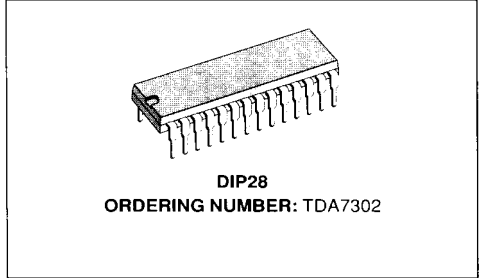


**DIGITAL CONTROLLED STEREO AUDIO PROCESSOR**

- INPUT AND OUTPUT PINS FOR EXTERNAL EQUALIZER
- THREE STEREO INPUT SOURCE SELECTION PLUS MONO INPUT
- TREBLE, BASS, VOLUME AND BALANCE CONTROL
- FOUR INDEPENDENT SPEAKER CONTROL (FRONT/REAR)
- SINGLE SUPPLY OPERATION
- ALL FUNCTIONS PROGRAMMABLE VIA SERIAL BUS
- VERY LOW NOISE AND VERY LOW DISTORTION
- POP FREE SWITCHING



and Hi-Fi system.

Control is accomplished by serial bus microprocessor interface.

The AC signal setting is obtained by resistor networks and analog switches combined with operational amplifiers.

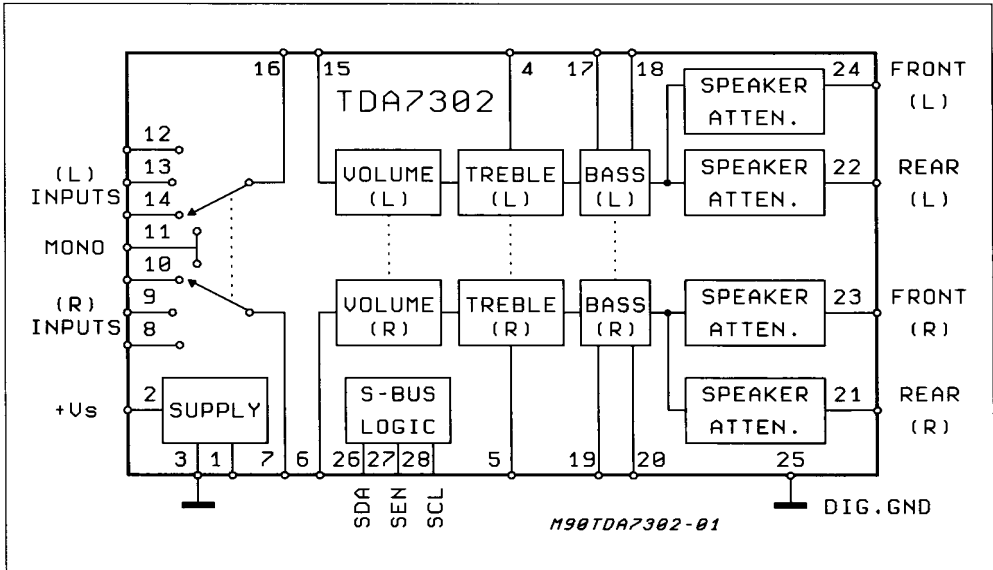
The results are: low noise, low distortion and high dynamic range.

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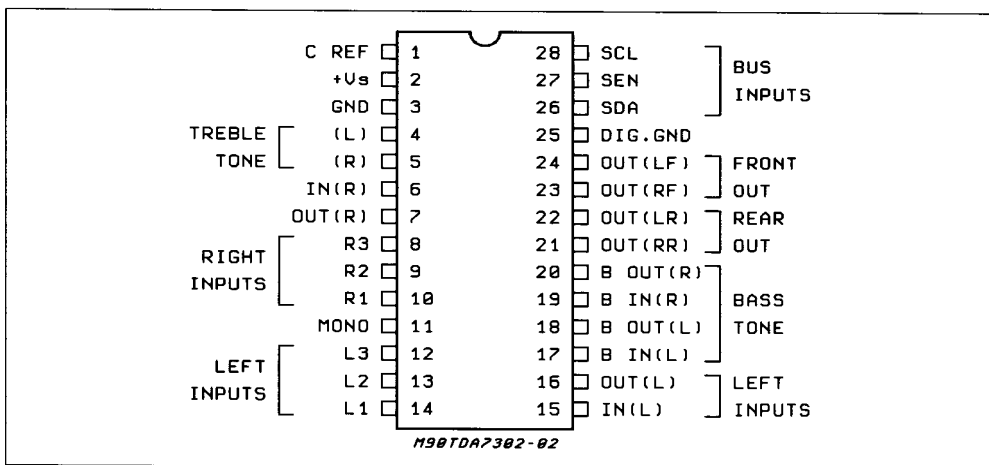
**DESCRIPTION**

