

TDA7403

BASIC SIGNAL PROCESSOR

- DEVICE INCLUDES AUDIO PROCESSOR, STEREO DECODER AND NOISEBLANKER
- HIGH PERFORMANCE SIGNAL PROCES-SOR WITH BASIC FUNCTIONS
- AM, FM, (MPX) AND CASSETTE INPUTS
- NO EXTERNAL COMPONENTS REQUIRED
- FULLY PROGRAMMABLE VIA I²C BUS
- LOW DISTORTION AND NOISE

DESCRIPTION

The TDA7403 is a high performance signal processor specifically designed for car radio applications focused on the low-end market.

The device includes a complete audioprocessor and a stereo decoder with noiseblanker.

Switched-capacitors design technique allows to obtain all these features without external components or adjustments. Using TDA7403 results is in a very performant low-cost signal processing

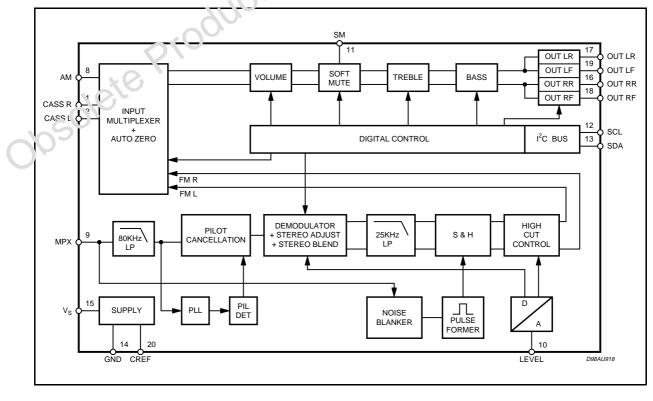


application

The device is fully programmable by I²C bus interface allowing to customize key device parameterc and especially filter characteristics.

The BICMOS process combined with the optimized signal processing assure low noise and low distortion performances.

BLOCK DIAGRAM



November 2002

ABSOLUTE MAXIMUM RATINGS

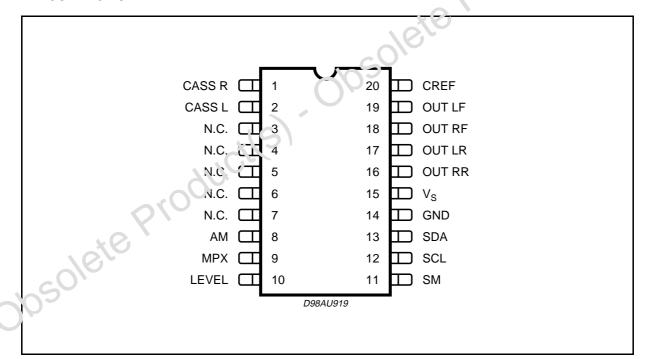
Symbol	Parameter	Value	Unit
Vs	Operating Supply Voltage	10.5	V
Tamb	Operating Ambient Temperature Range	-40 to 80	°C
Tstg	Operating Storage Temperature Range	-55 to 150	°C
	-	•	

SUPPLY

Symbol	Parameter	Test Condition	Min.	Тур.	Max.	Unit
Vs	Supply Voltage		7.5	9	10	V
ls	Supply Current	$V_{\rm S} = 9V$	25	30	35	mA
SVRR	Ripple Rejection @ 1KHz	Audioprocessor (all filters flat)		60		dВ
		Stereodecoder + Audioprocessor		55		dı3
ESD All pins are protected against ESD according to the MIL883 standard. PIN CONNECTION						

ESD

PIN CONNECTION



THERMAL DATA

I	Symbol	Parameter	Value	Unit
	Rth-j pins	Thermal Resistance Junction-pins Max	85	°C/W

PIN DESCRIPTION

N.	Name	Function	Туре
1	CASSR	Cassette Input Right	I
2	CASSL	Cassette Input Left	I
3	n.c.	not connected	
4	n.c.	not connected	
5	n.c.	not connected	
6	n.c.	not connected	
7	n.c.	not connected	
8	AM	AM Input	SI
9	MPX	FM Input (MPX)	
10	LEVEL	Level Input Stereodecoder	I
11	SM	Soft Mute Drive	I
12	SCL	I ² C Clock Line	I/O
13	SDA	I ² C Data Line	I/O
14	GND	Supply Ground	S
15	VS	Supply Voltage	S
16	OUTRR	Right Rear Speaker Output	0
17	OUTLR	Left Rear Speaker Output	0
18	OUTRF	Right Front Spaeaker Output	0
19	OUTLF	Left Front Speaker Output	0
20	CREF	Reference Capachor Pin	S

(1) See input configuration tree and data by tes recification "configuration"

Pin type legenda:

I = Input

- O = Output
- I/O = Input/Ou'.ρυ

S = Sup ily

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AUDIO PROCESSOR PART

Input Multiplexer

- MPX input
- Cassette stereo input
- AM mono

Volume control

- 1dB attenuator
- Max. gain 20dB
- Max. attenuation 79dB

Soft-step gain control

Bass Control

- 2nd order frequency response
- Center frequency programmable in 4(5) steps

- DC gain programmable
- 7 x 2dB steps

Treble Control

- 2nd order frequency response
- Center frequency programmable in 4 steps
- 7 x 2dB steps

Speaker Control

 4 independent speaker controls (1dB steps control range 50dB)

Mute Functions

- Direct mute driven by pin SM
- Digitally controlled softmute with 4 programmable time constants

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ELECTRICAL CHARACTERISTICS (Vs = 9V; Tamb = 25°C; R	$R_L = 10K\Omega$; at cains = 0dB; f = 1KHz;
unless otherwise specified).	C.Y

