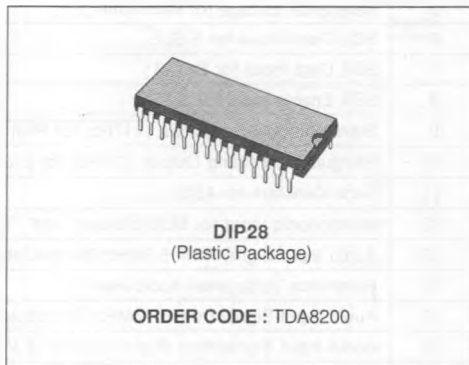


TV STEREO DECODER

ADVANCE DATA

- IDENTIFICATION OF TRANSMISSION MODE (mono/stereo/bilingual)
- DEMATRIXING OF THE STEREO AUDIO SIGNAL, WITH AN INTERNAL PROGRAMMABLE S-BUS NETWORK FOR MINIMAL CROSS-TALK
- DE-EMPHASIS OF THE AUDIO SIGNAL WITHOUT EXTERNAL COMPONENTS
- FILTERS FOR PSEUDOSTEREO AND ENLARGED STEREO BASE SPECIAL EFFECTS
- MONOPHONIC INPUT FOR MULTISTANDARD APPLICATIONS
- STEREO INPUT/OUTPUT FOR VCR
- VOLUME AND BALANCE CONTROL FOR EARPHONE OUTPUT
- ALL FUNCTION PROGRAMMABLE THROUGH USE OF S-BUS

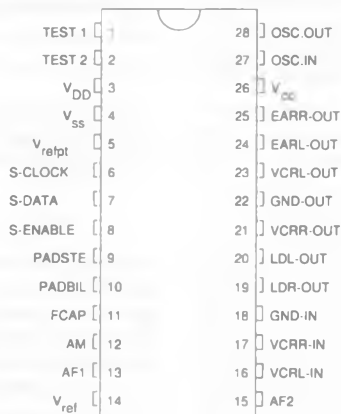


DESCRIPTION

The TDA8200 combines the functions of audio dematrixing and stereo decoder for the European 2-

carrier B/G system ; moreover, the device also includes input and a stereophonic output for reception of the audio signal coming from the SCART connector as well as a monophonic input for MULTISTANDARD applications (e.g., B/G and L).

