

Middle Power Class-D Speaker Amplifier series

20W+20W Full Digital Speaker Amplifier with built-in DSP

BM28720MUV

General Description

BM28720MUV is a Full Digital Speaker Amplifier with built-in DSP (Digital Sound Processor) designed for Flat-panel TVs in particular for space-saving and low-power consumption, delivers an output power of 20W+20W. This IC employs Bipolar, CMOS, and DMOS (BCD) process technology that eliminates turn-on resistance in the output power stage and internal loss due to line resistances up to an ultimate level. With this technology, the IC can achieve high efficiency. In addition, the IC is packaged in a compact reverse heat radiation type power package to achieve low power consumption and low heat generation and eliminates necessity of external heat-sink up to a total output power of 40W. This product satisfies both needs for drastic downsizing, low-profile structures and many function, high quality playback of sound system.

Key Specifications

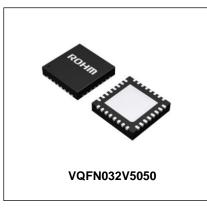
 Supply voltage (VCC) 	10V to 24V
Speaker output power	20W+20W (Typ.)
(VCC=18.5V, R _L =8Ω)	
■ THD+N	0.07 [%] (Typ.)

Applications

- Flat Panel TVs (LCD, OEL)
- Home Audio
- Desktop PC
- Amusement equipments
- Electronic Music equipments, etc.

Package VQFN032V5050

W(Typ) x D(Typ) x H(Max) 5.00mm x 5.00mm x 1.00mm



Features

- This IC includes the DSP (digital sound processor) for Audio signal processing for Flat TVs.
 12 Band/ch BQ, 3 Band DRC, Pre-Scaler, Channel Mixer, Fine Master Volume, Hard Clipper, Level Meter etc.
- This IC has one input systems of digital audio interface. (No needs of Master Clock)
 - I²S / LJ / RJ format
 - LRCLK: 32k/44.1k/48KHz
 - BCLK: 32fs / 48fs / 64fs
 - SDATA: 16 / 20 / 24bit
- This IC has one output systems of digital audio interface.
 - I²S format
 - SDATA: 16 / 20 / 24bit
- With wide range of power supply voltage.
- The monaural output that can reduce the number of external parts can be used.
- With high efficiency and low heat dissipation contributing to miniaturization, slim design, and also power saving of the system.
- Eliminates pop-noise generated during the power supply on/off. High quality muting performance is realized by using the soft-muting technology.
- This IC is built-in with various protection functions for highly reliability design.
 - High temperature protection
 - Under voltage protection
 - Output short protection
 - DC voltage protection
 - Clock stop protection
- Small package

Typical Application Circuit

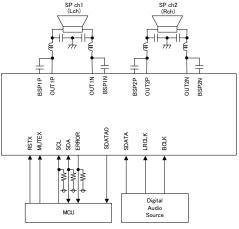


Figure 1. Typical application circuits

OProduct structure : Silicon monolithic integrated circuit OThis product has not designed protection against radioactive rays

Pin configuration and Block diagram

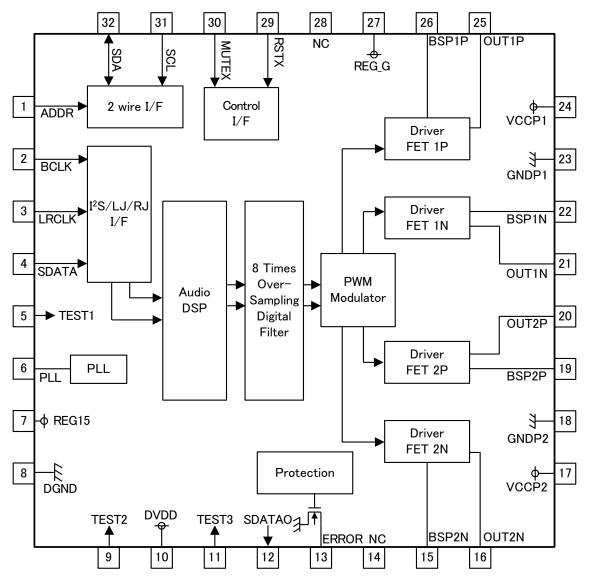


Figure 2. Pin configurations and Block diagram (Top View)

Pin Description

No.	Name	I/O	No.	Name	I/O	No.	Name	I/O	No.	Name	I/O
1	ADDR	Ι	9	TEST2	Ι	17	VCCP2	-	25	OUT1P	0
2	BCLK	I	10	DVDD	-	18	GNDP2	-	26	BSP1P	Ι
3	LRCK	I	11	TEST3	I	19	BSP2P	Ι	27	REG_G	0
4	SDATA	Ι	12	SDATAO	0	20	OUT2P	0	28	NC	-
5	TEST1	I	13	ERROR	0	21	OUT1N	0	29	RSTX	Ι
6	PLL	-	14	NC	-	22	BSP1N	I	30	MUTEX	Ι
7	REG15	0	15	BSP2N	Ι	23	GNDP1	-	31	SCL	Ι
8	DGND	-	16	OUT2N	0	24	VCCP1	-	32	SDA	I/O
	Q	- 41									

I = input; O = output; - = others

Absolute Maximum Ratings (Ta=25°C)

Item	Symbol	Limit	Unit	Conditions	
0 1 1	VCC	-0.3 to 34	V	Pin 17, 24	(Note 1) (Note 2)
Supply voltage	DVDD	-0.3 to 4.5	V	Pin 10	(Note 1)
		3.26	W		(Note 3)
Power dissipation	Pd	4.56	W		(Note 4)
Input voltage 1	VIN1	-0.3 to DVDD+0.3	V	Pin 1 - 5, 9, 11, 12, 13, 29 -32	(Note 1)
Terminal voltage 1	VPIN1	-0.3 to 7.0	V	Pin 27	(Note 1)
Terminal voltage 2	VPIN2	-0.3 to 29	V	Pin 16, 20, 21, 25	(Note 1)(Note 5)
Terminal voltage 3	VPIN3	OUTxx+6.0	V	Pin 15, 19, 22, 26	(Note 1)
Operating temperature range	Topr	-25 to +85	°C		
Storage temperature range	Tstg	-55 to +150	°C		
Maximum junction temperature	Tjmax	+150	°C		

(Note 1) The voltage that can be applied reference to GND (Pin 8, 18, 23).

(Note 2) Do not exceed Pd and Tjmax=150°C.

(Note 3) 74.2mm × 74.2mm × 1.6mm, FR4, 4-layer glass epoxy board

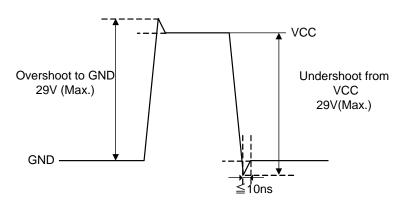
(Top and bottom layer back copper foil size: 20.2mm², 2nd and 3rd layer back copper foil size: 5505mm²)

Derating in done at 26.1 mW/°C for operating above Ta = 25°C. There are thermal via on the board.

