

## LOW COST DIGITALLY CONTROLLED AUDIO PROCESSOR

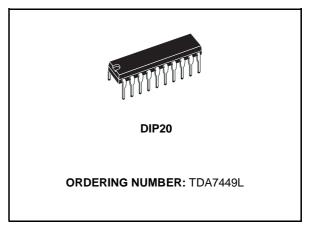
- INPUT MULTIPLEXER
  - 2 STEREO INPUTS
  - SELECTABLE INPUT GAIN FOR OPTIMAL ADAPTATION TO DIFFERENT SOURCES
- ONE STEREO OUTPUT
- VOLUME CONTROL IN 1.0dB STEPS
- TWO SPEAKER ATTENUATORS:
- TWO INDEPENDENT SPEAKER CONTROL IN 1.0dB STEPS FOR BALANCE FACILITY - INDEPENDENT MUTE FUNCTION
- ALL FUNCTION ARE PROGRAMMABLE VIA SERIAL BUS

#### DESCRIPTION

The TDA7449L is a volume control and balance (Left/Right) processor for quality audio applica-tions in TV systems.

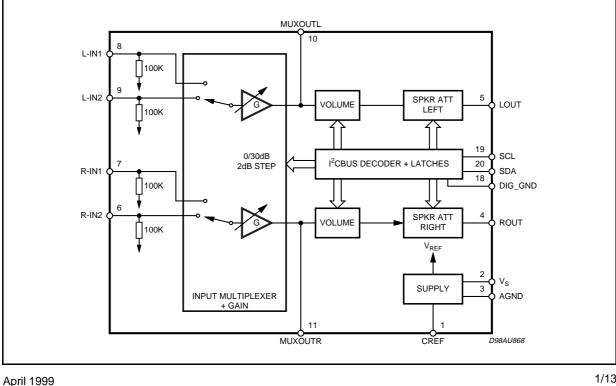
Selectable input gain is provided. Control of all the functions is accomplished by serial bus.

The AC signal setting is obtained by resistor net-



works and switches combined with operational amplifiers.

Thanks to the used BIPOLAR/CMOS Technology, Low Distortion, Low Noise and DC stepping are obtained.

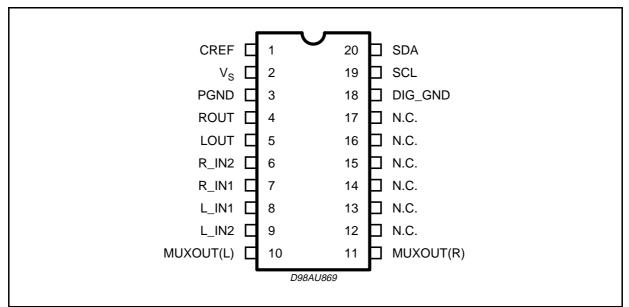


### **BLOCK DIAGRAM**

#### **ABSOLUTE MAXIMUM RATINGS**

Symbol	Parameter	Value	Unit
Vs	Operating Supply Voltage	10.5	V
T <sub>amb</sub>	Operating Ambient Temperature	-10 to 85	°C
T <sub>stg</sub>	Storage Temperature Range	-55 to 150	°C

#### **PIN CONNECTION**



#### THERMAL DATA

Symbol	Parameter	Value	Unit
R <sub>th</sub> j-pin	Thermal Resistance Junction-pins	150	°C/W

#### QUICK REFERENCE DATA

Symbol	Parameter	Min.	Тур.	Max.	Unit
Vs	Supply Voltage	6	9	10.2	V
V <sub>CL</sub>	Max. input signal handling	2			Vrms
THD	Total Harmonic Distortion V = 1Vrms f = 1KHz		0.01	0.1	%
S/N	Signal to Noise Ratio V out = 1Vrms (mode = OFF)		106		dB
Sc	Channel Separation f = 1KHz		90		dB
	Input Gain in (2dB step)	0		30	dB
	Volume Control (1dB step)	-47		0	dB
	Balance Control 1dB step	-79		0	dB
	Mute Attenuation		100		dB