The TDA7302 is a volume, tone (bass and treble), balance (left/right) and fader (front/rear) processor for high quality audio applications in car radio

**BLOCK DIAGRAM**



## PIN CONNECTION (Top view)



## ABSOLUTE MAXIMUM RATINGS

Symbol	Parameter	Value	Unit
$V_S$	Supply Voltage	14	V
$P_{tot}$	Total Power Dissipation $T_{amb} = 25^\circ\text{C}$	2	W
$T_{amb}$	Operating Ambient Temperature Range	-40 to +85	$^\circ\text{C}$
$T_{stg}$	Storage Temperature	-40 to 150	$^\circ\text{C}$

## THERMAL DATA

Symbol	Description	Value	Unit
$R_{th\ j-pins}$	Thermal Resistance Junction-pins	65	$^\circ\text{C}/\text{W}$

ELECTRICAL CHARACTERISTICS ( $T_{amb} = 25^\circ\text{C}$ ,  $V_S = 10\text{V}$ ,  $R_L = 10\text{k}\Omega$ ,  $R_g = 600\Omega$ ,  $f = 1\text{KHz}$  unless otherwise specified)

Symbol	Parameter	Test Condition	Min.	Typ.	Max.	Unit
$V_S$	Supply Voltage		6	10	14	V
$I_S$	Supply Current		15	30	40	mA
SVR	Ripple Rejection	$f = 300\text{Hz to } 10\text{KHz}$	50	58		dB

## INPUT SELECTORS

Symbol	Parameter	Test Condition	Min.	Typ.	Max.	Unit
$R_i$	Input Resistance		30	45		$\text{K}\Omega$
$V_{IN\ max}$	Max. Input Signal	$GV = 0\text{dB}$ $d = 0.3\%$	1.5	2.2		$V_{rms}$
$IN_S$	Input Separation	$f = 1\text{KHz}$ (2)	90	100		dB
		$f = 10\text{KHz}$ (2)	70	80		dB
$R_L$	Output Load Resistance		5			$\text{K}\Omega$
$V_i\ (DC)$	Input DC Voltage		3.5	4.3	5	V

## ELECTRICAL CHARACTERISTICS (continued)

Symbol	Parameter	Test Condition	Min.	Typ.	Max.	Unit
<b>VOLUME CONTROLS</b>						
$R_{IN}$	Input Resistance		5	10	20	K $\Omega$
	Control Range			78		mA
$G_{max}$	Max Gain		8	10	12	dB
	Max Attenuation		64	68		dB
	Step Resolution			2	3	dB
	Attenuator Set Error	$G_V = -50$ to 10dB			2	dB
	Tracking Error				2	dB

## SPEAKER ATTENUATORS

	Control Range		35	38	41	dB
	Step Resolution			2	3	dB
	Attenuator Set Error				2	dB
	Tracking Error				2	dB

## BASS AND TREBLE CONTROL (1)

	Control Range			$\pm 15$		dB
	Step Resolution			2.5	3.5	dB

## AUDIO OUTPUT

$V_O$	Max. Output Voltage	$d = 0.3\%$	1.5	2.2		Vrms
$R_L$	Output Load Resistance		2			K $\Omega$
$C_L$	Output Load Capacitance				1	nF
$R_O$	Output Resistance			70	150	$\Omega$
$V_O(DC)$	DC Voltage Level		3	3.8	4.5	V

## GENERAL

$e_{NO}$	Output Noise	BW = 22Hz to 22KHz	$G_V = 0$ dB	6	15	$\mu V$
			Out atten. $\geq 20$ dB	3.5		
			$G_V = 0$ dB	4		
S/N	Signal to Noise Ratio	All gain = 0dB $V_O = 1V_{rms}$ BW = 22Hz to 22KHz		105		dB
d	Distortion	$f = 1$ KHz $V_O = 1V$ $G_V = 0$		0.01	0.1	%
	Frequency Response (-1dB)	$G_V = 0$	High Low	20	20	KHz Hz
$S_C$	Channnel Separation left/right	$f = 1$ KHz		90	100	dB
		$f = 10$ KHz		70	80	dB

## BUS INPUTS

$V_{IL}$	Input LOW Voltage				0.8	V
$V_{IH}$	Input HIGH Voltage		2.4			V
$V_O$	Output Voltage SDA Acknowledge	$I = 1.6$ mA			0.4	V
	Digital Input Current		-5		+5	$\mu A$

## Notes:

- (1) Bass and Treble response see attached diagram. The center frequency and quality of the resonance behaviour can be chosen by the external circuitry. A standard first order bass response can be realized by a standard feedback network.
- (2) The selected input is grounded thru the 2.2 $\mu F$  capacitor.