Symbol	Parameter	Test Condition	Min.	Тур.	Max.	Unit
INPUT SEL	ECTOR					
Rin	Input Resistance	all inputs except Phone	70	100	130	KΩ
Vcl	Clipping Level		2.2	2.6		Vrms
SIN	Input Separation		80	100		dB
GIN MIN	Min. Input Gain		-1	0	1	dB
GIN MAX	Max. Input Gain	51		14		dB
GSTEP	Step Resolution			2		dB
VDC	DC Steps	Adjacent Gain Step		0		mV
		GMIN tO GMAX		1		mV
VOLUME C	CONTROL					
Gмах	Maa Gain			20		dB
Амах	າ 1ລະ: Attenuation			79		dB
ASTEP	Step Resolution			1		dB
E,	Attenuation Set Error	G = -20 to 20dB	-1.25	0	1.25	dB
0		G = -60 to 20dB	-4	0	3	dB
Ет	Tracking Error				2	dB
VDC	DC Steps	Adjacent Attenuation Steps		0.1	3	mV
		From 0dB to G _{MIN}		0.5	5	mV
SOFT MUT	E					
Амите	Mute Attenuation		70	100		dB
TD	Delay Time	T1		0.48		ms
		T2		0.96		ms
		Т3		40.4		ms
		T4		324		ms
VTHIow	Low Threshold for SM Pin ⁽¹⁾				1	V
VTHhigh	High Threshold for SM Pin		2.5			V
Rpu	Internal Pull-up Resistor		70	100	130	KΩ
Vpu	Pull-up Voltage			4.7		V

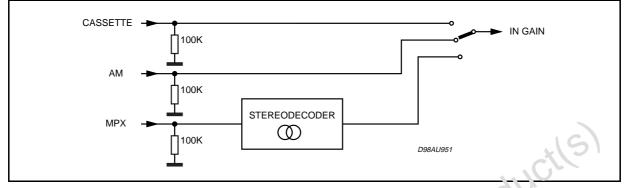
ELECTRICAL CHARACTERISTICS	(continued)
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Symbol	Parameter	Test Condition	Min.	Тур.	Max.	Unit
SOFT STE	P					
Tsw	Switch Time			10		ms
BASS CON						
				11		٩D
	Control Range Step Resolution			±14 2		dB dB
Astep fc	Center Frequency	fc1		2 60		<u>ив</u> Hz
IC IC	Center Frequency	fc2		70		Hz
		fc3		80		Hz
		fc4		100 ⁽²⁾		Hz
QBASS	Quality Factor	Q1		1		
CDAGO		Q2		1.25		9
		Q ₃		1.5	-C - 1	-
		Q4		2	<u>) </u>	
	Bass-Dc-Gain	DC = off		6		dB
		DC = on		4.4		dB
TREBLE C CRANGE	Control Range		· · · · ·	±1.4		D
	Step Resolution			±14 2		dB dB
Astep fc	Center Frequency	fc1		2 10		KHz
IC IC	Center r requeitcy	fc2		12.5		KHz
		fc3		12.5		KHz
		fc4		17.5		KHz
		104		17.0		1112
	ATTENUATORS					
	Control Range			50		dB
ASTEP	Step Resolution			1		dB
AMUTE	Output Mute Atten vation			80	0	dB
EE	Attenuation Set Erro	Adjacent Attenuation Stone		0.1	3	dB
Vdc	DC Steps	Adjacent Attenuation Steps		0.1	10	mV
AUDIO OU		11		1		
VCLIP	ຼດ'ເວລາ່ng Level	d = 0.3%	2.2	2.6		VRMS
R	Cutput Load Resistance		2			KΩ
<u> </u>	Output Load Capacitance				10	nF
KOUT	Output Impedance			30	120	Ω
VDC	DC Voltage Level			3.8		V
GENERAL						
e _{NO}	Output Noise	BW = 20 Hz to 20 KHz output muted		5		μV
		BW = 20 Hz to 20 KHz all gain = 0dB		10		μV
S/N	Signal to Noise Ratio	all gain = 0dB flat; Vo = 2V _{RMS}		106		dB
		bass treble at 12dB; Vo = 2.6V _{RMS}		96		dB
d	Distortion	VIN = 1VRMS; all stages 0dB		0.002		%
		VIN = 1VRMS; Bass & Treble = 12dB		0.05		%
Sc	Channel separation Left/Right		80	100		dB
Ет	Total Tracking Error	Av = 0 to -20dB		0	1	dB
		Av = -20 to -60dB		0	2	dB

SM pin is active low (mute condition)
See description of Audioprocessor Part - Bass & Treble filter characteristics programming

DESCRIPTION OF THE AUDIOPROCESSOR

Figure 1. Input stages



Input stages

The input circuits are the same as in preceeding ST audioprocessors with exception of the CD inputs (see figure 1).

The typical input impedance is $100k\Omega$.

AutoZero

In order to reduce the number of pins there is no AC coupling between the In-Gain and the following stage, so that any offset generated by or bofore the In-Gain stage would be transferred or even amplified to the output.

To avoid that effect a special offset cancellation stage called AutoZero is implemented.

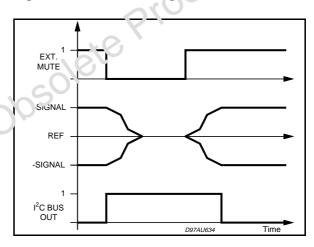
To avoid audible clicks the audioprocessor is muted before the loudness stage during this time. In some cases, for example if the μ P is executing a refresh cycle of the l²C bus programming, it is not useful to start a new AutoZero action because no new source is selected and an undesired mute would ecocer at the outputs. For such applications the TDA7403 could be switched in the "Auto Zero Remain" mode (Bit 6 of the subaddress Lvte). If this bit is set to high, the DATABYTE 0 could be loaded without invoking the AutoZero and the old adjustment value remains.

Softmute

The digitally controlled softmute stage allows muting/demuting the signal with a I^2C bus programmable slope. The mute process can either be activated by the softmute pin or by the I^2C bus. The slope is realized in a special S shaped curve to mute slow in the critical regions (see figure 2).

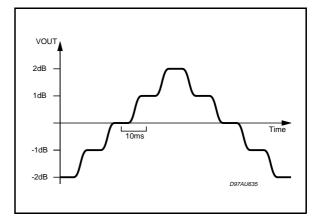
For timing purposes the Bit 3 of the I^2C bus output register is set to 1 from the start of muting until the end of demuting.

Figure 2. Softmute Timing



Note: Please notice that a started Mute action is always terminated and could not be interrupted by a change of the mute signal.

Figure 3. Soft Step Timing



Note: For steps more than 1dB the softstep mode should be deactivated because it could generate a 1dB error during the blend-time

Softstep Volume

When volume level is changed often an audible click appears at the output. The root cause of those clicks could be either a DC offset before the volume stage or the sudden change of the envelope of the audio signal. With the Softstep feature both kinds of clicks could be reduced to a minimum and are no more audible (see figure 3).