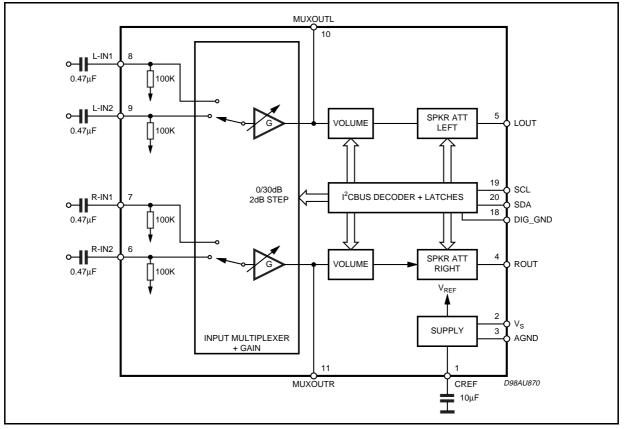
**ELECTRICAL CHARACTERISTICS** (refer to the test circuit  $T_{amb} = 25^{\circ}C$ ,  $V_S = 9V$ ,  $R_L = 10K\Omega$ ,  $R_G = 600\Omega$ , all controls flat (G = 0dB), unless otherwise specified)

Symbol	Parameter	Test Condition	Min.	Тур.	Max.	Unit
SUPPLY						
Vs	Supply Voltage		6	9	10.2	V
ls	Supply Current			7		mA
SVR	Ripple Rejection		60	90		dB
INPUT STA	AGE					
R <sub>IN</sub>	Input Resistance			100		KΩ
V <sub>CL</sub>	Clipping Level	THD = 0.3%	2	2.5		Vrms
S <sub>IN</sub>	Input Separation	The selected input is grounded through a 2.2µ capacitor	80	100		dB
Ginmin	Minimum Input Gain		-1	0	1	dB
G <sub>inman</sub>	Maximum Input Gain			30		dB
G <sub>step</sub>	Step Resolution			2		dB
VOLUME C	CONTROL					
CRANGE	Control Range		45	47	49	dB
A <sub>VMAX</sub>	Max. Attenuation		45	47	49	dB
A <sub>STEP</sub>	Step Resolution		0.5	1	1.5	dB
E <sub>A</sub>	Attenuation Set Error	$A_V = 0$ to -24dB	-1.0	0	1.0	dB
		$A_V = -24 \text{ to } -47 \text{dB}$	-1.5	0	1.5	dB
ΕT	Tracking Error	$A_V = 0$ to -24dB		0	1	dB
		$A_V = -24 \text{ to } -47 \text{dB}$		0	2	dB
V <sub>DC</sub>	DC Step	adjacent attenuation steps from 0dB to $A_V$ max		0 0.5	3	mV mV
A <sub>mute</sub>	Mute Attenuation		80	100		dB
SPEAKER	ATTENUATORS					
CRANGE	Control Range			76		dB
SSTEP	Step Resolution		0.5	1	1.5	dB
EA	Attenuation Set Error	$A_V = 0$ to -20dB	-1.5	0	1.5	dB
		$A_V = -20$ to $-56$ dB	-2	0	2	dB
V <sub>DC</sub>	DC Step	adjacent attenuation steps		0	3	mV
A <sub>mute</sub>	Mute Attenuation		80	100		dB
AUDIO OU	TPUTS					
VCLIP	Clipping Level	d = 0.3%	2.1	2.6		Vrms
RL	Output Load Resistance		2			KΩ
Ro	Output Impedance		10	40	70	Ω
Vdc	DC Voltage Level			3.8		V
GENERAL						
E <sub>NO</sub>	Output Noise	All gains = 0dB; BW = 20Hz to 20KHz flat		5	15	μV
Et	Total Tracking Error	$A_V = 0$ to -24dB		0	1	dB
		$A_V = -24 \text{ to } -47 \text{dB}$		0	2	dB
S/N	Signal to Noise Ratio	All gains 0dB; Vo = 1VRMS ;		106		dB
S <sub>C</sub>	Channel Separation Left/Right		80	100		dB
d	Distortion	Av = 0; Vi = 1Vrms ;		0.01	0.08	%
BUS IN- PUT						

#### ELECTRICAL CHARACTERISTICS (continued.)

Symbol	Parameter	Test Condition	Min.	Тур.	Max.	Unit
BUS INPUT	Г					
VIL	Input Low Voltage				1	V
VIH	Input High Voltage		3			V
lin	Input Current	VIN = 0.4V	-5		5	μA
Vo	Output Voltage SDA Acknowledge	lo = 1.6mA		0.4	0.8	V

#### **TEST CIRCUIT**



#### **APPLICATION SUGGESTIONS**

The first and the last stages are volume control blocks. The control range is 0 to -47dB (mute) for the first one, 0 to -79dB (mute) for the last one. Both of them have 1dB step resolution.