Figure 1: Application Circuit

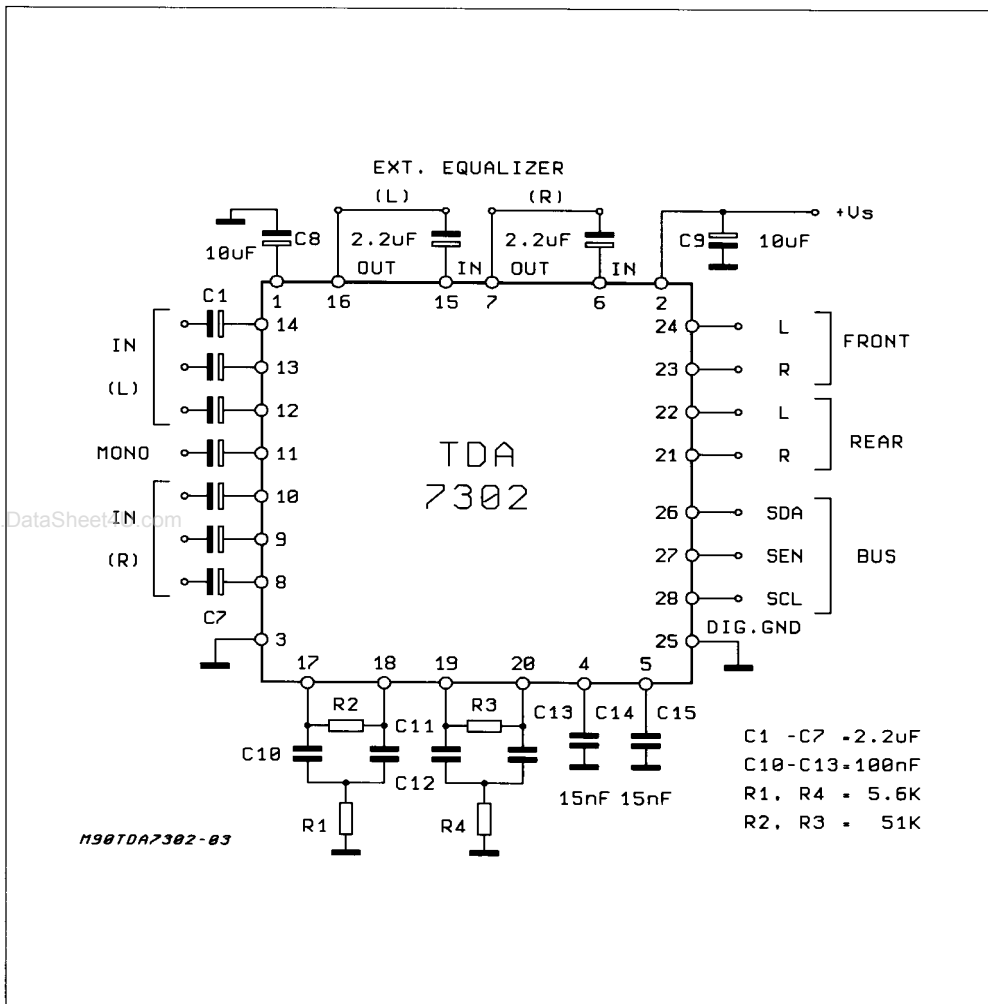


Figure 2: P.C. Board and Components Layout of the Fig.1 (1:1 scale)