(Note 4) 74.2mm × 74.2mm × 1.6mm, FR4, 4-layer glass epoxy board (Copper area 5505mm²)

Derating in done at 36.5 mW/°C for operating above Ta=25°C. There are thermal via on the board.

(Note 5) It should use it below this ratings limit including the AC peak waveform (overshoot) for all conditions. At only undershoot, it is admitted using at ≦10ns and ≦29V by the VCC reference. (Please refer following figure.)





Recommended Operating Ratings (Ta=25°C)

Item	Symbol	Limit	Unit	Conditions	
Supply voltage	VCC	10 to 24	V	Pin 17, 24	(Note 1) (Note 2)
	DVDD	3 to 3.6	V	Pin 10	(Note 1)
		E 4	0	Pin 16, 20, 21, 25	
Minimum load impedance	Р	5.4	Ω	VCC = 18V to 24V	(Note 6)
Minimum load impedance	R∟	2.6	0	Pin 16, 20, 21, 25	
		3.6	Ω	VCC < 18V	(Note 6)

(Note 6) Do not exceed Pd.

Electrical Characteristics

(Unless otherwise specified Ta=25°C, VCC=18V, DVDD=3.3V, RSTX=3.3V, MUTEX=3.3V, f=1kHz, R_L=8 Ω , DSP : Through, fs=48kHz, Snubber circuit for output terminal : R_{snb}=5.6 Ω , C_{snb}=680pF)

lite res	Currente e l		Limit		1.1	Conditions		
Item	Symbol	Min	Тур	Max	Unit			
Total circuit								
Circuit current 1	I _{CC1}	-	45	90	mA	Pin 17, 24, No load		
(Normal mode)	I _{DD1}	-	9	19	mA	Pin 10, -infinity dBFS input, No load		
		-	10	40		Pin 17, 24, No load		
Circuit current 2	I _{CC2}	-	10	40	μA	RSTX=0V, MUTEX=0V		
(Reset mode)			2.5	7.0		Pin 10, -infinity dBFS input, No load		
	DD2	-	2.5	7.0	mA	RSTX=0V, MUTEX=0V		
Open-drain terminal	N/			0.0	V			
Low level voltage	V _{ERR}	-	-	0.8	V	Pin 13, I ₀ =0.5mA		
Regulator output voltage 1	V _{REG_G}	4.9	5.7	6.5	V	Pin 27		
Regulator output voltage 2	V _{REG15}	1.3	1.5	1.7	V	Pin 7		
High level input voltage	V _{IH}	2.5	-	3.3	V	Pin 1 - 5, 9, 11, 29 -32		
Low level input voltage	VIL	0	-	0.8	V	Pin 1 - 5, 9, 11, 29 -32		
Input current		150	100	50		Pin 2 – 4 VIN = 0V		
(Input pull-up terminal)	I _{UP}	-150	-100	-50	μA			
Input current		35	70	105				
(Input pull-down terminal)	I _{DN}		70	105	μA	Pin 1, 29, 30, VIN = 3.3V		
Input current	1	-1	0					
(SCL, SDA terminal)	IIL	-1	0	-	μA	Pin 31, 32, VIN = 0V		
Input current			0	1	μA	Din 21 22 \/IN 2 2\/		
(SCL, SDA terminal)	IIH	-	0	I	μΑ	Pin 31, 32, VIN = 3.3V		
Speaker amplifier output								
Movimum output nowor 1	D		10		W	VCC=13V,THD+N=10%		
Maximum output power 1	P ₀₁	-	10	-	VV	(Note 7)		
Maximum output nower 2	D		20		W	VCC=18.5V,THD+N=10%		
Maximum output power 2	P _{O2}	-	20	-	VV	(Note 7)		
Total harmonic distortion 1	THD1		0.07	-	%	P _o =1W, AES17		
	וטחי	-	0.07	-	70	(Note 7)		
Crosstalk 1	CT1	60	80	-	dB	VCC=13V, Po=1W, A-weighted		
UIUSSIAIN I		00	00	-	UD	(Note 7)		
Output noise voltage 1	V _{NO1}		80	-	µVrms	-infinity dBFS input, A-weighted		
	V NO1	-	00	-	μνιτις	(Note 7)		
	f _{PWM1}	-	256	-	kHz	fs=32 kHz		
PWM sampling frequency	f _{PWM2}	-	352.8	-	kHz	fs=44.1 kHz		
	f _{PWM3}	-	384	-	kHz	fs=48 kHz		

(Note 7) These items show the typical performance of device and depend on board layout, parts, and power supply. The standard value is in mounting device and parts on surface of ROHM's board directly.

Speaker output(Ta=25°C, VCC=18V, DVDD=3.3V, RSTX=0V/3.3V, MUTEX=0V/3.3V, f=1kHz, DSP : Through, fs=48kHz, Snubber circuit for output terminal : R_{snb} =5.6 Ω , C_{snb} =680pF) Measured by ROHM designed 4 layer board.

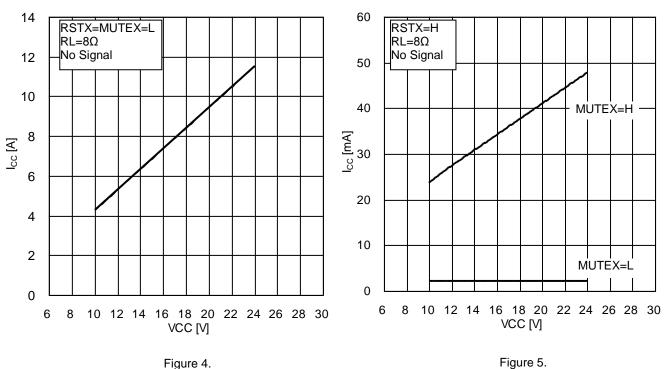
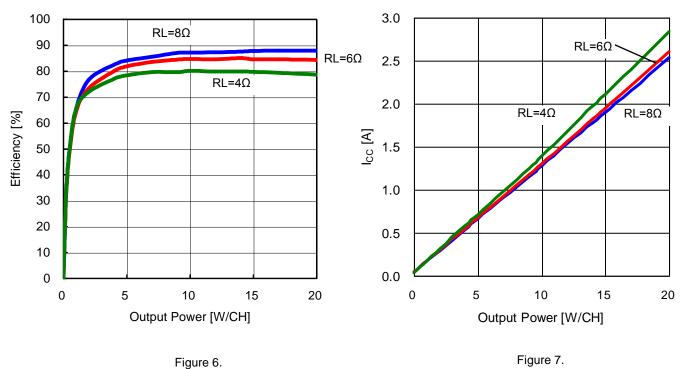


Figure 4. Power supply voltage- Current consumption

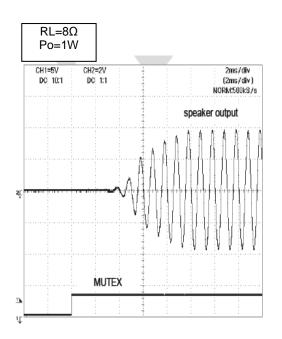
Power supply voltage- Current consumption

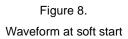


Output power - Efficiency

Output power - Current consumption

Speaker output(Ta=25°C, VCC=18V, DVDD=3.3V, RSTX=3.3V, MUTEX=3.3V, f=1kHz, DSP : Through, fs=48kHz, Snubber circuit for output terminal : R_{snb} =5.6 Ω , C_{snb} =680pF) Measured by ROHM designed 4 layer board.