PIN CONNECTION

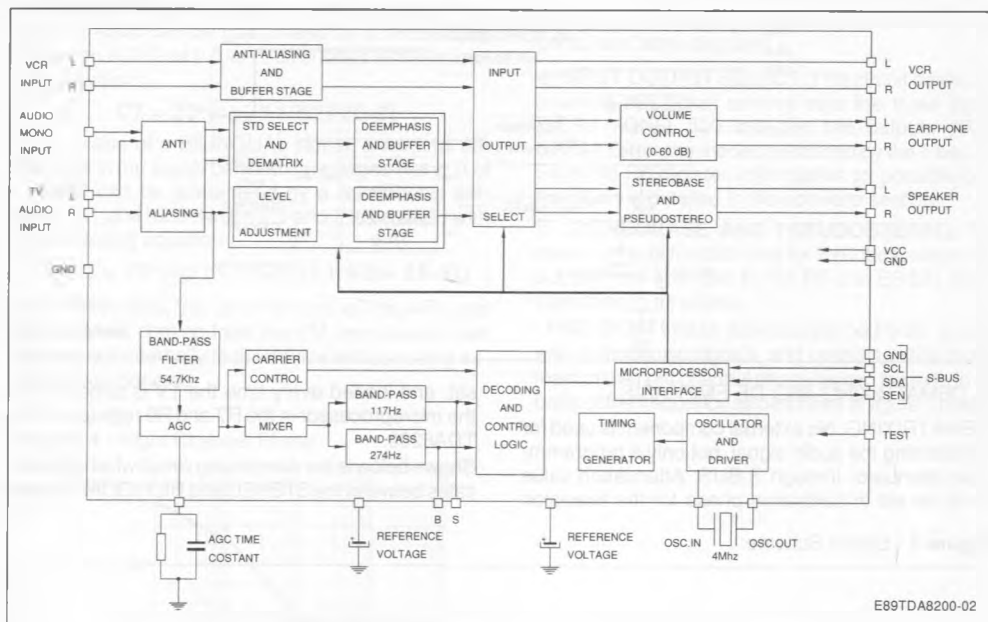


E89TDA8200-01

PIN DESCRIPTION

Pin	Function
1	Reserved for device testing, normally should be connected to mass.
2	Reserved for device testing, normally should be unconnected.
3	Power Voltage for Digital Part and Pilot Tone $V_d = 8.5V$
4	Mass for Digital Part and for Pilot Tone
5	Reference Voltage for Pilot Tone
6	SCL Clock Input for S-BUS
7	SDA Data Input for S-BUS
8	SEN Enable Input for S-BUS
9	Stereo Modulating Output (117Hz) for Pilot Tone
10	Bilingual Modulating Output (274Hz) for pilot Tone
11	Time Constant for AGC
12	Monophonic Input for Multistandard "AM"
13	Audio Input Signal from 5.5MHz Demodulator, AF1
14	Reference Voltage for Audio Part
15	Audio Input Signal from 5.74MHz Demodulator, AF2
16	Audio Input Signal from Right Channel of VCRR-IN Videotape Machine
17	Audio Input Signal from Left Channel of VCRL-IN Videotape Machine
18	Mass for Audio Input Signal
19	Audio Output Signal from Right Channel of LDR-OUT Loudspeakers
20	Audio Output Signal from Left Channel of LDL-OUT Loudspeakers
21	Audio Output Signal from Right Channel of VCRR-OUT Videotape Machine
22	Mass for Audio Output Signal
23	Audio Output Signal from Left Channel of VCRL-OUT Videotape Machine
24	Audio Output Signal from Right Channel of EARL-OUT Earphones
25	Audio Output Signal from Left Channel of EARR-OUT Earphones
26	Input Voltage for Analog Part $V_a = 8.5V$
27	Oscillator Input at 4MHz
28	Oscillator Output at 4MHz

BLOCK DIAGRAM



CIRCUIT DESCRIPTION

The device is made up of 5 main sections :

- 1 S-BUS interface
- 2 Oscillator and power on reset
- 3 Pilot tone decoder
- 4 Dematrixing and de-emphasis of the audio signal
- 5 Reception of the audio signal

1) S-BUS INTERFACE

All of the TDA8200 functions are activated by micro-processors, using a 3-wire serial bus (SCL, SDA, SEN). For further information, see the software section.

2) OSCILLATOR AND POWER ON RESET

OSCILLATOR. The device functions with an external 4MHz crystal quartz connected to pins 27 and 28. A possible alternative is to connect pin 27 to an external 4MHz generator. If the clock frequency does not remain stable, there will be variations in the audio response and pilot tone decoding.

POWER ON RESET. About 120ms after VCC power has reached the required level for correct device operation, the power on reset circuit signals correct operating status, using the RES bit in the RR

register. For further information, see the software section.

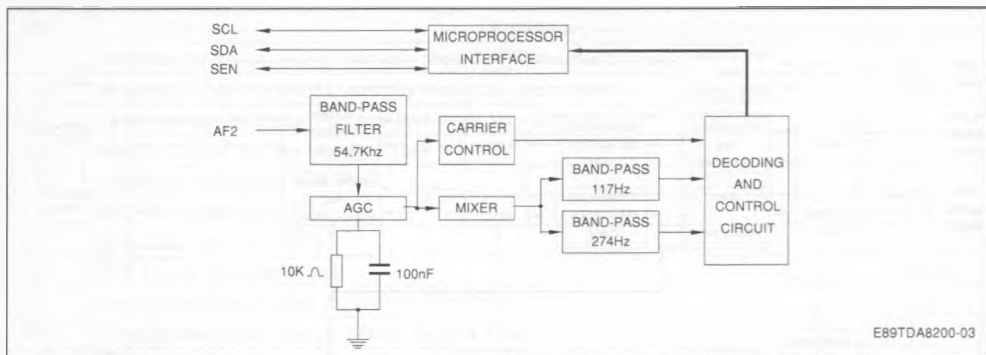
3) PILOT TONE DECODING

The pilot tone decoding section is used to identify the transmission mode (monophonic, stereophonic or bilingual) in B/G standard.