Bass

There are three parameters programmable in the bass stage (see figs 4, 5, 6, 7):

- Attenuation
- Center Frequency (60, 70, 80 and 100Hz)
- Quality Factors (1, 1.25, 1.5 and 2)

DC Mode

In this mode the DC gain is increased by 4.4dB. In addition the programmed center frequency and quality factor is decreased by 25% which can be used to reach alternative center frequencies or quality factors.

Treble

There are two parameters programmable in the treble stage (see figs 8, 9):

- Attenuation
- Center Frequency (10, 12.5, 15 and 17.5kr z).

Speaker Attenuator

Due to practical aspects the steps in the speaker attenuators are not linear over the full range. At to 10de attenuations more than 2403 the steps increase from 1.5dB to 10dE (p.e.se see data byte specifi-

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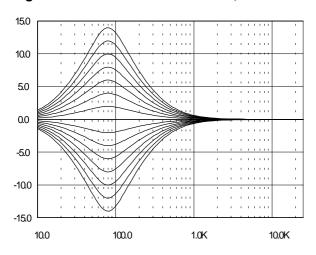
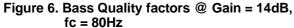
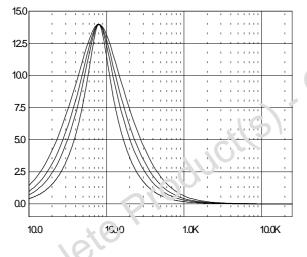


Figure 4. Bass Control @ fc = 80Hz, Q = 1





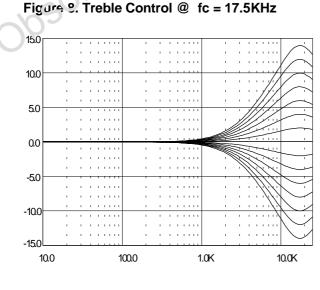


Figure 5. Bass Center @ Gain = 14dB, Q = 1

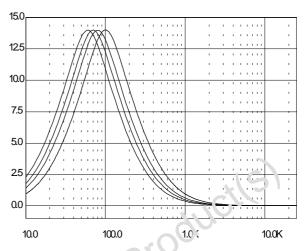
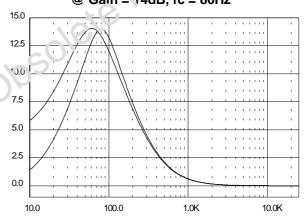
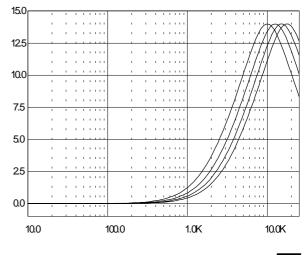


Figure 7. Bass no: mail and DC Mode @ Gain = 14dB, fc = 80Hz



Note: In general the center frequency, Q and DC-mode can be set independently. The exception from this rule is the mode (5/xx1111xx) where the center frequency is set to 150Hz instead of 100Hz.

Figure 9. Treble Center Frequencies @ Gain = 14dB



STEREODECODER PART

- No external components necessary
- PLL with adjustment free fully integrated VCO
- Automatic pilot dependent MONO/STEREO switching
- Very high suppression of intermodulation and interference
- Programmable Roll-Off compensation

- Dedicated RDS Softmute
- Highcut and Stereoblend characterisctics programmable in a wide range
- Internal Noiseblanker with threshold controls
- Multipath detector with programmable internal/external influence
- I²C bus control of all necessary functions

ELECTRICAL CHARACTERISTICS (V_S = 9V; deemphasis time constant = 50μ s, V_{MPX} = 500mV, 75KHz deviation, f = 1KHz. GI = 6dB, Tamb = 25° C; unless otherwise specified).

Symbol	Parameter	Test Condition	Min.	Тур.	Max	Jr it
Vin	MPX Input Level	Input Gain = 3.5dB		0.5	<u>1.2.</u>	VRMS
Rin	Input Resistance			160		KΩ
Gmin	Minimum Input Gain			<u> </u>		dB
Gmax	Max Input Gain		<u>کم ا</u>	11		dB
GSTEP	Step Resolution			2.5		dB
SVRR	Supply Voltage Ripple Rejection	Vripple = 100mv, f = 1khz		60		dB
α	Max Channel Separation		·	50		dB
THD	Total Harmonic Distortion			0.05		%
$\frac{S+N}{N}$	Signal plus Noise to Noise Ratio	S = 2V _{rms}		86		dB
MONO/STE	EREO SWITCH					
VPTHST1	Pilot Threshold Voltage	fo. Stereo, PTH = 1		15		mV
VPTHST0	Pilot Threshold Voltage	for Stereo, PTH = 0		25		mV
VPTHMO1	Pilot Threshold Voltage	for Mono, PTH = 1		12		mV
VPTHMO0	Pilot Threshold Valle go	for Stereo, PTH = 0		19		mV
PLL						
∆f/f	Capture Range		0.5			%
DEEMPHA	SIS ar.d HIGHCUT					
THC50	Deemphasis Time Constant	Bit = 7, Subadr. 10 = 0 V _{LEVEL} >> V _{HCH}		50		μs
1H:275	Deemphasis Time Constant	Bit = 7, Subadr. 10 = 1 VLEVEL >> V _{HCH}		75		μs
THC50	Highcut Time Constant	Bit = 7, Subadr. 10 = 0 VLEVEL >> VHCL		150		μs
THC75	Highcut Time Constant	Bit = 7, Subadr. 10 = 1 VLEVEL >> V _{HCL}		225		μs
STEREOBL	END and HIGHCUT-CONTRO	DL				
REF5V	Internal Reference Voltage			5		V
TCREF5V	Temperature Coefficient			3300		ppm
LGmin	Min. LEVEL Gain			0		dB
LGmax	Max. LEVEL Gain			10		dB
LGstep	LEVEL Gain Step Resolution			0.67		dB
VSBLmin	Min.Voltage for Mono			33		%REF5V
VSBLmax	Max. Voltage for Mono			58		%REF5V
VSBLstep	Step Resolution			8.4		%REF5V

ELECTRICAL CHARACTERISTICS (continued)