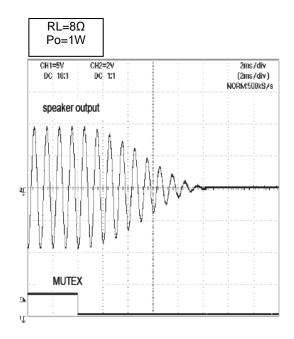
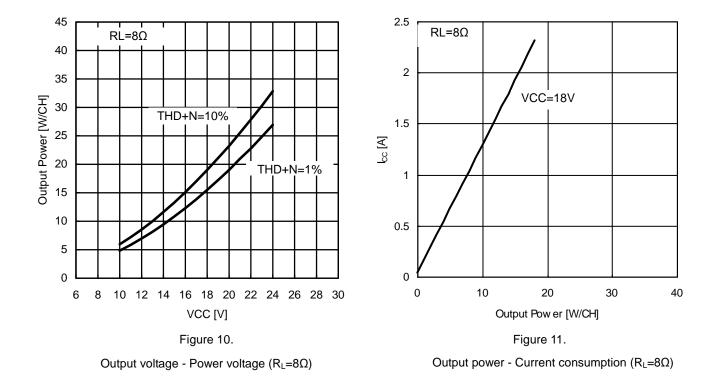
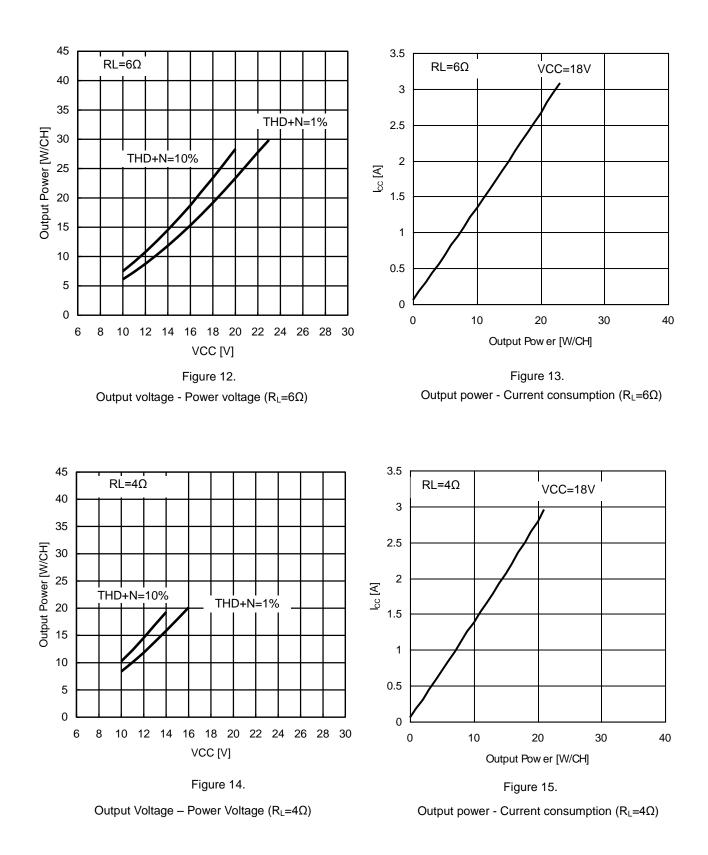


Figure 9. Waveform at soft mute



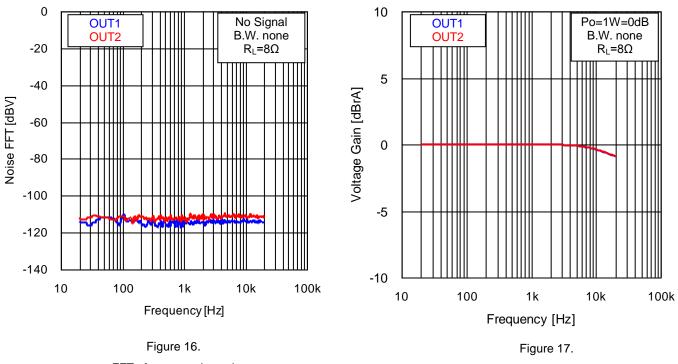
*Dotted line means internal dissipation is over package power.

Speaker output(Ta=25°C, VCC=18V, DVDD=3.3V, RSTX=3.3V, MUTEX=3.3V, f=1kHz, DSP : Through, fs=48kHz, Snubber circuit for output terminal : R_{snb} =5.6 Ω , C_{snb} =680pF) Measured by ROHM designed 4 layer board.

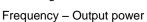


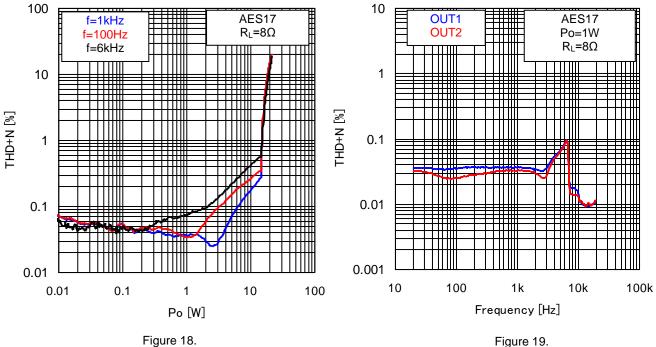
*Dotted line means internal dissipation is over package power.

Speaker output(R_L=8 Ω , Ta=25°C, VCC=18V, DVDD=3.3V, RSTX=3.3V, MUTEX=3.3V, f=1kHz, DSP : Through, fs=48kHz, Snubber circuit for output terminal : R_{snb}=5.6 Ω , C_{snb}=680pF) Measured by ROHM designed 4 layer board.