Figure 1 shows the block diagram for the circuit described hereafter :

The decoding signal goes through the pass-band filter at a frequency of 54.7kHz to the AF2 input in order to eliminate undesirable frequencies.

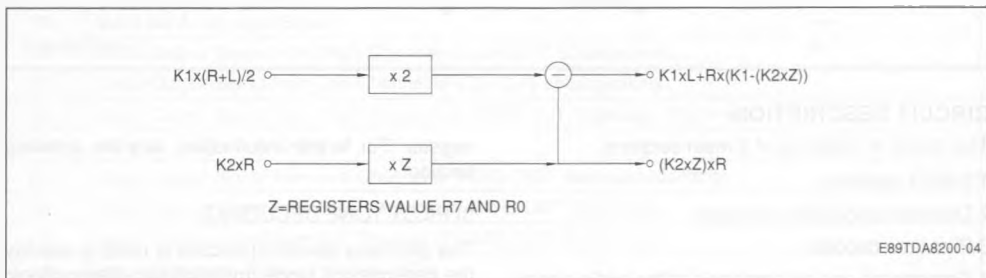
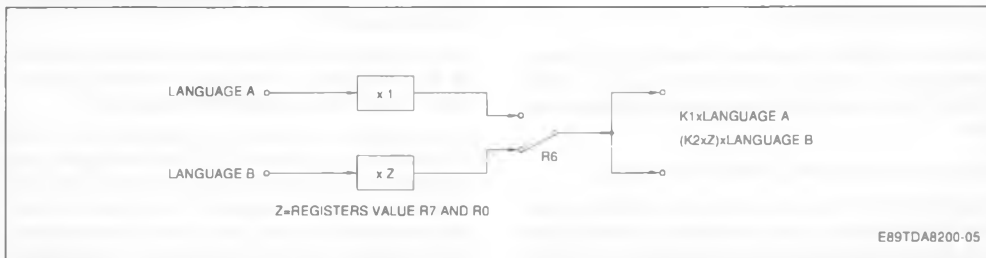
To guarantee a recognition range from 10 to 300mVeff for a carrier at 54.7kHz, the signal goes through an AGC circuit the control velocity of which is based on pin 11's capacity. To be able to establish transmission type, the signal is demodulated by the MIXER and filtered into the 117Hz (stereo transmission) and 274Hz (bilingual transmission) fixed frequencies. If, at the output of only one of the two filters, the signal is present, the "DECODING AND CONTROL CIRCUIT" block verifies if the carrier frequency is correct, using the "CARRIER CONTROL" block, and transmits the information to bits B and S of the RR register.

Figure 1 : Pilot Tone Decoder.**4) DEMATRIXING AND DE-EMPHASIS**

DEMATRIXING. No external component is used for dematrixing the audio signal, but only a programmable attenuator through S-BUS. Attenuation value must be set in calibration phase for the television

set, and loaded every time the TV is turned on by the microprocessor in the R7 and R0 registers of the TDA8200.

Shown below is the dematrixing circuit which differentiates between the STEREO and BILINGUAL signals.

Figure 2 : Stereo Scheme.**Figure 3 : Bilingual Scheme.**

In both circuits, account is taken of the K1 and K2 constants which represent the removal of the AF1 and AF2 signals arriving from the FM demodulator in favor of the real AF1 and AF2 signals transmitted by the generator.

From the output equations can be deduced that, for $K1 = K2 \times Z$, can be obtained. based on precision of Z, the left (L) channel and the correct amplitude of the outputs for the STEREO signal, as well as for the BILINGUAL signals.