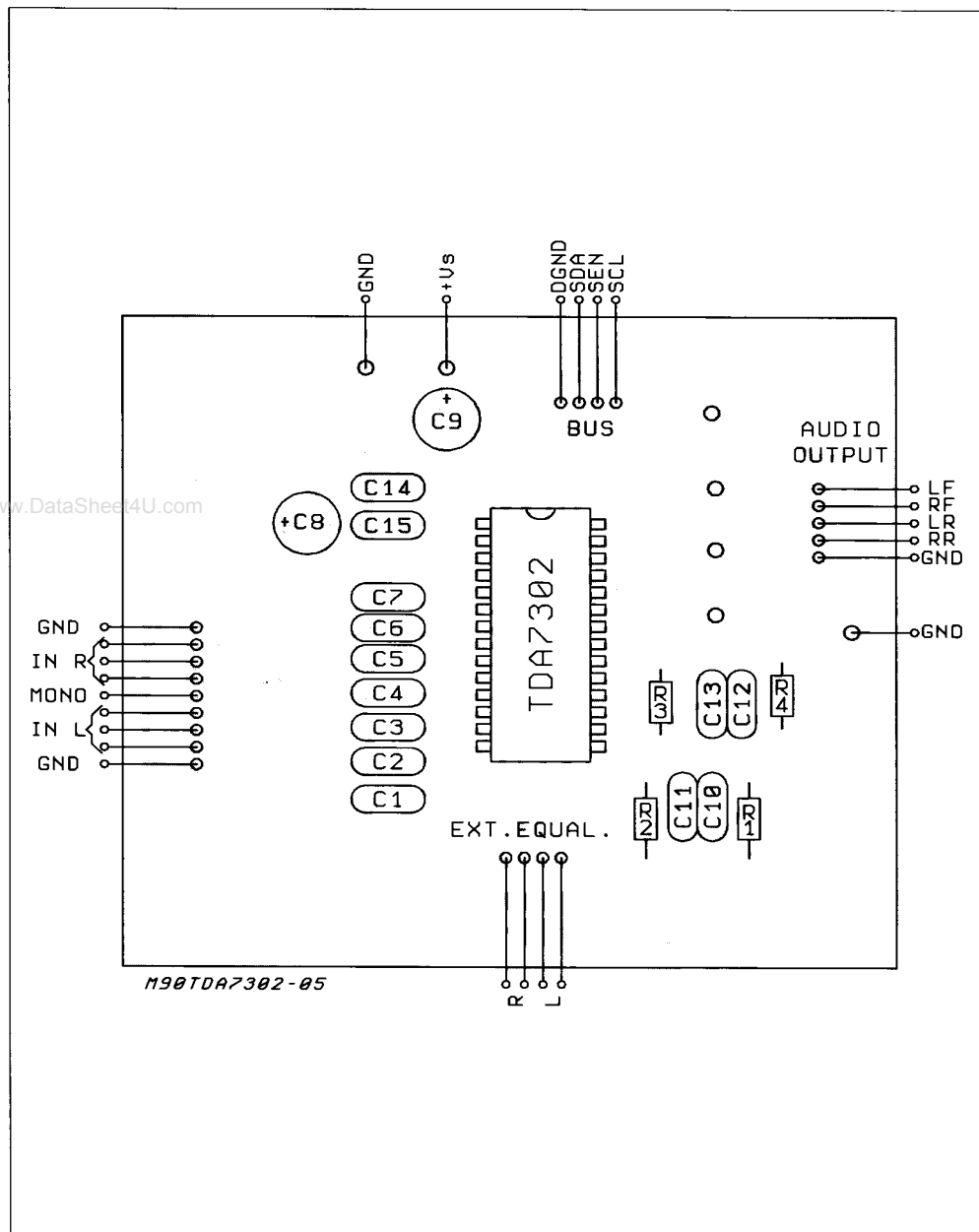


Figure 3: Total Output Noise vs. Volume Setting

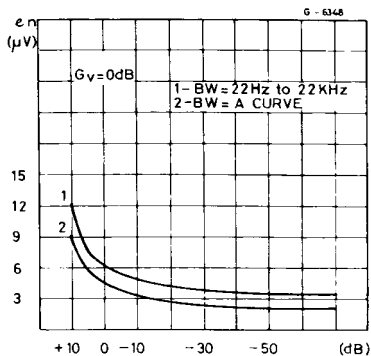


Figure 4: Signal to Noise Ratio vs. Volume Setting

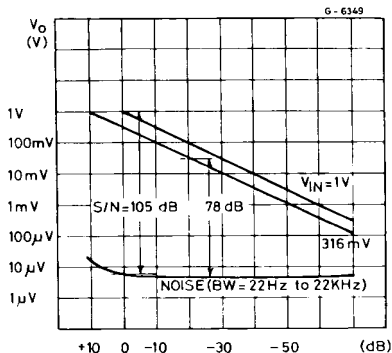


Figure 5: Distortion + Noise vs. Frequency

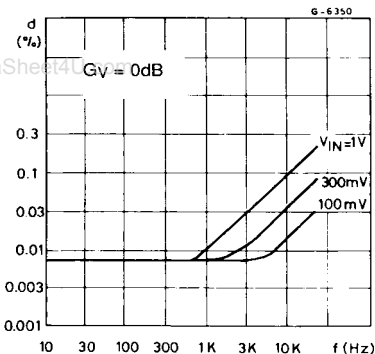


Figure 6: Distortion vs. Output Voltage

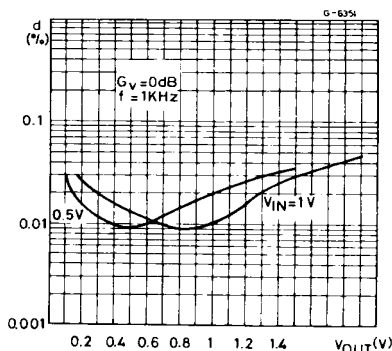


Figure 7: Distortion vs. Load Resistance

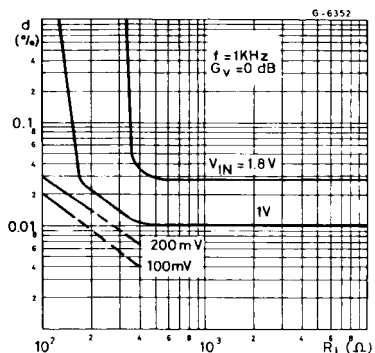
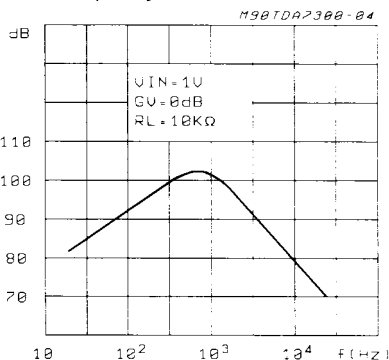
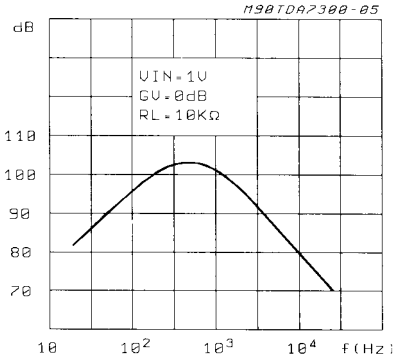


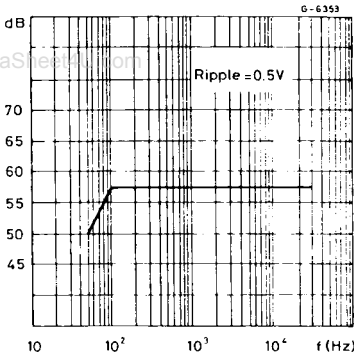
Figure 8: Channel Separation (L1 - R1) vs. Frequency