FFT of output noise voltage

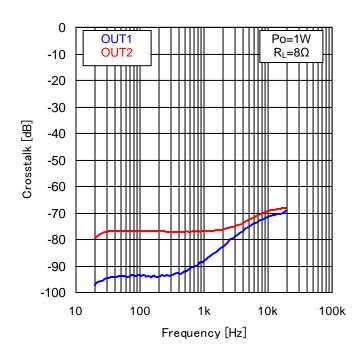


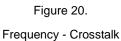


Output Power - THD+N

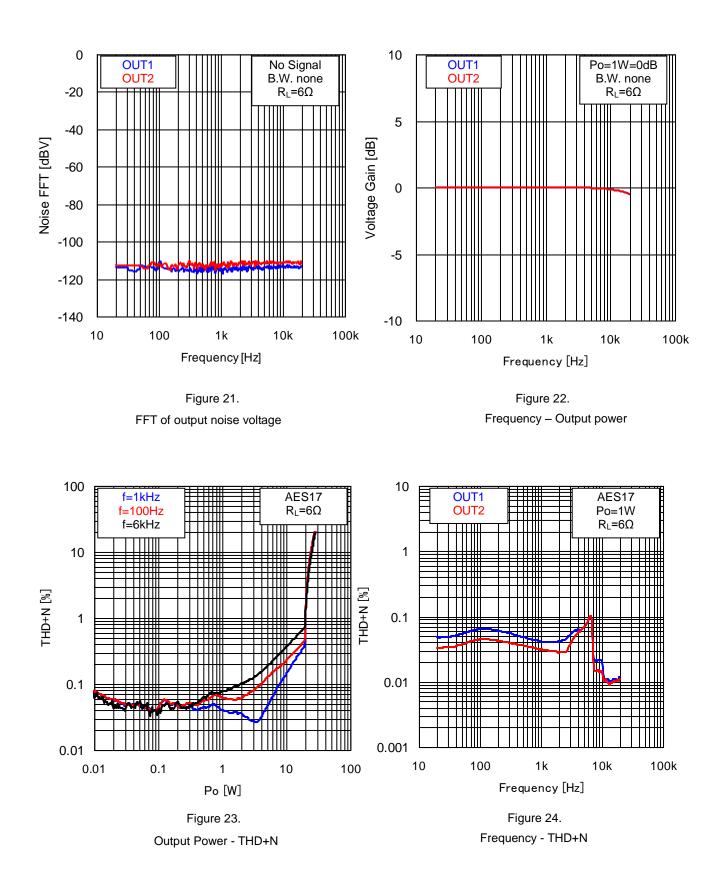
Figure 19. Frequency - THD+N

Speaker output(R_L=8 Ω , Ta=25°C, VCC=18V, DVDD=3.3V, RSTX=3.3V, MUTEX=3.3V, f=1kHz, DSP : Through, fs=48kHz, Snubber circuit for output terminal : R_{snb}=5.6 Ω , C_{snb}=680pF) Measured by ROHM designed 4 layer board.

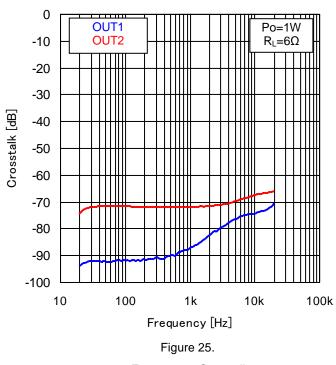




Speaker output(R_L=6 Ω , Ta=25°C, VCC=18V, DVDD=3.3V, RSTX=3.3V, MUTEX=3.3V, f=1kHz, DSP : Through, fs=48kHz, Snubber circuit for output terminal : R_{snb}=5.6 Ω , C_{snb}=680pF) Measured by ROHM designed 4 layer board.

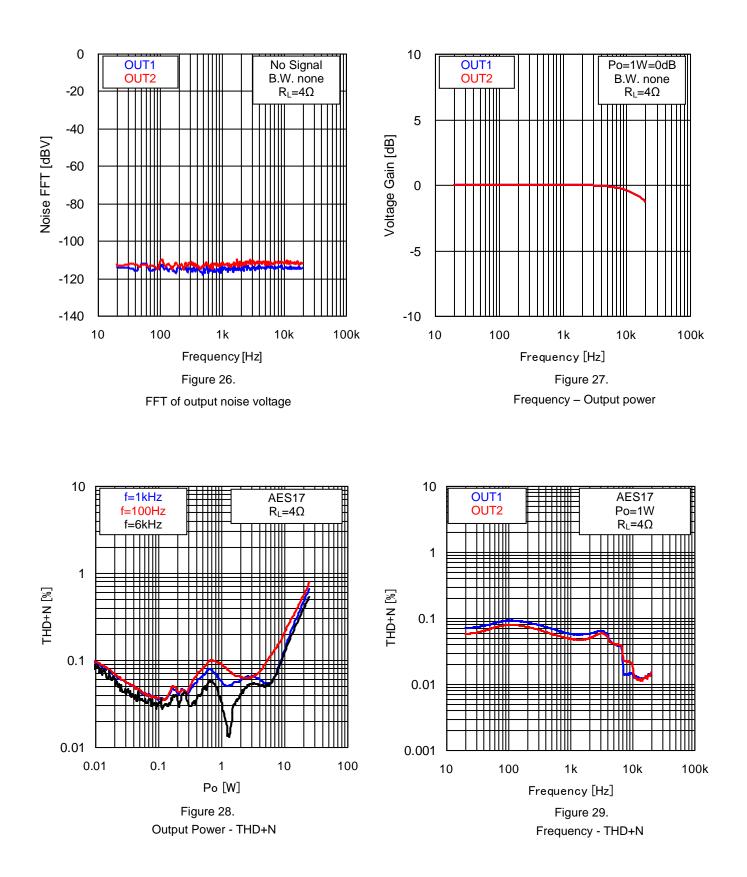


Speaker output R_L=6 Ω , Ta=25°C, VCC=18V, DVDD=3.3V, RSTX=3.3V, MUTEX=3.3V, f=1kHz, DSP : Through, fs=48kHz, Snubber circuit for output terminal : R_{snb}=5.6 Ω , C_{snb}=680pF) Measured by ROHM designed 4 layer board.

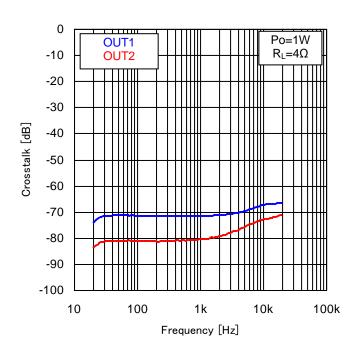


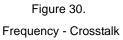
Frequency - Crosstalk

Speaker output(R_L=4 Ω , Ta=25°C, VCC=18V, DVDD=3.3V, RSTX=3.3V, MUTEX=3.3V, f=1kHz, DSP : Through, fs=48kHz, Snubber circuit for output terminal : R_{snb}=5.6 Ω , C_{snb}=680pF) Measured by ROHM designed 4 layer board.