As far as the STEREO signal is concerned, a typical value of diaphony between left and right channels of 48dB can be guaranteed by a relationship between K1/K2 of 0.5 to 2, calculated in the following equation :

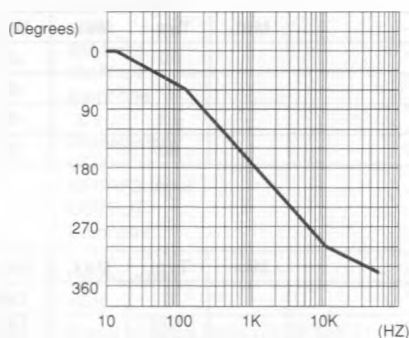
$$CT = 20 \cdot \log\{(K1/K2)/2E-9\}$$

In the case of a BILINGUAL signal, a minimal difference in the signal between languages A and B of 0.05dB can be guaranteed by a relationship between K1 and K2 between 0.5 and 2, calculated with the following equation :

$$K(I-r) = 20 \cdot \log\{(K1/K2)/[(K1/K2) + 2E-9]\}$$

DE-EMPHASIS. The de-emphasis of the AF1 and AF2 signals coming from the FM demodulator are carried out internally in the device without using external components.

Figure 4 : Right Channel Phase.



- **ENLARGED STEREO BASE**, activated by the ES bit, is used for stereophonic signals, and the function is that of making the stereo effect evident, even when the distance between the loudspeakers is reduced. For that purpose, diaphony is introduced in opposition to the right phase and vice versa. The extent of the diaphony to be introduced depends on the distance between the loudspeakers ; in the TDA8200, the diaphony is in the range of 50%.

ABSOLUTE MAXIMUM RATINGS

Symbol	Parameter	Value	Unit
V28	Analog. Supply Voltage	Max 10	V
Vi	Input Voltage (all input)	- 0.3 to Vs + 0.3	V
Ii	Input Current (all input)	Max 5	mA
Io	Output Current (all output)	Max 10	mA
Tstg	Storage Temperature	- 25 to 125	°C

5) AUDIO SIGNAL RECEPTION

Audio signal reception is made up of three essential parts (see block diagram) :

a) **INPUT OUTPUT SELECT.** This circuit deals with shunting the signal coming from the three inputs (VCR, MONO, TV) into the two outputs (VCR, SPEAKER) in the modes described by the 5 bits S0-S4 in the R6 register. Information on possible configuration is detailed in the software section.

b) **STEREOBASE AND PSEUDOSTEREO.** The special effects function only for the loudspeaker output, and are activated by the PS and ES bits of R6. Operation is as follows :

- **PSEUDOSTEREO**, activated by the PS bit, is used with monophonic signals, and consists in the movement of the right channel phase toward the left, on the basis of the frequency as described in figure 4 below :

c) **VOLUME CONTROL.** The earphone output also has a circuit to control and balance volume, carried out by two logarithmic attenuators of 2dB a step with a maximum attenuation value of 60dB. The attenuation depends on the configuration of the 5 bits, CS1-CS5 of R3 for the left channel and the 5 bits, CD1-CD5 of R4 for the right channel ; the values in dB are shown in table 3 shown in the software section.

ELECTRICAL CHARACTERISTICS

Refer to the test circuit, $V_s = 8.5V$, $T_{amb} = 25^{\circ}C$, without special effects, unless otherwise specified.

DC CHARACTERISTICS

Symbol	Parameter	Min.	Typ.	Max.	Unit
V_s	Supply Voltage	8	8.5	9	V
I_s	Supply Current		30		mA
SVR	Supply Voltage Rejection (Frippl = 100Hz)		35		dB

AUDIO INPUT CHARACTERISTICS

Symbol	Parameter	Min.	Typ.	Max.	Unit
V_{in}	Voltage Amplitude on all Inputs		1		Vrms
n	Amplitude Ratio between AF1/AF2	0.5		2	
R_{in}	Resistance on all Audio Inputs	10	40		k Ω

VOLUME CONTROL CHARACTERISTICS**Earphone**

Symbol	Parameter	Min.	Typ.	Max.	Unit
KV	Volume Control Range ; K_{vmax}/K_{vmin}		60		dB
K_{vmin}	Attenuator Resolution/step		2		dB
K_e	Tracking Error $KV = 0$ to 60dB		± 1	± 2	dB
KB	Balance Control Range ; $KV = 0$ dB		60		dB