Symbol	Parameter	Test Condition	Min.	Тур.	Max.	Unit
STEREOBI	_END and HIGHCUT CONTR	OL				
VHCHmin	Min.Voltage for NO Highcut			42		%REF5V
VHCH _{max}	Max. Voltage for NO Highcut			66		%REF5V
VHCH _{step}	Step Resolution			8.4		%REF5V
VHCLmin	Min. Voltage for FULL High cut			17		%VHCH
VHCLmax	Max. Voltage for FULL High cut			33		%VHCH
Carrier and	harmonic suppression at the	output				
α19	Pilot Signal	f = 19KHz		50		arz
α38	Subcarrier	f = 38KHz		75		JB
α57	Subcarrier	f = 57KHz		62	\Box	dB
α76	Subcarrier	f = 76KHz		90		dB
Intermodula	ation (Note1)			0		
α2	Pilot Signal	f _{mod} = 10KHz; f _{spur} = 1KHz;		65		dB
α3		f _{mod} = 13KHz; f _{spur} = 1KHz;		75		dB
Traffic Radi	io (Note 2)	101				
α57	Signal	f = 57KHz		70		dB
SCA - Sub	sidiary Communications Auth	orization (Note 3)				
α67	Signal	f = 67KH z		75		dB
ACI - Adjac	ent Channel Interference (No	te 4)				
α114	Signal	f = 114KHz		95		dB
α190	Signal	f = 190KHz		84		dB
Notes to the ch	aracteristics:					

1. Intermodulation Suppression: measured with: 91% pilot signal; fm = 10kHz or 13kHz.

2. Traffic Radio (V.F.) Suppression. ...easured with: 91% stereo signal; 9% pilot signal; fm=1kHz; 5% subcarrier (f = 57kHz, fm = 23Hz AM, m = 60.1

3. SCA (Subsidiary Communications Authorization) measured with: 81% mono signal; 9% pilot signal; fm = 1kHz; 10%SCA - subcarrier (fs = 67kHz, immodulated).

4. ACI (/ ឃុំ ac an Channel Interference) measured with: 90% mono signal; 9% pilot signal; fm =1kHz; 1% spurious signal (fc = 10 .Hz or 186kHz, unmodulated).

NOISE BLANKER PART

- internal 2nd order 140kHz high pass filter
- programmable trigger threshold

- additional circuits for trigger adjustment (deviation, field-strenght)
- very low offset current during hold time
- four selectable pulse suppression times

Symbol	Parameter	Test Condition		Min.	Тур.	Max.	Unit
Vtr	Trigger Threshold ^{0) 1)}	meas. with VPEAK = 0.9V	NBT = 111		30		тVор
			NBT = 110		35		тVор
			NBT = 101		40		n.Vop
			NBT = 100		45		T IV OP
			NBT = 011		50		mVop
			NBT = 010		55		тVор
			NBT = 001		60		тVор
			NBT = 000		35		тVор
VTRNOISE	Noise Controlled Trigger	meas. with VPEAK = 1.5V	NCT = 00		260		тVор
	Threshold ²⁾		NCT = 01		220		тVор
			NCT _ 1.7		180		тVор
			NCT - 11		140		тVор
Vrect	Rectifier Voltage	V _{MPX} = 0mV	.0.		0.9		V
		VMPX = 50mV; f - 1CJK'H2	2		1.7		V
		V _{MPX} = 100mV f = 5JKH	lz		2.5		V
Vrect dev	deviation dependent	means. with	OVD = 11		0.9(off)		Vop
	rectifier Voltage 3)	$V_{MPX} = 300 \text{mV}$	OVD = 10		1.2		Vop
		(751 Hz a zv.)	OVD = 01		2.0		Vop
			OVD = 00		2.8		Vop
VRECT FS	Fieldstrength Cont. o. ea	means. with	FSC = 11		0.9(off)		V
	Rectifier Voltage	$V_{MPX} = 0mV$	FSC = 10		1.3		V
		VLEVEL << VSBL (fully mono)	FSC = 01		1.8		V
			FSC = 00		2.3		V

ELECTRICAL CHARACTERISTICS (continued)

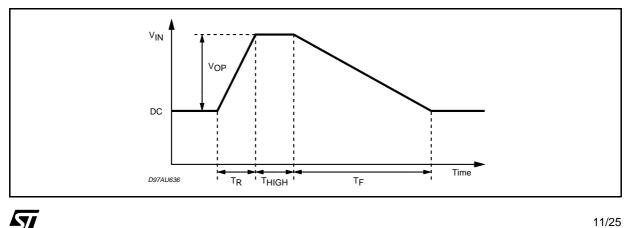
0) All thresholds a terms are sured using a pulse with $T_R = 2 \mu s$, $T_{HIGH} = 2 \mu s$ and $T_F = 10 \mu s$.

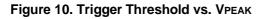
1) NBT represents the Noiseblanker-Byte bits D2; D0 for the noise blanker trigger threshold

2) Nor coresents the Noiseblanker-Byte bit pair D4,D3 for the noise controlled trigger adjustment

.) C /D represents the Noiseblanker-Byte bit pair D7,D6 for the over deviation detector

4) FSC represents the Fieldstrength-Byte bit pair D1,D0 for the fieldstrength control





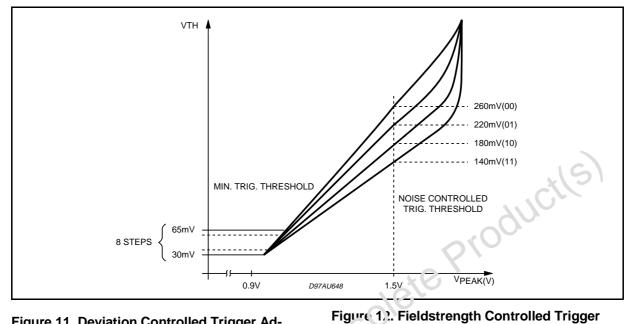


Figure 11. Deviation Controlled Trigger Adjustment

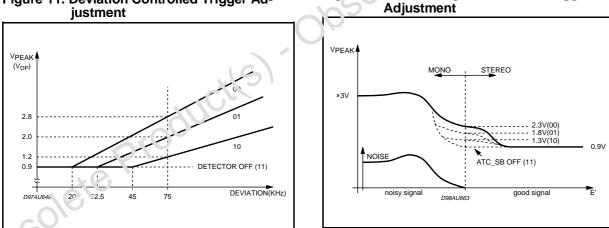
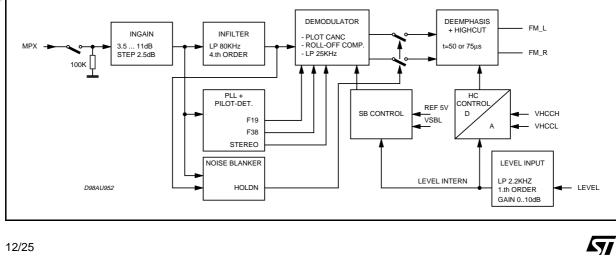


Figure 13. Block diagram of the stereo decoder



DESCRIPTION OF STEREODECODER

The stereodecoder part of the TDA7403 (see Fig. 16) contains all functions necessary to demodulate the MPX signal like pilot tone dependent MONO/STEREO switching as well as "stereoblend" and "highcut" functions.