Speaker output(R_L=4 Ω , Ta=25°C, VCC=18V, DVDD=3.3V, RSTX=3.3V, MUTEX=3.3V, f=1kHz, DSP : Through, fs=48kHz, Snubber circuit for output terminal : R_{snb}=5.6 Ω , C_{snb}=680pF) Measured by ROHM designed 4 layer board.





Digital Block Functional Overview

No.	Function	Specification
1	Pre-Scaler	Lch / Rch become same set point.
		 +48dB to -79dB (0.5dB step),-∞dB
2	Channel Mixer	 Lch <= Mute, Lch(default), Rch, (L+R)/2, L-R
		 Rch <= Mute, Lch, Rch(default), (L+R)/2, L-R
		Lch/Rch are independent phase reversal control available
3	12 Band BQ	 12 Band biquad type filter
		 Only 5 coefficient is required.(b0,b1,b2,a1,a2)
		 The Filter types which can be realized is
		Peaking/Low-shelf/High-shelf/Low-pass/High-pass/All-pass/Notch.
		Lch/Rch become same set point or independent set. There is soft transition function.
4	Fine Master Volume	 Lch / Rch become same set point or independent set.
		 +24dB to -103dB (0.125dB step),-∞dB
		There is soft transition function
5	3 Band DRC	Non clip output is achieved
		Available three band operation
		 Threshold level : +12dB to -32dB (0.5dB step)
6	Post-Scaler	Lch / Rch become same set point
		 +48dB to -79dB (0.5dB step),-∞dB
7	Fine Post-Scaler	Lch / Rch become independent set point
		 +0.7dB to -0.8dB (0.1dB step)
8	DC Cut HPF	• Fc : 1Hz
9	Clipper	Lch / Rch become same set point
		Clip level : +3dB to -22.5dB (-0.1dB step)

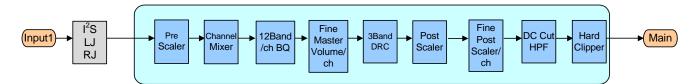


Figure 31. DSP Block diagram

RSTX pin, **MUTEX** pin function

RSTX	MUTEX	DSP block	Speaker output			
(29pin)	(30pin)	condition	condition			
Low	Low	Reset ON	HiZ_low			
Low	Low	Reset ON	(Low consumption)			
	Low	Normal operation	HiZ_low ^(Note 4)			
High	Low	(Mute ON)	(Mute ON)			
Link	Llink	Normal operation	Normal operation			
High	High	(Mute OFF)	(Mute OFF)			
Low	High	Don't use.				

(Note 1) RSTX is set to low, internal registers are initialized.

(Note 2) VCCP1, VCCP2< 2.5V, IC latched by protection circuit and ERROR terminal condition are initialized.

(Note 3) If DVDD is under 3V, RSTX is set to low once for 10ms(min), and set high again. Then DSP is needed to set parameter again.

(Note 4) Speaker output becomes HiZ-low after elapse of PWM stop time after setting MUTEX low. Please refer to PWM sampling frequency.

PWM sampling frequency

PWM sampling frequency of Speaker output and Soft-mute transition time depends on sampling frequency (fs) of the digital sound input. These transition times are changed by sending select address 0x15[1:0].

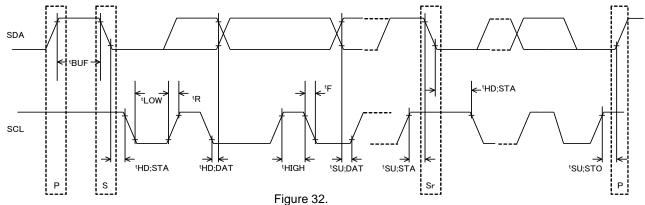
Sampling		0x15[1:0]	Soft mute trai	nsition time	DW/M stop time	
frequency (fs)	PWM frequency	value	Mute ON	Mute OFF	PWM stop time	
		0x0	10.7ms	10.7ms	86ms	
	004111-	0x1	21.4ms	10.7ms	106ms	
48kHz	384kHz	0x2	42.7ms	10.7ms	125ms	
		0x3	85.4ms	10.7ms	162ms	
		0x0	11.7ms	11.7ms	93ms	
		0x1	23.3ms	11.7ms	113ms	
44.1kHz	352.8kHz	0x2	46.5ms	11.7ms	135ms	
		0x3	92.9ms	11.7ms	177ms	
		0x0	16.1ms	16.1ms	116ms	
22 64-		0x1	32.1ms	16.1ms	148ms	
32 kHz	256kHz	0x2	64.1ms	16.1ms	178ms	
		0x3	128.1ms	16.1ms	241ms	

Wait time(Twait) rules from releasing RSTX to releasing MUTEX.

When the time from releasing RSTX to releasing MUTEX prescribes **Twait**, please secure **Twait** by all means more than 450ms.

2 wire Bus control signal specification

1) Electrical characteristics and Timing of Bus line and I/O stage



SDA and SCL bus line characteristics	Unless otherwise specified Ta=25°C, VDD=3.3V)
	$\frac{10}{10}$

	Parameter	Symbol	High spee	Unit	
	Falameter	Symbol	Min.	Max.	Unit
1	SCL clock frequency	fSCL	0	400	kHz
2	Bus free time between 「Stop」 condition and 「Start」 condition	tBUF	1.3	—	μs
3	Hold-time of (sending again) Start condition. After this period the first clock pulse is generated.	tHD;STA	0.6	_	μs
4	SCL clock's LOW state Hold-time	tLOW	1.3	—	μs
5	SCL clock's HIGH state Hold-time	tHIGH	0.6	—	μs
6	Set-up time of sending again Start condition	tSU;STA	0.6	—	μs
7	Data hold time	tHD;DAT	0 (Note 1)	—	μs
8	Data set-up time	tSU;DAT	250	—	ns
9	Rise-time of SDA and SCL signal	tR	20+0.1Cb	300	ns
10	Fall-time of SDA and SCL signal	tF	20+0.1Cb	300	ns
11	Set-up time of Stop condition	tSU;STO	0.6	—	μs
12	Capacitive load of each bus line	Cb	_	400	pF

The above-mentioned numerical values are all the values corresponding to VIH min and the VIL max level.

(Note 1) To exceed an undefined area on the fall-edge of SCL (VIH min of the SCL signal), the transmitting set should internally

offer the holding time of 300ns or more for the SDA signal.

(Note 2) SCL and SDA pin is not corresponding to threshold tolerance of 5V.

Please use it within 4.5V of the absolute maximum rating.

2)Command interface

2 wire Bus control is used for command interface. It not only writes but also it is possible to read it excluding a part of register. In addition to "Slave Address", set and write 1 byte of "Select Address" to read out the data. 2 wire Bus Slave mode format is illustrated below.

_		MSB	LSB		MSB	LSB		MSB	LSB		
	S	Slave Addre	ess	А	Select Ac	ddress	А	Data		А	Ρ
								•			

Figure 33.