SPECIAL EFFECT CHARACTERISTICS**Pseudostereo**

Symbol	Parameter	Min.	Typ.	Max.	Unit
Psh	Phase Shifter Response at : 100HZ 1kHz 10kHz		0 180 360		Deg Deg Deg
$K(l-r)$	Amplitude Tracking Error between Left and Right Channels ($KV = 0$ dB)		± 0.5	± 1	dB

Enlarged Stereo Base

Symbol	Parameter	Min.	Typ.	Max.	Unit
$K(l-r)$	Amplitude Tracking Error between Left and Right Channel ($KV = 0$ dB)		± 0.5	± 1	dB

DE-EMPHASIS CHARACTERISTICS (active only in AF1/AF2 input)

Symbol	Parameter	Min.	Typ.	Max.	Unit
t_d	Time Constant	45	50	55	μs

ELECTRICAL CHARACTERISTICS (continued)**AUDIO OUTPUT CHARACTERISTICS** (continued)

Symbol	Parameter	Min.	Typ.	Max.	Unit
d	Total Harmonic Distortion				
	VCR Input : ($V_i = 1V_{rms}$, $F_i = 1kHz$)				
	OUTPUT				
	Earphone		0.2	0.4	%
	Loudspeaker		0.2	0.4	%
	VCR		0.2	0.4	%
	AM Input : ($V_i = 1V_{rms}$, $F_i = 1kHz$)				
	OUTPUT				
	Earphone		0.2	0.4	%
	Loudspeaker		0.2	0.4	%
	VCR		0.2	0.4	%
	AF1/AF2 Input Bilingual : ($V_i = 1V_{rms}$, $F_i = 1kHz$)				
CT	OUTPUT				
	Earphone		0.2	0.4	%
	Loudspeaker		0.2	0.4	%
	VCR		0.2	0.4	%
	AF1/AF2 Input Bilingual : ($V_i = 1V_{rms}$, $F_i = 1kHz$)				
	OUTPUT				
	Earphone		0.2	0.4	%
	Loudspeaker		0.2	0.4	%
	VCR		0.2	0.4	%
	Crosstalk between Left and Right Channels				
	($V_i = 1V_{rms}$, $F_i = 50Hz + 15kHz$, $Bw = 10Hz$)				
	VCR Input Stereo :				
CTb	OUTPUT				
	Earphone		70	80	dB
	Loudspeaker		60	70	dB
	VCR		70	80	dB
	AF1/AF2 Input Stereo :				
	OUTPUT				
	Earphone		70	80	dB
	Loudspeaker		60	70	dB
	VCR		70	80	dB
	Bilingual Crosstalk				
	($V_i = 1V_{rms}$, $F_i = 50Hz + 15kHz$, $Bw = 10Hz$)				
	VCR Input Bilingual :				
CTI	OUTPUT				
	Earphone		75	85	dB
	Loudspeaker		75	85	dB
	VCR		75	85	dB
	AF1/AF2 Input Bilingual :				
	OUTPUT				
	Earphone		70	80	dB
	Loudspeaker		70	80	dB
	VCR		70	80	dB
	Crosstalk between VCR and TV Section				
		70	80		dB

S-BUS CHARACTERISTICS

Symbol	Parameter	Min.	Typ.	Max.	Unit
Vil	Input Voltage Low Level	Vss		0.8	V
Vih	Input Voltage High Level	2		VDD	V
Iil-ih	Input Current			1	μA
Iol	Output Low Current Capability ($V_{ol} = 0.45V$)	5			mA
Ioh	Leakage Output Current ($V_o = 5.25V$, output off, $V_s = 8.5V$)			10	μA

ELECTRICAL CHARACTERISTICS (continued)

PILOT TONE CHARACTERISTICS $V_1 = 50\text{mVrms}$, $m = 0.5$, unless otherwise specified