The very high resolution allows the implementation of systems free from any noisy acoustical effect.

The TDA7449L audioprocessor provides 2 bands tones control.

#### CREF

The suggested  $10\mu$ F reference capacitor (CREF) value can be reduced to  $4.7\mu$ F if the application requires faster power ON.



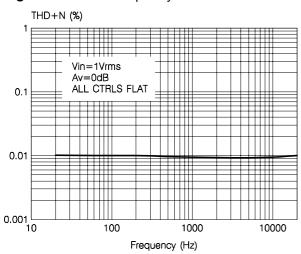




Figure 3: THD vs. RLOAD

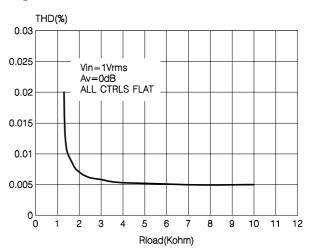
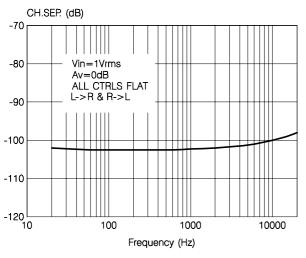


Figure 4: Channel separation vs. frequency



#### **1<sup>2</sup>C BUS INTERFACE**

Data transmission from microprocessor to the TDA7449L and vice versa takes place through the 2 wires  $I^2C$  BUS interface, consisting of the two lines SDA and SCL (pull-up resistors to positive supply voltage must be connected).

#### **Data Validity**

As shown in fig. 3, 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.

#### **Start and Stop Conditions**

As shown in fig.4 a start condition is a HIGH to 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.

#### **Byte Format**

Every byte transferred on the SDA line must contain 8 bits. Each byte must be followed by an ac-

**Figure 3:** Data Validity on the I<sup>2</sup>CBUS

knowledge 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. 5). The peripheral (audio processor) that acknowledges has to pull-down (LOW) the SDA line during this clock pulse.

The audio processor 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.

#### Transmission without Acknowledge

Avoiding to detect the acknowledge of the audio processor, 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.

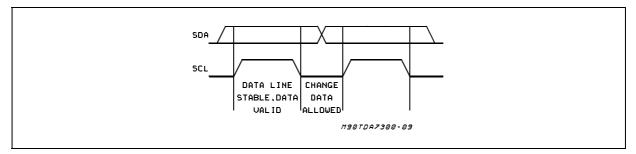
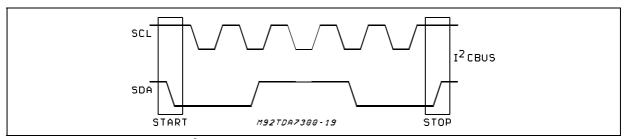
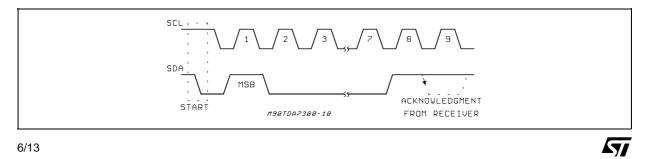


Figure 4: Timing Diagram of I<sup>2</sup>CBUS







#### SOFTWARE SPECIFICATION

Interface Protocol

The interface protocol comprises:

- A start condition (S)
- A chip address byte, containing the TDA7449L

address

- A subaddress bytes
- A sequence of data (N byte + acknowledge)
- A stop condition (P)

CHIP ADDR	ESS	SUBAD	DRESS	DATA 1 to DATA n				
MSB	LSB	MSB	LSB M	MSB	LSB			
S 1 0 0 0 1	0 0 0 ACK	X X X B	DATA ACK	DATA	ACK P			
D96AU420								

ACK = Acknowledge S = Start P = Stop A = AddressB = Auto Increment

#### **EXAMPLES**

**No Incremental Bus** 

The TDA7449L receives a start condition, the cor-

rect chip address, a subaddress with the B = 0 (no incremental bus), N-data (all these data concern the subaddress selected), a stop condition.