Adaptations like programmable input gain, roll-off compensation, selectable deemphasis time constant and a programmable fieldstrength input allow to use different IF devices.

Stereodecoder Mute

The TDA7403 has a fast and easy to control RDS mute function which is a combination of the audioprocessor softmute and the high-ohmic mute of the stereodecoder. If the stereodecoder is selected and a softmute command is sent (or activated through the SM pin) the stereodecoder will be set automatically to the high-ohmic mute condition after the audio signal has been softmuted.

Hence a checking of alternate frequencies could be performed. To release the system from the mute condition simply the unmute command must be sent: the stereodecoder is unmuted immediately and the audioprocessor is softly unmuted. Fig. 14 shows the output signal Vo as well as the internal stereodecoder mute signal. The influence of Softmute on the stereodecoder mute can be switched off by setting bit 3 of the Softmute byte to "0". A stereodecoder mute command (bit 0, stereodecoder byte set to "1") will set the stereodecoder in any case independently to the high-ohmic mute state.

If any other source than the stereodecoder is selected the decoder remains muted and the MPX pin is connected to Vref to avoid any discharge of the coup'ing capacitor through leakage currents.

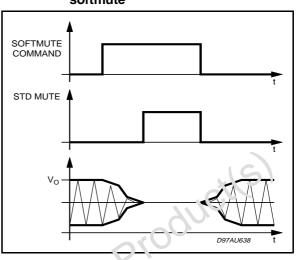
h:put Stages

The Ingain stage allows to adjust the MPX signal to a magnitude of about 1Vrms internally which is the recommended value. The 4.th order input filter has a corner frequency of 80kHz and is used to attenuate spikes and noise and acts as an antialiasing filter for the following switch capacitor filters.

Demodulator

In the demodulator block the left and the right channel are separated from the MPX signal. In this stage also the 19 kHz pilot tone is cancelled. For reaching a high channel separation the TDA7403 offers an I2C bus programmable roll-off adjustment which is able to compensate the low-





pass behaviour of the tuner section. If the tuner attenuation a, 33kHz is in a range from 20.2% to 31% the T237403 needs no external network before the 3325 pin. Within this range an adjustment to cotain at least 40dB channel separation is posible.

The bits for this adjustment are located together with the fieldstrength adjustment in one byte. This gives the possibility to perform an optimization step during the production of the carradio where the channel separation and the fieldstrength control are trimmed.

Deemphasis and Highcut

The lowpass filter for the deemphasis allows to choose between a time constant of $50\mu s$ and $75\mu s$ (bit D7, Stereodecoder byte).

The highcut control range will be in both cases $t_{HC} = 2 \cdot t_{Deemp}$. Inside the highcut control range (between VHCH and VHCL) the LEVEL signal is converted into a 5 bit word which controls the lowpass time constant between t_{Deemp} ... $3 \cdot t_{Deemp}$. There by the resolution will remain always 5 bits independently of the absolute voltage range between the VHCH and VHCL values.

The highcut function can be switched off by I2C bus (bit D7, Fieldstrength byte set to "0").

PLL and Pilot Tone Detector

The PLL has the task to lock on the 19kHz pilotone during a stereo transmission to allow a correct demodulation. The included detector enables the demodulation if the pilot tone reaches the selected pilottone threshold VPTHST. Two different thresholds are available. The detector output (signal STEREO, see block diagram) can be checked

Figure 14. Signals during stereodecoder's softmute

by reading the status byte of the TDA7403 via I2C bus.

Fieldstrength Control

The fieldstrength input is used to control the high cut and the stereoblend function. In addition the signal can be also used to control the noiseblanker thresholds.

LEVEL Input and Gain

To suppress undesired high frequency modulation on the highcut and stereoblend function the LEVEL signal is lowpass filtered firstly. The filter is a combination of a 1st order RC lowpass at 53kHz (working as anti-aliasing filter) and a 1storder switched capacitor lowpass at 2.2kHz. The second stage is a programmable gain stage to adapt the LEVEL signal internally to different IF.

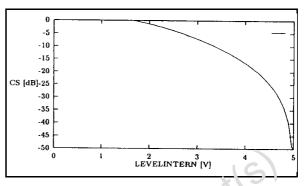
The gain is widely programmable in 16 steps from 0dB to 10dB (step = 0.67dB). These 4 bits are located together with the Roll-Off bits in the "Stereodecoder Adjustment" byte to simplify a possible adaptation during the production of the carradio.

Stereoblend Control

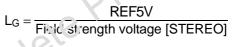
The stereoblend control block converts the internal LEVEL voltage (LEVEL INTERN) into an demodulator compatible analog signal which is used to control the channel separation between 0dB and the maximum separation internally this control range has a fixed upper limit which is the internal reference voltage REF5V. The lower limit can be programmed to be 33%, 42%, 50% or 58% of REF5V (see 19. 16).

To adjust the errie nal LEVEL voltage to the internal range two values must be defined: the LEVEL





gain L_G and VSBL. To adjust the voluce where the full channel separation is reached (VST) the LEVEL gain L_G has to be defined. The following equation can be used to estimate the gain:



The gain can be programmed through 4 bits in the "Scereodecoder-Adjustment" byte.

The MONO voltage VMO (0dB channel separation) can be choosen selecting 33, 42, 50 or 58% of REF5V.

All necessary internal reference voltages like REF5V are derived from a bandgap circuit. Therefore they have a temperature coefficient near zero. This is useful if the fieldstrength signal is also temperature compensated.

But most IF devices apply a LEVEL voltage with a TC of 3300ppm. The TDA7403 offers this TC for the reference voltages, too. The TC is selectable with bit D7 of the "stereodecoder adjustment" byte.