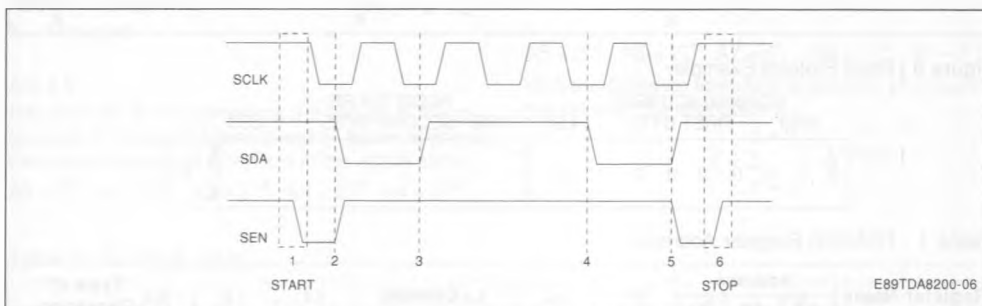
Symbol	Parameter	Min.	Typ.	Max.	Unit
V1	Input Voltage Amplitude (without AM modulation)	10		300	mVrms
d _{fp}	Frequency Range of Carrier to 54687.5Hz		± 1		kHz
d _{fs}	Frequency Range of Stereo Filter (117.4Hz)		± 3		Hz
d _{fb}	Frequency Range of Bilingual Filter (274Hz)		± 6		Hz
dm	Range of AM Modulation Index	40	50	60	%

SOFTWARE INFORMATION

S-BUS DESCRIPTION

Shown below is the timing diagram of the S-BUS protocol :

Figure 5 : S-BUS Timing Diagram.



The START/STOP conditions (points 1 and 6) occur only by a transmission of SEN wire (10 and 01 respectively) while the SCL wire is in high state (1).

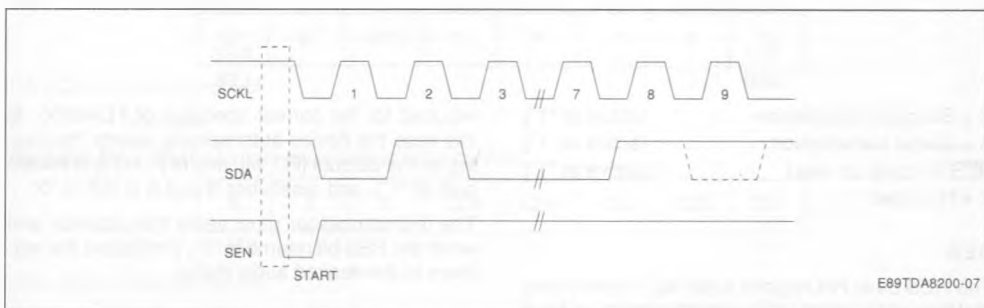
During transmission, the SDA wire can change only when SCL wire is in low state (points 2, 3, 4, 5). After START condition (point 1), the SEN wire must change to high state (point 2) and it remains in this condition for the whole transmission.

At the end of transmission (point 5), the SDA wire

change to high state and in the mean time, the SEN wire must go to low state and then return to high state to generate the STOP condition (point 6).

After the transmission of each byte (composed of 8 bits), there is an ACKNOWLEDGE bit appointed by a high state on SDA wire generated by the transmitter. The device that acknowledges, forces to low state the SDA wire during the time of the acknowledge clock pulse, as described in figure 5 :

Figure 6 : Acknowledge Bit.



In normal conditions, the addressed device must acknowledge after each byte received. If the SDA wire, during the 9th pulse, remains in high state, the master-transmitter can generate the STOP condition to abort the transfer.

Interface between the microprocessor and the TDA8202 occurs through use of the following protocol :

- a start condition (START) see figure 7 and 8
- an address byte, containing the address reserved for the TDA8200 (1000000x) and the bus trans-

mission direction (this information is located in the byte's 8th bit, in which "0" indicates a write, and "1", a read, by the microprocessor). At the end of each byte, the TDA8200 must give the ACKNOWLEDGE signal

- a byte for addressing the registers in write, or the content of register RR in a read
- a third byte containing the information to be written on the register
- a stop condition (STOP)

Figure 7 : Write Protocol Example.