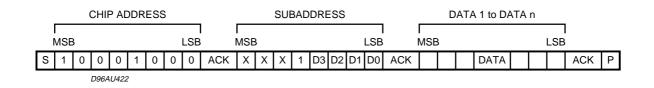
			C	CHI	P AE	DDR	ESS	3			SUBADDRESS					DATA												
	Г MS	в							LSB	I	l Mse	3						LSB		l MSE	3					LSB	1	
S	1	(	)	0	0	1	0	0	0	ACK	Х	Х	Х	0	D3 [	D2	D1	D0	ACK				DATA				ACK	Ρ
			L	D96A	U421																							

#### **Incremental Bus**

The TDA7449L receive a start conditions, the correct chip address, a subaddress with the B = 1 (incremental bus): now it is in a loop condition with an autoincrease of the subaddress whereas

SUBADDRESS from "XXX1000" to "XXX1111" of DATA are ignored.

The DATA 1 concern the subaddress sent, and the DATA 2 concern the subaddress sent plus one in the loop etc, and at the end it receivers the stop condition.



#### POWER ON RESET CONDITION

INPUT SELECTION	IN2
INPUT GAIN	28dB
VOLUME	MUTE
SPEAKER	MUTE

#### **DATA BYTES**

Address = 88 HEX (ADDR:OPEN). FUNCTION SELECTION: First byte (subaddress)

MSB							LSB	SUBADDRESS
D7	D6	D5	D4	D3	D2	D1	D0	SUBADDILESS
Х	Х	Х	В	0	0	0	0	INPUT SELECT
Х	Х	Х	В	0	0	0	1	INPUT GAIN
Х	Х	Х	В	0	0	1	0	VOLUME
Х	Х	Х	В	0	0	1	1	NOT USED
Х	Х	Х	В	0	1	0	0	NOT USED
Х	Х	Х	В	0	1	0	1	NOT USED
Х	Х	Х	В	0	1	1	0	SPEAKER ATTENUATE "R"
Х	Х	Х	В	0	1	1	1	SPEAKER ATTENUATE "L"

B = 1: INCREMENTAL BUS ACTIVE B = 0: NO INCREMENTAL BUS X = DON'T CARE

In Incremental Bus Mode, the three "not used" functions must be addressed in any case. For example to refresh "Volume = 0dB" and Speaker\_R = -40dB", the following bytes must be sent:

SUBADDRESS	XXX10010
VOLUME DATA	X000000
NOT USED 1 DATA	XXXX1111
NOT USED 2 DATA	XXXX1111
NOT USED 3 DATA	XXXX1111
SPEAKER_R DATA	X0000010

#### INPUT SELECTION

MSB							LSB	INPUT MULTIPLEXER
D7	D6	D5	D4	D3	D2	D1	D0	
Х	Х	Х	Х	Х	Х	0	0	NOT ALLOWED
Х	Х	Х	Х	Х	Х	0	1	NOT ALLOWED
Х	Х	Х	Х	Х	Х	1	0	IN2
Х	Х	Х	Х	Х	Х	1	1	IN1

## DATA BYTES (continued)

INPUT GAIN SELECTION

MSB							LSB	INPUT GAIN
D7	D6	D5	D4	D3	D2	D1	D0	2dB STEPS
				0	0	0	0	0dB
				0	0	0	1	2dB
				0	0	1	0	4dB
				0	0	1	1	6dB
				0	1	0	0	8dB
				0	1	0	1	10dB
				0	1	1	0	12dB
				0	1	1	1	14dB
				1	0	0	0	16dB
				1	0	0	1	18dB
				1	0	1	0	20dB
				1	0	1	1	22dB
				1	1	0	0	24dB
				1	1	0	1	26dB
				1	1	1	0	28dB
				1	1	1	1	30dB

GAIN = 0 to 30dB

#### VOLUME SELECTION

MSB							LSB	VOLUME
D7	D6	D5	D4	D3	D2	D1	D0	1dB STEPS
					0	0	0	0dB
					0	0	1	-1dB
					0	1	0	-2dB
					0	1	1	-3dB
					1	0	0	-4dB
					1	0	1	-5dB
					1	1	0	-6dB
					1	1	1	-7dB
	0	0	0	0				0dB
	0	0	0	1				-8dB
	0	0	1	0				-16dB
	0	0	1	1				-24dB
	0	1	0	0				-32dB
	0	1	0	1				-40dB
	Х	1	1	1	Х	Х	Х	MUTE

VOLUME = 0 to 47dB/MUTE

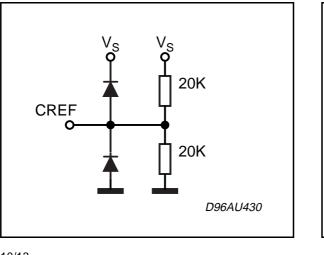
## DATA BYTES (continued)

