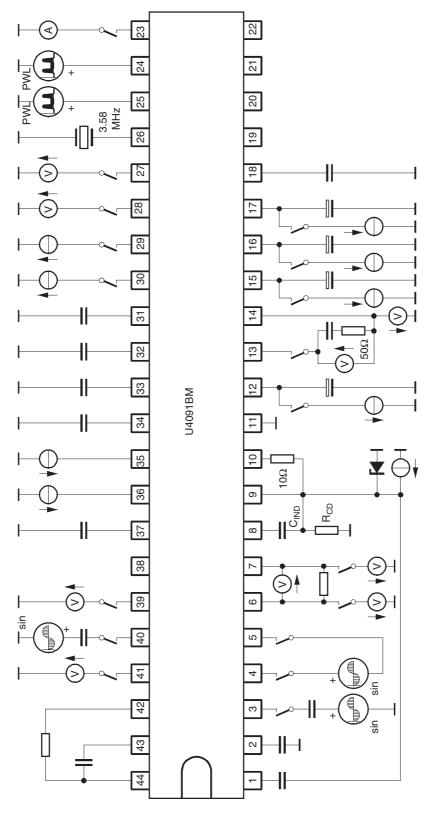
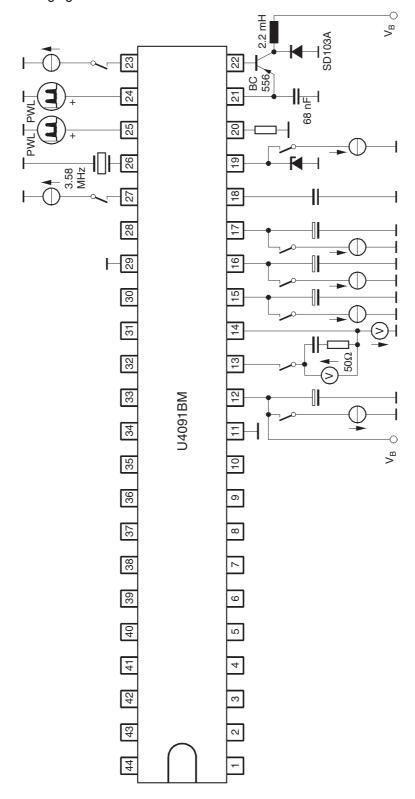




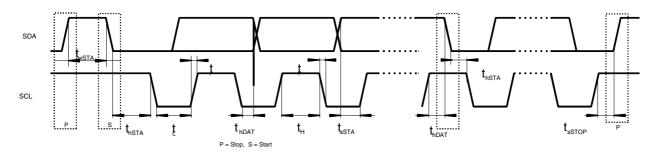


Figure 19-2. Test Circuit for Ringing



20. Bus Timing

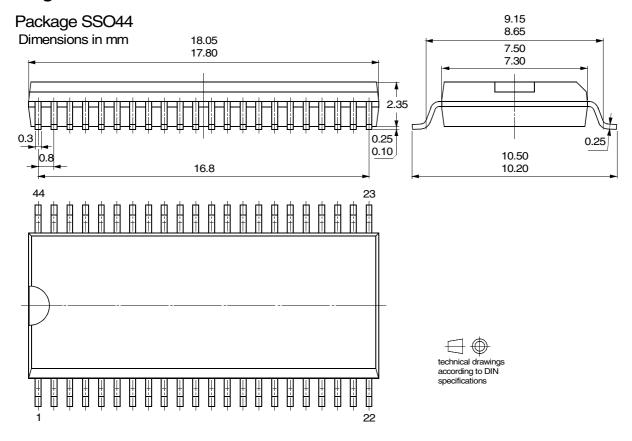
Figure 20-1. Bus Timing Diagram



21. Ordering Information

Extended Type Number	Package	Remarks
U4091BM-RFNGY	SSO44	Tube
U4091BM-RFNG3Y	SSO44	Taped and reeled
T4091R-DDB	Die	Die on foil

22. Package Information







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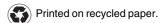
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