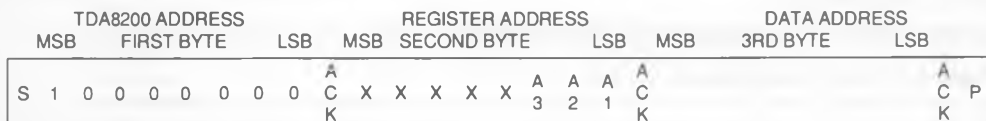


Figure 8 : Read Protocol Example.

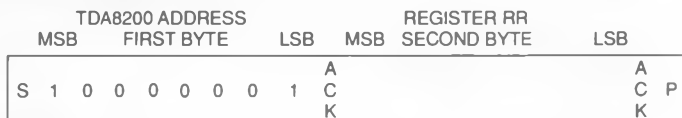


Table 1 : TDA8200 Register Address.

Register Name	Address			Content	Type of Operation
	A3	A2	A1		
R0	0	0	0	Dematrixing and Muting	WRITE ONLY
R3	0	1	1	Left Earphone Volume Channel (60dB)	WRITE ONLY
R4	1	0	0	Right Earphone Volume Channel (60dB)	WRITE ONLY
R6	1	1	0	Configuration Switch and Special Effects	WRITE ONLY
R7	1	1	1	Dematrixing	WRITE ONLY
RR				Transmission Conditions and Power on Reset	READ ONLY

REGISTER CONTENTS

RR register : (Transmission conditions and power on reset)



B = Bilingual transmission

S = Stereo transmission

RES = Power on reset

X = Not used

(active at "1")

(active at "1")

(active at "1")

required for the correct operation of TDA8200. In this case the device automatically inserts the muting on the outputs (R0, ML and MV ; R3 and R4 are both at "1"), and resets bits B and S in RR to "0".

The microprocessor must verify this situation and when the RES bit returns to "0", it initializes the registers to the desired audio status.

RES

The RES bit in RR register is set to "1" every time that the supply voltage falls under the minimum level

B and S

B, and S are the bits reserved to identify the transmission type in B/G standard. If both bits are at "0",

the transmission is MONO, while if only one bit is at "1", the transmission is STEREO (S at "1") or BILINGUAL (B at "1").

Register R0 : (Dematrixing and muting)

X	X	X	TS	ML	MV	A8	A9
MSB							LSB

Register R7 : (Dematrixing)

A0	A1	A2	A3	A4	A5	A6	A7
MSB							LSB

TS = Testing bit (active at "1")
 ML = Loudspeaker muting (active at "1")
 MV = VCR muting (active at "1")
 A0-A9 = Dematrixing (active at "1")
 X = Not used

$$A5 = 2^{-5} \quad A6 = 2^{-6} \quad A7 = 2^{-7} \quad A8 = 2^{-8} \quad A9 = 2^{-9}$$

A0-A9

Bits from A0 to A9 represents the values of the attenuator Z shown in figure 2 and 3. Below is listed the value of every bit when it is in the active state :

$$A0 = 2^0 \quad A1 = 2^{-1} \quad A2 = 2^{-2} \quad A3 = 2^{-3} \quad A4 = 2^{-4}$$

With the following formula it is possible to calculate the value of the attenuator Z :

$$Z = \sum_{n=0}^{n=9} A^n \cdot 2^n$$

Table 2 : Example Table.

	A0	A1	A2	A3	A4	A5	A6	A7	A8	A9	Z
Minimum	0	1	0	0	0	0	0	0	0	0	0.5
Medium	1	0	0	0	0	0	0	0	0	0	1
Maximum	1	1	1	1	1	1	1	1	1	1	2

ML and MV

Bits ML and MV are respectively the mute for the loudspeaker output and VCR output. To have the mute active it is necessary to put the relative bit to

the level "1". At the switch on the device automatically sets both bits to "1" putting the outputs in mute ; so it is necessary to set them to "0" after every switch on.