SPEAKER ATTENUATE SELECTION

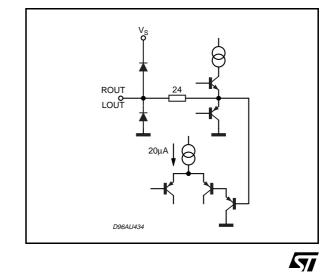
MSB							LSB	SPEAKER ATTENUATION	
D7	D6	D5	D4	D3	D2	D1	D0	1dB	
					0	0	0	0dB	
					0	0	1	-1dB	
					0	1	0	-2dB	
					0	1	1	-3dB	
					1	0	0	-4dB	
					1	0	1	-5dB	
					1	1	0	-6dB	
					1	1	1	-7dB	
	0	0	0	0				0dB	
	0	0	0	1				-8dB	
	0	0	1	0				-16dB	
	0	0	1	1				-24dB	
	0	1	0	0				-32dB	
	0	1	0	1				-40dB	
	0	1	1	0				-48dB	
	0	1	1	1				-56dB	
	1	0	0	0				-64dB	
	1	0	0	1				-72dB	
	1	1	1	1	Х	Х	Х	MUTE	

SPEAKER ATTENUATION = 0 to -79dB/MUTE

**PIN:** 1

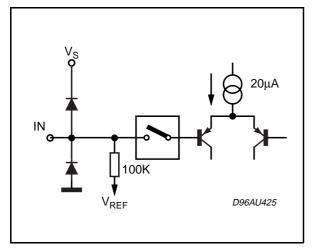




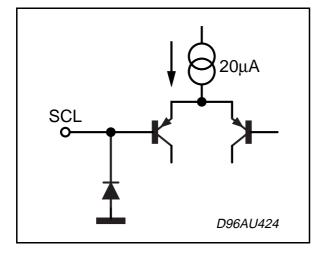


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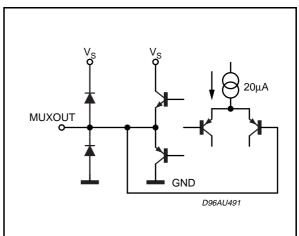
## **PINS:** 6,7,8,9



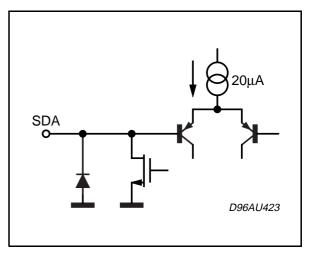




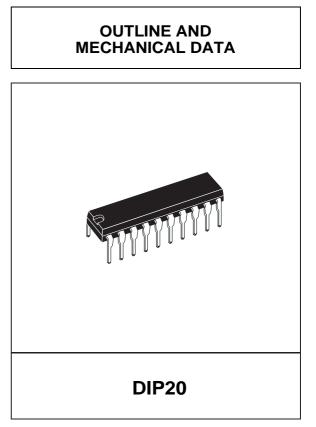


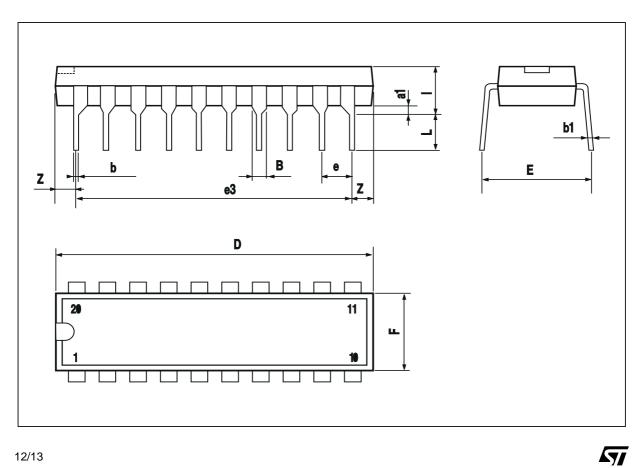






DIM.		mm		inch			
	MIN.	TYP.	MAX.	MIN.	TYP.	MAX.	
a1	0.254			0.010			
В	1.39		1.65	0.055		0.065	
b		0.45			0.018		
b1		0.25			0.010		
D			25.4			1.000	
E		8.5			0.335		
е		2.54			0.100		
e3		22.86			0.900		
F			7.1			0.280	
I			3.93			0.155	
L		3.3			0.130		
Z			1.34			0.053	





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