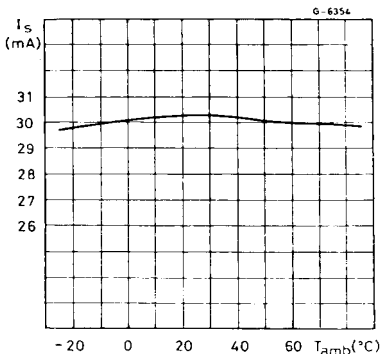
**Figure 9: Input Separation (L1 - L2) vs. Frequency**



**Figure 10: Supply Voltage Rejection vs. Frequency**



**Figure 11: Quiescent Current vs. Temperature**



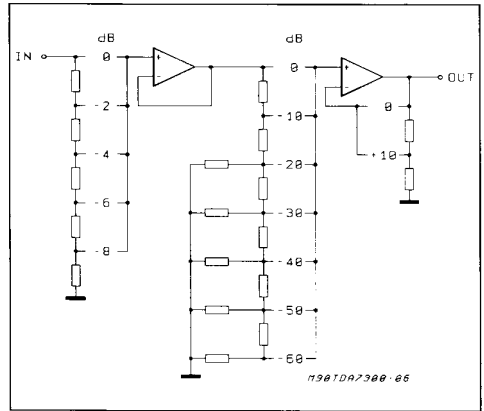
**APPLICATION INFORMATION**

**Volume Control Concept**

Traditional electronic volume control circuits use a multiplier technique with all the disadvantages of high noise and distortion.

The used concept, as shown in Fig. 12 with digital switched resistor dividers, provides extremely low noise and distortion. The multiplexing of the resistive dividers is realized with a multiple-input operational amplifier.