S : Start Condition

Slave Address : The data of eight bits in total is sent putting up bit of Read mode (high) or Write mode (low) after slave address (7bit) set with the terminal ADDR. (MSB first)

- A : The acknowledge bit adds to data that the acknowledge is sent and received in each byte. When data is correctly sent and received, "low" is sent and received.
 - There was no acknowledgement for "high".
- Select Address : The select address in one byte is used.(MSB first)
- Data : Data byte is sent and received data(MSB first)

P : Stop Condition

S		M	SB 6		5			LSB		f	_	
	Start Conditi	on							Stop	conditio	on	
3)Slave Ac	SDA↓ SCL="ł ddress	nigh"		Fi	gure 34.				SDA † S	SCL="h	nigh"	
• While	ADDR pin is "lov	v"										
MSB									LSB			
A	6 A5	A4	A3		A2	A1	A0	R/\	N			
1	0	0	0		0	0	0	1/0	0			
	ADDR pin is "hig	Jh"										
MSB	0								LSB			
A		A4 0	A3 0		A2 0	A1 0	A0	R/\ 1/0				
	0	0	0		0	0	1	1/	5			
4) Writing • Basic				Fi	gure 35.							
S	Slave Address	A Sel	ect Address	a A	Data	a A	Р					
				: M	aster to S	Slave,	: SI	ave to I	Master			
• Auto-i	ncrement format											
S	Slave Address	A Sel	ect Address	A	Data	1 A	Data 2	А	Da	ta 3…I	N A P	
				: M	aster to S	Slave,	: SI	ave to I	Master			
	6 1 .			Fi	gure 36.							
5)Reading First of a	of data all, the destinatio	n addres	s (0x20 in tł	ne exa	mple) for	reading	is written ir	n the re	gister of	the 0xl	D0 address at th	าย
	eading. In the fol	llowing st	ream, data i	s read	after the	slave ad	dress. Plea	ase do r	not return	ackno	wledge when yo	วน
end the	reception.											
S	Slave Address	A F	Req_Addr	А	Select	Address	A P					
(ex.)	0x80		0xD0			0x20						
S	Slave Address	A C	Data 1 A	۱ (Data 2	Α		A	Data N	Ā	P	
(ex.)	0x81		0x**		0x**				0x**			
	: Master to Sla	ave,	: Slave	to Ma	ster, A :	With Ack	nowledge,	Ā : W	ithout Ac	knowle	edge	

Figure 37.

Format of digital audio input

- LRCLK: It is L/R clock input signal. It corresponds to 32kHz/44.1kHz/48kHz and that is same as the sampling frequency (fs). This signal shows left or right of audio data.
- BCLK: It is Bit Clock input signal.
 It is used for the latch of data in rising edge. Frequency of BCLK is 64 times of sampling frequency (64fs) or 48 times (48fs) or 32 times (32fs). If BCLK frequency is 32fs, only 16 bit data width is available.
- SDATA: It is data input signal.
 It is amplitude data. The data length is different according to the resolution of the input digital data.
 Data width is selectable as 16 bit, 20 bit or 24 bit.

The digital input has I2S, Left-justified and Right-justified formats. The figure below shows the timing chart of each transmission mode.

SDATAO: Audio data after DSP processing.
 Data is audio data after DSP processing.
 This output is synchronous to LRCK and BCLK.
 Output supports only I2S format.

BCLK clock 64fs

I²S 64fs Format

LRCLK	Left Channel]	Right Channel	
BCLK∐	MSB LSB	22 23 24 25 26 27 28 29 30 31 32	33 34 35 36 37 38 39 40 41 42 43 44 45 46 47 48 49 50 51 52 MANANANANANANANANANANANANANANANANANANAN	2 53 54 55 56 57 58 59 60 61 62 63 64
SDATA	S 1413121110 9 8 7 6 5 4 3 2 1 0		S 1413121110 9 8 7 6 5 4 3 2 1 0	
	16bit Mode 20bit Mode 24bit Mode	1	16bit Mode 20bit Mode 24bit Mode	

Left-Justified 64fs Format

LRCLK	Left Channel		Right Channel	
BCLK	1 2 3 4 5 6 7 8 9 10 11 12 13 14 15 16 17 18 19 20 MSB LSB	0 21 22 23 24 25 26 27 28 29 30 31 32	33 34 35 36 37 38 39 40 41 42 43 44 45 46 47 48 49 50 51 52	
SDATA	S 1413121110 9 8 7 6 5 4 3 2 1 0		S 14131211109876543210	
l I	16bit Mode	1	16bit Mode	1
	20bit Mode		20bit Mode	
ı	24bit Mode		24bit Mode	

Right-Justified 64fs Format

LRCLK	Left Channel	Right Channel
12345678	3 9 10 11 12 13 14 15 16 17 18 19 20 21 22 23 24 25 26 27 28 29 30 31 32	33 34 35 36 37 38 39 40 41 42 43 44 45 46 47 48 49 50 51 52 53 54 55 56 57 58 59 60 61 62 63 64
	MSB	MSB LSB
SDATA	S 1413121110 9 8 7 6 5 4 3 2 1 0	S 14131211109876543210
	16bit Mode	16bit Mode
	20bit Mode	20bit Mode
	24bit Mode	24bit Mode

Figure 38.

BCLK clock 48fs

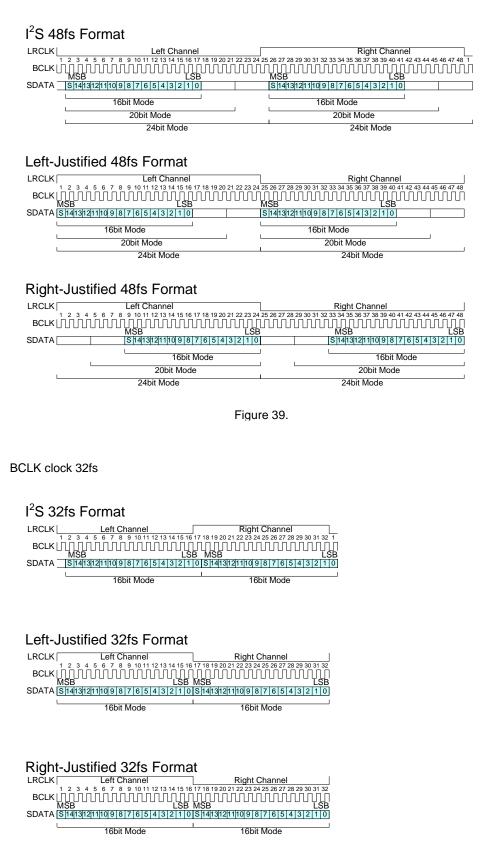


Figure 40.

Format setting for Digital Audio Interface

Please set BCLK clock fs, Data length and Format by transmitting command according to inputted Digital Serial Audio signal.

SDATAO output data bit width is able to be set independently.

Output supports only I²S format.