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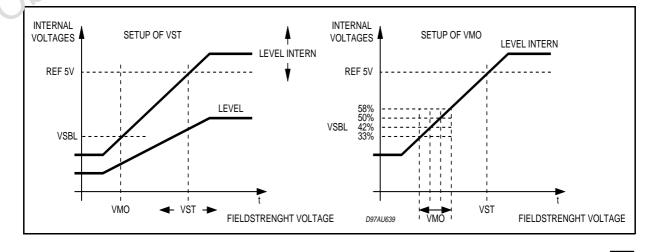
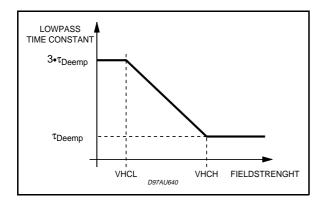


Figure 16. Relation between internal and external LEVEL voltage and setup of Stereoblend

Highcut Control

The highcut control setup is similar to the stereoblend control setup : the starting point VHCH can be set with 2 bits to be 42, 50, 58 or 66% of REF5V whereas the range can be set to be 17 or 33% of VHCH (see fig. 17).

Figure 17. Highcut characteristics



FUNCTIONAL DESCRIPTION OF THE NOISE-BLANKER

In the automotive environment the MPX signal is disturbed by spikes produced by the ignition and for example the wiper motor. The aim of the noiseblanker part is to cancel the autible influence of the spikes. Therefore the output of the stereodecoder is held at the actual voltage for 40μ s.

In a first stage the spikes must be detected but to avoid a wrong trogeting on high frequency (white) noise a complex trigger control is implemented. Behind the triggerstage a pulse former generates the "blanking" pulse. To avoid any crosstall to the signalpath the noiseblanker is

Figure 18. Block diagram of the noiseblanker

supplied by his own biasing circuit.

Trigger Path

The incoming MPX signal is highpass filtered, amplified and rectified. This second order highpass-filter has a corner frequency of 140kHz. The rectified signal, RECT, is lowpass filtered to generate a signal called PEAK. Also noise with a frequency 140kHz increases the PEAK voltage. The PEAK voltage is fed to a threshold generator, which adds to the PEAK voltage a DC dependent threshold VTH. Both signals, RECT and PEAK+VTH are fed to a comparator which triggers a re-triggerable monoflop. The monot.op's output activates the sample-and-ho'd circuits in the signalpath for 40µs.

The block diagram of the noise lanker is given in fig.18.

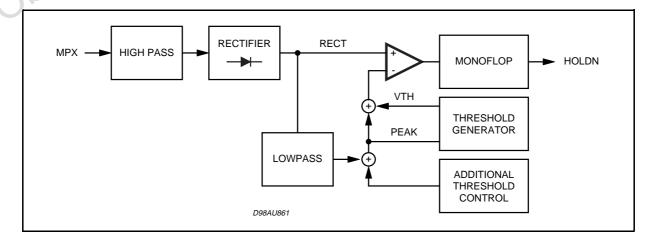
Automatic Noise Controlled Threshold Adjustment (ATC)

There are mainly two independent possibilities for programming the trigger threshold:

- a the low threshold in 8 steps (bits D0 to D2 of the noiseblanker byte)
- b the noise adjusted threshold in 4 steps (bits D3 and D4 of the noiseblanker byte, see fig. 13).

The low threshold is active in combination with a good MPX signal without any noise; the PEAK voltage is less than 1V. The sensitivity in this operation is high.

If the MPX signal is noisy the PEAK voltage increases due to the higher noise, which is also rectified. With increasing of the PEAK voltage the trigger threshold increases, too. This particular gain is programmable in 4 steps (see fig. 10).



Automatic Threshold Control

Besides the noise controlled threshold adjustment there is an additional possibility for influencing the trigger threshold. It is depending on the stereoblend control.

The point where the MPX signal starts to become noisy is fixed by the RF part. Therefore also the starting point of the normal noise-controlled trigger adjustment is fixed (fig. 15). In some cases the behaviour of the noiseblanker can be improved by increasing the threshold even in a region of higher fieldstrength. Sometimes a wrong triggering occures for the MPX signal often shows distortion in this range which can be avoided even if using a low threshold.

Because of the overlap of this range and the range of the stereo/mono transition it can be controlled by stereoblend. This threshold increase is programmable in 3 steps or switched off with bits D0 and D1 of the fieldstrength control byte.

Over Deviation Detector

If the system is tuned to stations with a high deviation the noiseblanker can trigger on the higher frequencies of the modulation. To avoid this wrong behaviour, which causes noise in the output signal, the noiseblanker offers a deviation dependent threshold adjustment.

By rectifying the MPX signal a further signal representing the actual deviation is obtained. It is used to increase the PEAK voltage. Offset and gain of this circuit are programmable in 3 steps with the bits D6 and D7 of the stereodecoder byte (the first step turns off the detector, see fig. 15).

TEST MODE

During the test mode which can be activated by setting bit D0 of the testing k yte and bit D5 of the subaddress byte to "1" se eral internal signals are available at the CASSR pin. During this mode the input resistance of 100kOhm is disconnected from the pin. The internal signals available are shown in the software specification.

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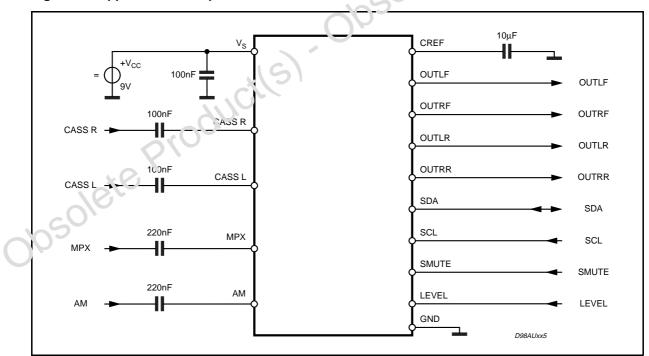


Figure 19. Application Example.

I²C BUS INTERFACE DESCRIPTION

Interface Protocol

The interface protocol comprises:

-a start condition (S)

/ write transmission)

-a subaddress byte

-a sequence of data (N-bytes + acknowledge) -a stop condition (P)

-a chip address byte (the LSB bit determines read

	CHIP ADDRESS						SUBADDRESS						DATA 1 to DATA n												
l MSB	3						LSB		l MSI	3						LSB		MSE	3				LSB		
S 1	0	0	0	1	1	0	R/W	ACK	Х	AZ	Т	Ι	A3	A2	A1	A0	ACK				DATA			ACK	Ρ
		D97A	NU627	,																					

S = Start

ACK = Acknowledge

AZ = AutoZero-Remain

T = Testing

I = Autoincrement

P = Stop

MAX CLOCK SPEED 500kbits/s

The transmitted data is automatically updated after each ACK.