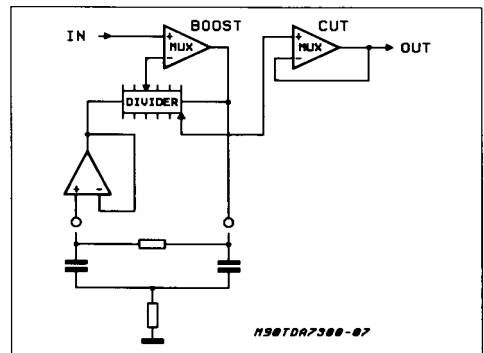
**Figure 12: Volume Control**



**Bass and Treble Control**

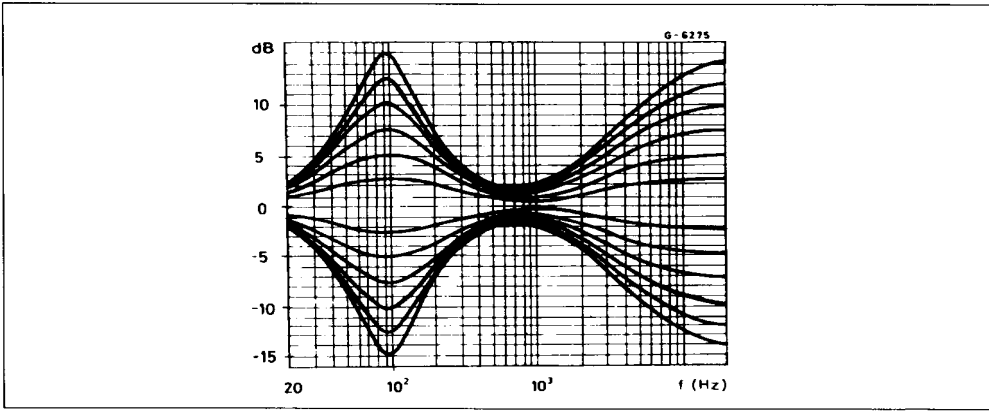
The principle operation of the bass control is shown in Fig. 12. The external filter together with the internal buffer allows a flexible filter design according to the different requirements in car radios. The function of the treble is similar to the bass. A typical curve is shown in Fig.14.

**Figure 13: Bass Control**



APPLICATION INFORMATION (continued)

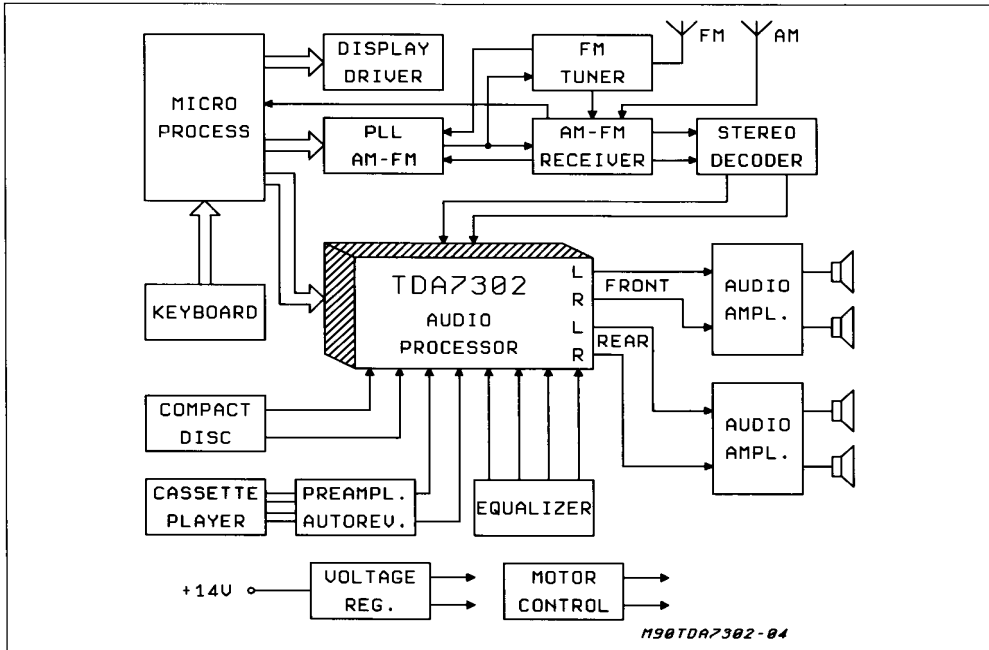
Figure 14: Typical Tone Response



Outputs

A special class-A output amplifier with a modulated sink current provides low distortion and ground compatibility with low current consumption.

Figure 15: Complete Car-Radio System using Digital Controlled Audio Processor





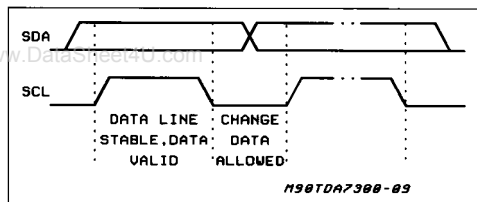
**APPLICATION INFORMATION** (continued)**SERIAL BUS INTERFACE****S-BUS Interface and I<sup>2</sup>C BUS Compatibility**

Data transmission from microprocessor to the TDA7302 and viceversa takes place thru the 3-wire S-BUS interface, consisting of the three lines SDA, SCL, SEN. If SDA and SEN inputs are short-circuited together, then the TDA7302 appears as a standard I<sup>2</sup>C BUS slave.