BCLK clock

Default = 0x0 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x03[5:4]	0x0	64fs
	0x1	48fs
	0x2	32fs
	0x3	Don't use

Data format

Default = 0x0 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x03[3:2]	0x0	I ² S format
	0x1	Left-justified format
	0x2	Right-justified format
	0x3	Don't use

Data width

Default = 0x2 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x03[1:0]	0x0	16 bit
	0x1	20 bit
	0x2	24 bit
	0x3	Don't use

SDATAO output data width

Default = 0x2 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x78[1:0]	0x0	16 bit
	0x1	20 bit
	0x2	24 bit
	0x3	Don't use

Audio Interface format and timing

Recommended timing and operating conditions (BCLK, LRCLK, SDATA)

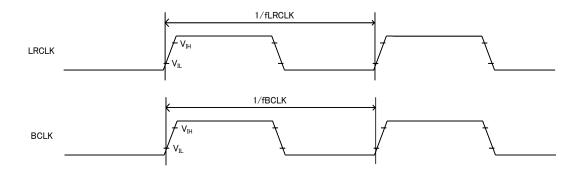
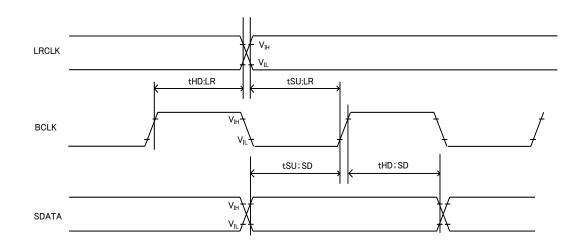
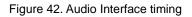


Figure 41. Clock timing

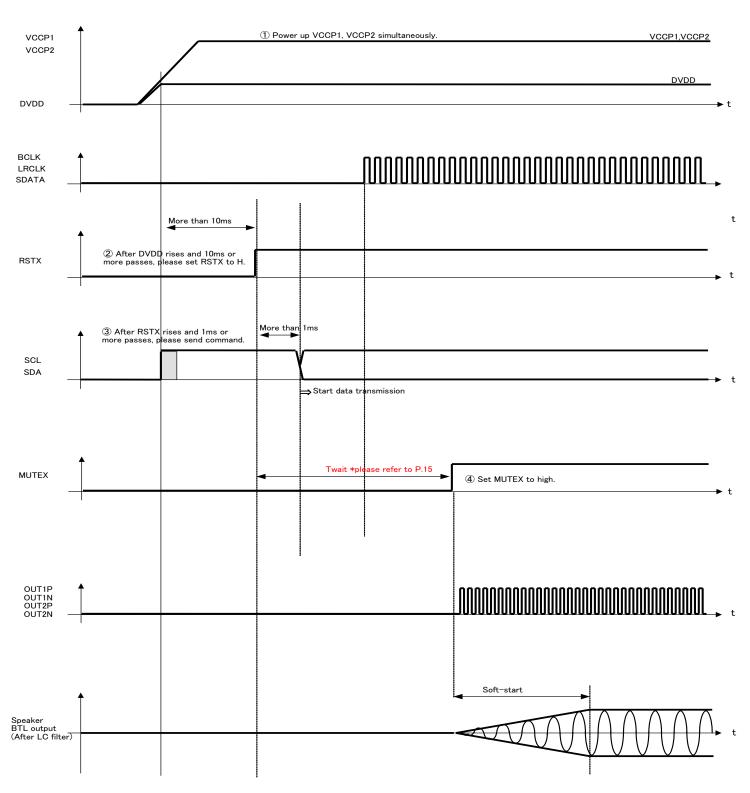




No.	Parameter	Symbol	L	imit	Unit
INO.	Falailletei	Symbol	Min.	Max.	Unit
1	LRCLK frequency	fLRCLK	32	48	kHz
2	BCLK frequency	fBCLK	2.048	3.072	MHz
3	Setup time, LRCLK ^(Note 1)	tSU;LR	20	—	ns
4	Hold time, LRCLK ^(Note 1)	tHD;LR	20	—	ns
5	Setup time, SDATA	tSU;SD	20	_	ns
6	Hold time, SDATA	tHD;SD	20	_	ns
7	LRCLK, DUTY	dLRCLK	40	60	%
8	BCLK, DUTY	dBCLK	40	60	%

(Note 1) This regulation is to keep rising edge of LRCK and rising edge of BCLK from overlapping.

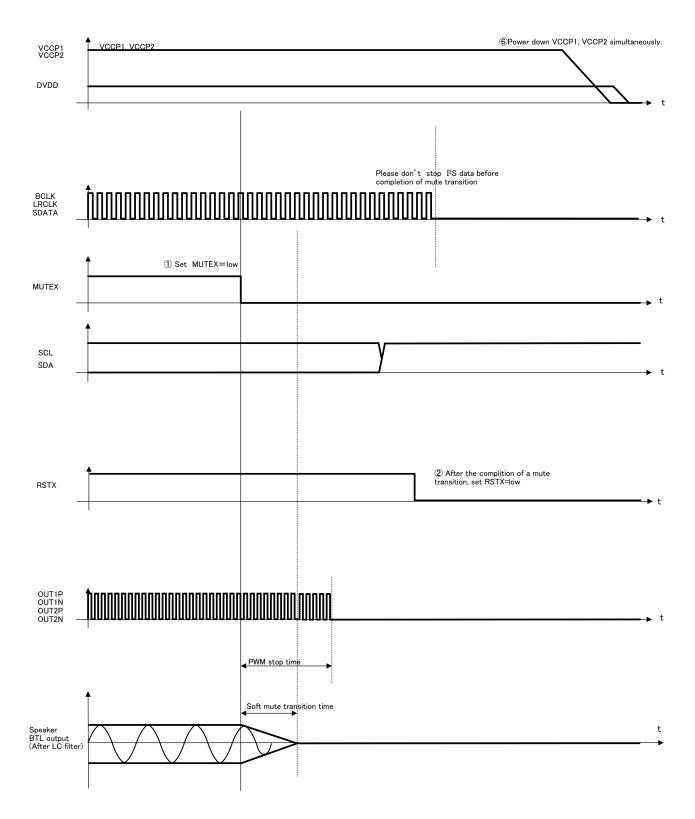
Power supply start-up sequence



*Please make sure to input low to RSTX terminal from external at the time power up DVDD. *Please refer to [7. The wake-up Procedure of power-up].

Figure 43.

Power supply shut-down sequence



%Please make sure to input low to RSTX terminal when DVDD is powered down.

Figure 44.