Transmission can be repeated without new chip address.

Auto i	increment
--------	-----------

If bit I in the subaddress byte is set to "1", the autoincrement of the subaddress is enabled.

TRANSMITTED DATA (sand mode)

MSB	. 0	K				LSB
Х	XXX	Х	ST	SM	Х	Х

SM = Soft mute activated

ST - Stereo

✓ = Not Used

MSB			.C			LSB	FUNCTION
Х	AZ	Т	 A:,	A2	A1	A0	
			0	0	0	0	Not allowed
			0	0	0	1	Auto-Zero
			0	0	1	0	Volume
		\sim	0	0	1	1	Softmute
		0	0	1	0	0	Bass / Treble Attenuator
			0	1	0	1	Bass / Treble Configuration
6			0	1	1	0	Speaker attenuator LF
G	\mathcal{O}^{+}		0	1	1	1	Speaker attenuator LR
			1	0	0	0	Speaker attenuator RF
			1	0	0	1	Speaker attenuator RR / Blanktime adjust
			1	0	1	0	Stereodecoder
			1	0	1	1	Noiseblanker
			1	1	0	0	Fieldstrength Control
			1	1	0	1	Configuration
			1	1	1	0	Stereodecoder Adjustment
			1	1	1	1	Testing

T = Testmode

I = Autoincrement

AZ = Auto Zero Remain

X = not used

DATA BYTE SPECIFICATION Input Selector

MSB							LSB	FUNCTION
D7	D6	D5	D4	D3	D2	D1	D0	
					0 0 0 1 1 1	0 0 1 1 0 0 1 1	0 1 0 1 0 1 0 1	Source Selector don't use Cassette don't use AM Stereo Decoder don't use Mute don't use
				Х				Don't Care
			1 0 0 1		0 0 1 1	1 1 0 0	1 1 0 0	AM/FM Mode AM mono AM stereo AM through Stereo, 'De co der FM- Stereo/De co der
0 0 : 1 1	0 0 1 1	0 1 0 1						In-Gain 14dB 12dF 2 cB 0 dB

For example to select the CD input in quasi-differential mode with g_{ϵ} in of 3dB the Data Byte is: 0/01111000

Autozero

				16				
				5	1			
		~	3					
Autoze	ro	<u>KO</u>						
	X					LSB		LOUDNESS
MSB								
MSB D7	D6 25	D4	D3	D2	D1	D0		
	<u>D6 ン5</u> 1	D4 1	D3 0	D2 0	D1 0	D0 0	setting fix	
	D6 5 1	D4 1					setting fix	

Mute, Beep and Mixing

MSB							LSB	MUTE/BEEP/MIXING
D7	D6	D5	D4	D3	D2	D1	D0	
				0 1	0 0 1 1	0 1 0 1	0 1	Mute Enable Softmute Disable Softmute Mute time =0.48 ms Mute time =0.96 ms Mute time =40.4 ms Mute time =324 ms Stereo Decoder Softmute Influence = off Stereo Decoder Softmute Influence = on
1	1	1	1					Must be "1"

Volume

ote: for more information to the Stereodecoder-Softmute-Influence please refer to the stereodecoder description										
MSB							LSB	ATTENUATION		
D7	D6	D5	D4	D3	D2	D1	D0	0.		
	0 0 0 0 0 0 0 1	0 0 0 0 1 1 1	00.000.100.00	00:11100:11	00	0 0 0 1 : 1 0 0 : 1 1	0 1 0 1 0 1 0 1 0 1 0 1	Gain/Attenuation +32dB +31dB : +20dB +19dB +19dB +18dB : +1dB 0dB - 1dB : -78dB -79dB		
0 (96							Softstep Softstep Volume = off Softstep Volume = on		

Note: It is not recommended to use a gain more than 20dB for system performance reason. In general, the max. gain should be limited by software to the maximum value, which is needed for the system.

Bass & Treble Attenuation

MSB							LSB	BASS & TREBLE ATTENUATION
D7	D6	D5	D4	D3	D2	D1	D0	
				0 0 1 1 1	0 0 1 1 1 : 0 0	0 0 1 1 1 1 : 0 0	0 1 : 0 1 1 0 : 1 0	Treble Steps -14dB -12dB : -2dB 0dB 0dB +2dB : +12dB +14dB
0 0 1 1 1 1	0 0 1 1 1 1 : 0 0	0 : 1 1 : 0 0	0 1 1 0 1 0 1 0					Bass Steps -14dB -12dB : -2dB 0dB 0dB +.2dF +12cB +14dB

For example 12dB Treble and -8dB Bass give the following DATA E YTE . 0 0 1 1 1 0 0 1.

Bass &	Treble	Filter (Characteristics	16		
MSB			. C	R	,	LS

	MSB							LSB	BASS & TREBLE FILTER
	D7	D6	D5	D4	D; J	D2	D1	D0	
		X	eP	y O'			0 0 1	0 1 0 1	Treble Center Frequency = 10 KHz Center Frequency = 12.5 KHz Center Frequency = 15 KHz Center Frequency = 17.5 KHz
2	05	0	1 0 1 1	1 0 1 0 1	0 1 1	0 1 0 1			Bass Center Frequency = 60 Hz Center Frequency = 70 Hz Center Frequency = 80 Hz Center Frequency = 100Hz Center Frequency = 150Hz Quality factor = 1 Quality factor = 1.25 Quality factor = 1.5 Quality factor = 2 DC-Gain = $0dB$ DC-Gain = $\pm 4.4dB$
	1								must be "1"

For example Treble center frequency = 15kHz, Bass center frequency = 100Hz, Bass Q = 1 and DC = 0dB give the following DATA BYTE: 1 0 0 0 1 1 1 0

MSB							LSB				
D7	D6	D5	D4	D3	D2	D1	D0				
1 0 0 1 1	1 0 1 0 1	0 0 0 0 0 0 0 0 0 0 1	0 0 1 1 1 1 1 1 1 1	0 0 1 1 1 1 1 1 1	0 0 1 0 0 0 1 1 1 1	0 0 1 0 1 1 0 1 1	0 1 : 1 0 1 0 1 0 1	Attenuation OdB -1dB : -23dB -24.5dB -26dB -26dB -28dB -30 -32dB -35dB -40dB -50dB Speaker Mute Must be "1" (except RR speaker; see below) Blank Time a.'i. (on.) at RR speaker) 38 μ s 25.5 μ s 32 μ s			
Stereodecoder											
MOD				. NC				FUNCTION			