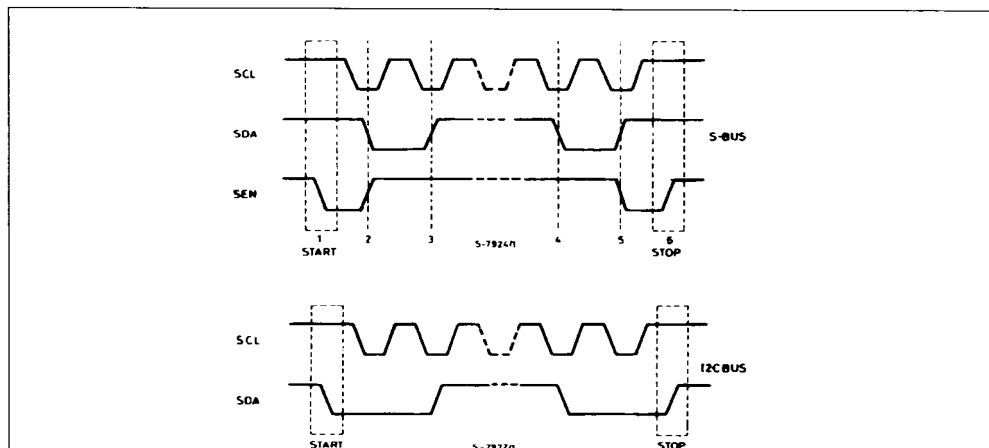
According to I<sup>2</sup>C BUS specification the S-BUS lines are connected to a positive supply voltage via pull-up resistors

**Data Validity**

As shown in fig. 16, the data on the SDA line must be stable during the high period of the clock. The HIGH and LOW state of the data line can only change when the clock signal on the SCL line is LOW.

**Figure 16:** Data Validity on the I<sup>2</sup>C BUS**Start and Stop Conditions****I<sup>2</sup>C BUS:**

as shown in fig.17 a start condition is a HIGH to

**Figure 17:** Timing Diagram of S-BUS and I<sup>2</sup>C BUS

LOW transition of the SDA line while SCL is HIGH. The stop condition is a LOW to HIGH transition of the SDA line while SCL is HIGH.

**S-bus:**

the start/stop conditions (points 1 and 6) are detected exclusively by a transition of the SEN line (1 → 0 / 0 → 1) while the SCL line is at the HIGH level.

The SDA line is only allowed to change during the time the SCL line is low (points 2, 3, 4, 5). After the start information (point 1) the SEN line returns to the HIGH level and remains unchanged for all the time the transmission is performed.

**Byte Format**

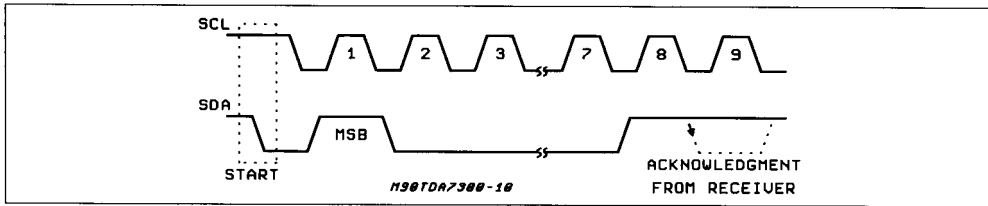
Every byte transferred on the SDA line must contain 8 bits. Each byte must be followed by an acknowledge bit. The MSB is transferred first.

**Acknowledge**

The master ( $\mu$ P) puts a resistive HIGH level on the SDA line during the acknowledge clock pulse (see fig. 18). The peripheral (audioprocessor) that acknowledges has to pull-down (LOW) the SDA line during the acknowledge clock pulse, so that the SDA line is stable LOW during this clock pulse.

The audioprocessor which has been addressed has to generate an acknowledge after the reception of each byte, otherwise the SDA line remains at the HIGH level during the ninth clock pulse time. In this case the master transmitter can generate the STOP information in order to abort the transfer.

Figure 18: Acknowledge on the I<sup>2</sup>C BUS



**Transmission without Acknowledge**

Avoiding to detect the acknowledge of the audioprocessor, the  $\mu$ P can use a simpler transmission: simply it waits one clock without checking the slave acknowledging, and sends the new data.

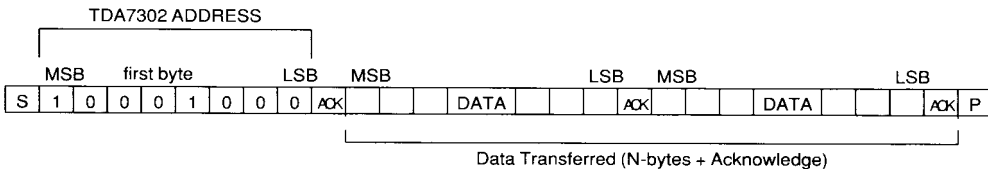
This approach of course is less protected from misworking and decreases the noise immunity.