Register R3 : (Left channel earphone volume control)

X	X	X	CS5	CS4	CS3	CS2	CS1
MSB							LSB

CS1-CS5 = Volume control

X = Not used

Register R4 : (Right channel earphone volume control)

X	X	X	CD5	CD4	CD3	CD2	CD1
MSB							LSB

CD1-CD5 = Volume control

X = Not used

In the table below are listed the values of the attenuation in dB depending on the bits set in registers R3 and R4 :

Table 3 : Earphone Volume Table.

Binary Code		Attenuation dB
MSB	LSB	
XXX00000		0
XXX00001		2
XXX00010		4
XXX00011		6
XXX00100		8
XXX00101		10
XXX00110		12
XXX00111		14
XXX01000		16
XXX01001		18
XXX01010		20
XXX01011		22
XXX01100		24
XXX01101		26
XXX01110		28
XXX01111		30
XXX10000		32
XXX10001		34
XXX10010		36
XXX10011		38
XXX10100		40
XXX10101		42
XXX10110		44
XXX10111		46
XXX11000		48
XXX11001		50
XXX11010		52
XXX11011		54
XXX11100		56
XXX11101		58
XXX11110		60
XXX11111		MUTE

Register R6 : Configuration switch and special effects)

X	S4	PS	ES	S3	S2	S1	S0
MSB				LSB			

S0-S4 = Configuration switch

PS = Pseudostereo

ES = Enlarged stereo base

X = Not used

S0-S4

Bits from S0 to S4 are used to select the desired audio condition of TDA8200. The following table shows all the possible conditions :

Table 4 : Configuration Switch.

Selection Mode S4 S0	AM Input	TV Input		VCR Input		VCR Output		Speaker Output		Earph. Output	
		AF1	AF2	L	R	L	R	L	R	L	R
00000		$\frac{L+R}{2}$	R	L°	R°	L	R	L°	R°	L°	R°
00001		$\frac{L+R}{2}$	R	1°	2°	L	R	2°	2°	1°	1°
00010		$\frac{L+R}{2}$	R	1°	2°	L	R	1°	1°	2°	2°
00011		$\frac{L+R}{2}$	R			M*	M*	L	R	L	R
00100		$\frac{L+R}{2}$	R			L	R	L	R	L	R
00101		1	2			1	2	2	2	1	1
00110		1	2			1	2	1	1	2	2
00111		1	2			1	1	1	1	1	1
01000		1	2	L°	R°	1	2	L°	R°	L°	R°
01001		1	2	1°	2°	1	2	2°	2°	1°	1°
01010		1	2	1°	2°	1	2	1°	1°	2°	2°
01011		1	2	1°	2°	2	2	2	2	1	1
01100		M		L°	R°	M	M	L°	R°	L°	R°
01101		M		1°	2°	M	M	2°	2°	1°	1°
01110		M		1°	2°	M	M	1°	1°	2°	2°
01111		M				M	M	M	M	M	M
11100	AM			L°	R°	AM	AM	L°	R°	L°	R°
11101	AM			1°	2°	AM	AM	2°	2°	1°	1°
11110	AM			1°	2°	AM	AM	1°	1°	2°	2°
11111	AM					AM	AM	AM	AM	AM	AM
10001		$\frac{L+R}{2}$	R	L°	R°	L°	R°	L	R	L	R
10010		M		L°	R°	L°	R°	M	M	M	M
10011		1	2	L°	R°	L°	R°	1	1	2	2
10100		1	2	L°	R°	L°	R°	2	2	1	1
10101	AM			L°	R°	L°	R°	AM	AM	AM	AM
10110				L°	R°	L°	R°	L°	R°	L°	R°
10111				1°	2°	1°	2°	1°	1°	2°	2°
11000				1°	2°	1°	2°	2°	2°	1°	1°

Where :

M = monophonic standard TV signal B/G

M* = reconstructed monophonic signal starting from L and R

L = left standard TV B/G stereo signal

R = right standard TV B/G stereo signal

1 = bilingual B/G TV standard 1st language

2 = bilingual B/G TV standard 2nd language

L° = VCR stereo left signal

R° = VCR stereo right signal

1° = bilingual VCR 1st language

2° = bilingual VCR 2nd language

AM = monophonic standard L signal

PACKAGE MECHANICAL DATA

28 PINS – PLASTIC DIP