Speaker Attenuation (LF, LR, RF, RR)

Stereodecoder

	MSB				110			LSB	FUNCTION
	D7	D6	D5	D1	D3	D2	D1	D0	
			Q	P.				0 1	STD Unmuted STD Muted
		jet	Q,			0 0 1 1	0 1 0 1		IN-Gain 11dB IN-Gain 8.5dB IN-Gain 6dB IN-Gain 3.5dB
	\sim				1				must be "1"
\mathbf{D}			1 1	0 1					Forced MONO MONO/STEREO switch automatically
		0 1							Pilot Threshold HIGH Pilot Threshold LOW
	0 1								Deemphasis 50μs Deemphasis 75μs

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Noiseblanker

MSB							LSB	FUNCTION				
D7	D6	D5	D4	D3	D2	D1	D0					
					0 0 0 1 1 1	0 0 1 1 0 0	0 1 0 1 0 1 0	Low Threshold 65mV Low Threshold 60mV Low Threshold 55mV Low Threshold 50mV Low Threshold 45mV Low Threshold 45mV Low Threshold 35mV				
		0	0 0 1 1	0 1 0 1		1	1	Low Threshold 30mV Noise Controlled Threshold 320mV Noise Controlled Threshold 260mV Noise Controlled Threshold 200mV Noise Controlled Threshold 140r iv Noise blanker OFF Noise blanker ON				
0 0 1 1	0 1 0 1							Over deviation Adjust 2.8V Over deviation Adjust 2.0V Over deviation # djust 1.2V Over deviation Detector OFF				
Fieldsti	Fieldstrength Control											
MSB					16	LSB	FUNCTION					

Fieldstrength Control

	MSB					16		LSB	FUNCTION
	D7	D6	D5	D4	D3	<u>.</u> 72	D1	D0	
				000			0 0 1 1	0 1 0 1	Noiseblanker Field strength Adj 2.3V Noiseblanker Field strength Adj 1.8V Noiseblanker Field strength Adj 1.3V Noiseblanker Field strength Adj OFF
		et	e		0 0 1 1	0 1 0 1			VSBL at 33% REF 5V VSBL at 42% REF 5V VSBL at 50% REF 5V VSBL at 58% REF 5V
\mathbf{i}	05	2	0 0 1 1	0 1 0 1					VHCH at 42% REF 5V VHCH at 50% REF 5V VHCH at 58% REF 5V VHCH at 66% REF 5V
		1 0							VHCL at 17% VHCH VHCL at 33% VHCH
	0 1								High cut OFF High cut ON

Configuration

MSB					FUNCTION			
D7	D6	D5	D4	D3	D2	D1	D0	
1	1	1	1	1	1	1	1	setting fix

(

Stereodecoder Adjustment

MSB							LSB	FUNCTION			
D7	D6	D5	D4	D3	D2	D1	D0				
					0 0 1 1	0 0 1 : 0	0 1 0 : 0	Roll-Off Compensation not allowed 20.2% 21.9% : 25.5% : 31.0%			
	0 0 : 1	0 0 : 1	0 0 1 : 1	0 1 0 : 1				LEVEL Gain OdB 0.66dB 1.33dB : 10dB			
0 1								Temperature compensation at LEVEL input TC = 0 TC = 16.7mV/k (3300ppm)			
Testing	Testing										

Testing

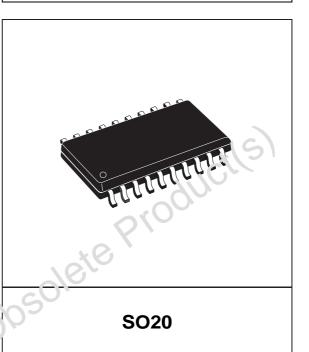
1									
	MSB							1.57	FUNCTION
	D7	D6	D5	D4	D3	D2	D1	DO	
						19		0 1	Stereodecoder test signals OFF Test signals enabled if bit D5 of the subaddress (test mode bit) is set to "1", too
							0 1		External Clock Internal Clock
	05	Jet		C 0 0 0 1 1 1 1 0 0 0 0 1 1 1 1 1 1 1 1	0 0 1 1 0 0 1 1 0 0 1 1 0 0 1 1	0 1 0 1 0 1 0 1 0 1 0 1 0			Testsignals at CASS_R VHCCH Level intern Pilot magnitude VCOCON; VCO Control Voltage Pilot threshold HOLDN NB threshold F228 VHCCL VSBL not used not used PEAK not used REF5V not used
		0 1							VCO OFF ON
	0 1								Audioprocessor test mode only if bit D5 of the subaddress (test mode bit) is set to "1" OFF

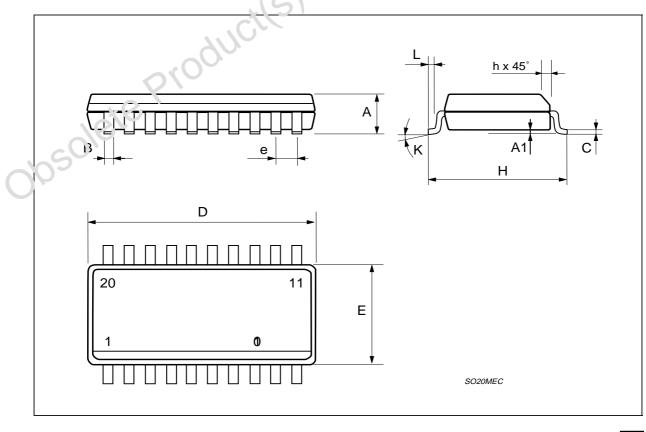
Note : This byte is used for testing or evaluation purposes only and must not be set to other values than the default "11111110" in the application!

TDA7403

DIM.		mm		inch				
	MIN.	TYP.	MAX.	MIN.	TYP.	MAX.		
А	2.35		2.65	0.093		0.104		
A1	0.1		0.3	0.004		0.012		
В	0.33		0.51	0.013		0.020		
С	0.23		0.32	0.009		0.013		
D	12.6		13	0.496		0.512		
E	7.4		7.6	0.291		0.299		
е		1.27			0.050			
н	10		10.65	0.394		0.419		
h	0.25		0.75	0.010		0.030		
L	0.4		1.27	0.016		0.050		
к		()° (min.)	8° (max.)				

OUTLINE AND MECHANICAL DATA





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