**Interface Protocol**

The interface protocol comprises:

- A start condition (S)
- A chip address byte, containing the TDA7302 address (the 8th bit of the byte must be 0). The TDA7302 must always acknowledge at the end of each transmitted byte.
- A sequence of data (N-bytes + acknowledge)
- A stop condition (P)

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ACK = Acknowledge  
S = Start  
P = Stop

MAX CLOCK SPEED 100kbits/s

**SOFTWARE SPECIFICATION**

Chip address (TDA7302 address)

1 0 0 0 1 0 0 0  
MSB LSB

**DATA BYTES**

MSB	LSB	Function
0 0 B2 B1 B0 A2 A1 A0		Volume Control
1 1 0 B1 B0 A2 A1 A0		Speaker ATT LR
1 1 1 B1 B0 A2 A1 A0		Speaker ATT RR
1 0 0 B1 B0 A2 A1 A0		Speaker ATT LF
1 0 1 B1 B0 A2 A1 A0		Speaker ATT RF
0 1 0 X X S2 S1 S0		Audio switch
0 1 1 0 C3 C2 C1 C0		Bass control
0 1 1 1 C3 C2 C1 C0		Treble control

X = don't care  
Ax = 2dB steps  
Bx = 10dB steps  
Cx = 2.5dB steps

**Status after power-on reset**

**STATUS AFTER POWER-ON-RESET**

Volume	- 68 dB
Speaker	- 38 dB
Audio Switch	Mono
Bass	+ 2.5 dB
Treble	+ 2.5 dB

**SOFTWARE SPECIFICATION** (continued)  
**DATA BYTES** (detailed description)

**VOLUME**

MSB				LSB				
0	0	B2	B1	B0	A2	A1	A0	Volume 2dB Steps
					0	0	0	0
					0	0	1	-2
					0	1	0	-4
					0	1	1	-6
					1	0	0	-8
					1	0	1	Not allowed
					1	1	0	Not allowed
					1	1	1	Not allowed
0	0	B2	B1	B0				Volume 10dB STEPS
		0	0	0				+10
		0	0	1				0
		0	1	0				-10
		0	1	1				-20
		1	0	0				-30
		1	0	1				-40
		1	1	0				-50
		1	1	1				-60

For example if you want setting the volume at -32dB the 8 bit string is: 0 0 1 0 0 0 0 1

**SPEAKER ATTENUATORS**

MSB				LSB				
1	0	0	B1	B0	A2	A1	A0	Speaker LF
1	0	1	B1	B0	A2	A1	A0	Speaker RF
1	1	0	B1	B0	A2	A1	A0	Speaker LR
1	1	1	B1	B0	A2	A1	A0	Speaker RR
					0	0	0	0
					0	0	1	-2
					0	1	0	-4
					0	1	1	-6
					1	0	0	-8
					1	0	1	Not allowed
					1	1	0	Not allowed
					1	1	1	Not allowed
		0	0					0
		0	1					-10
		1	0					-20
		1	1					-30

For example attenuation of 24dB on speaker RF is given by: 1 0 1 1 0 0 1 0

**SOFTWARE SPECIFICATION** (continued)

AUDIO SWITCH - Select the input Channel to Activate

MSB			LSB					
0	1	0	X	X	S2	S1	S0	Audio Switch
			X	X	0	0	0	Stereo 1
			X	X	0	0	1	Stereo 2
			X	X	0	1	0	Stereo 3
			X	X	0	1	1	Mute Input
			X	X	1	0	0	Mono
			X	X	1	0	1	Not Allowed
			X	X	1	1	0	Not Allowed
			X	X	1	1	1	Not Allowed

X = don't care

For example to set the stereo 2 channel the 8 bit string must be: 0 1 0 0 0 0 1

BASS AND TREBLE - Control Range of  $\pm 15$  dB (boost and cut) Steps of 2.5 dB

0	1	1	0	C3	C2	C1	C0	Bass Treble
0	1	1	0	C3	C2	C1	C0	
				0	0	0	0	- 15
				0	0	0	1	- 15
				0	0	1	0	- 12.5
				0	0	1	1	- 10
				0	1	0	0	- 7.5
				0	1	0	1	- 5
				0	1	1	0	- 2.5
				0	1	1	1	- 0
				1	1	1	1	+ 0
				1	1	1	0	+ 2.5
				1	1	0	1	+ 5
				1	1	0	0	+ 7.5
				1	0	1	1	+ 10
				1	0	1	0	+ 12.5
				1	0	0	1	+ 15
				1	0	0	0	+ 15

C3 = Sign

For example Bass at -12.5dB is obtained by the following 8 bit string: 0 1 1 0 0 1 0

Purchase of I<sup>2</sup>C Components of SGS-THOMSON Microelectronics, conveys a license under the Philips I<sup>2</sup>C Patent Rights to use these components in an I<sup>2</sup>C system, provided that the system conforms to the I<sup>2</sup>C Standard Specifications as defined by Philips.