About the protection function

Protection function		Detecting & Releasing condition	Speaker PWM output	ERROR ^(Note 1) output
Output short protection	Detecting condition	Detecting current = 7.2A (TYP.)	HiZ_low (latch) ^(Note 2)	Low (latch)
DC voltage protection	Detecting condition	PWM output Duty=0% or 100% for 12µs(TYP)and over	HiZ_low (latch) ^(Note 2)	Low (latch)
High	Detecting condition	Chip temperature to be above 150°C (TYP.)	HiZ_low	Low
temperature protection	Releasing condition	Chip temperature to be below 120°C (TYP.)	Normal operation	LOW
Under voltage	Detecting condition	Power supply voltage to be below 7.0V (TYP.)	HiZ_low	High
protection	Releasing condition	Power supply voltage to be above 7.5V (TYP.)	Normal operation	High
Clock stop	Detecting condition	BCLK signal have stopped among constant period. LRCLK signal have stopped among constant period. BCLK frequency is under constant value. BCLK frequency is over constant value. Please refer to P.55-58 about constant value.	HiZ_low	High
protection	Releasing condition	LRCLK signal haven't stopped among constant period and BCLK continues maximum 60ms or more of continuous appropriate frequency. Please refer to P.55-58 about constant value.	Normal operation	

(Note 1) The ERROR pin is Nch open-drain output.

(Note 2) Once an IC is latched, the circuit is not released automatically even after an abnormal status is removed.

The following procedures ① or ② is available for recovery.

①After MUTEX pin is made low once over PWM stop time, MUTEX pin is returned to high again.
 ②Turning on the power supply again (VCCP1, VCCP2<2.5V, 10ms(min)).

- 1) Output short protection (Short to the power supply)
 - This IC has the output short protection circuit that stops the PWM output when the PWM output is short-circuited to the power supply due to abnormality.
 - Detecting condition It will detect when MUTEX pin is set high and the current that flows in the PWM output pin becomes 7.2A(TYP.) or more. The PWM output instantaneously enters the state of HiZ-low if detected, and IC does the latch.

Releasing method - ①After MUTEX pin is set low once over the PWM stop time(see page 15), MUTEX pin is returned to high again.

②Turning on the power supply again (VCCP1, VCCP2<2.5V, 10ms(min)).

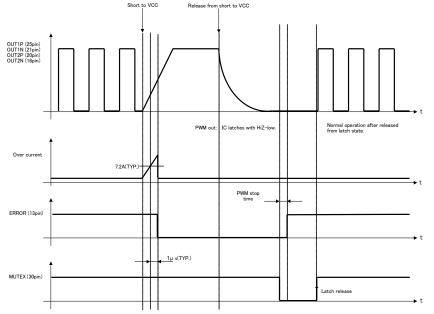


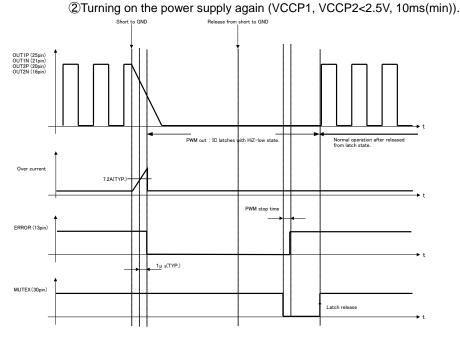
Figure 45.

2) Output short protection (Short to GND)

This IC has the output short protection circuit that stops the PWM output when the PWM output is short-circuited to GND due to abnormality.

Detecting condition - It will detect when MUTEX pin is set high and the current that flows in the PWM output terminal becomes 7.2A(TYP.) or more. The PWM output instantaneously enters the state of HiZ-low if detected, and IC does the latch.

Releasing method – ①After MUTEX pin is set low once over the PWM stop time(see page 15), MUTEX pin is returned to high again.



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Figure 46.

- 3) DC voltage protection in the speaker
 - When the DC voltage in the speaker is impressed due to abnormality, this IC has the protection circuit where the speaker is defended from destruction.
 - Detecting condition It will detect when MUTEX pin is set high and PWM output Duty=0% or 100% over 12µs.(fs=48kHz) Once detected, The PWM output instantaneously enters the state of HiZ-low, and IC does the latch.
 - Releasing method ①After MUTEX pin is set low once over the PWM stop time(see page 15), MUTEX pin is returned to high again.

(2) Turning on the power supply again (VCCP1, VCCP2<2.5V, 10ms(min)).

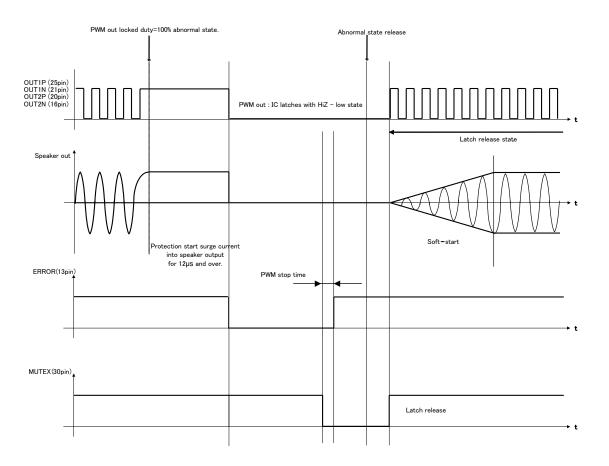


Figure 47.

4) High temperature protection

This IC has the high temperature protection circuit that prevents thermal reckless driving under an abnormal state for the temperature of the chip to exceed Tjmax=150°C.

Detecting condition - It will detect when MUTEX pin is set high and the temperature of the chip becomes 150°C(TYP.) or more. The speaker output is muted when detected.

Releasing condition - It will release when MUTEX pin is set high and the temperature of the chip becomes 120°C(TYP.) or less. The speaker output is outputted when released.

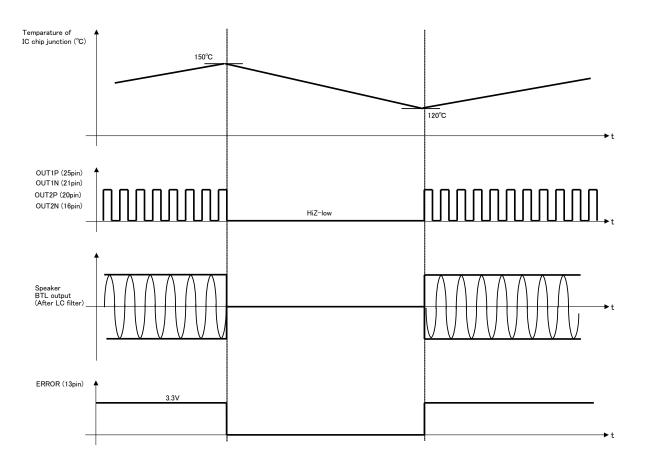


Figure 48.

5) Under voltage protection

This IC has the under voltage protection circuit that make speaker output mute once detecting extreme drop of the power supply voltage.

Detecting condition – It will detect when MUTEX pin is set high and the power supply voltage becomes lower than 7.0V. The speaker output is muted when detected.

Releasing condition – It will release when MUTEX pin is set high and the power supply voltage becomes more than 7.5V. The speaker output is outputted when released.

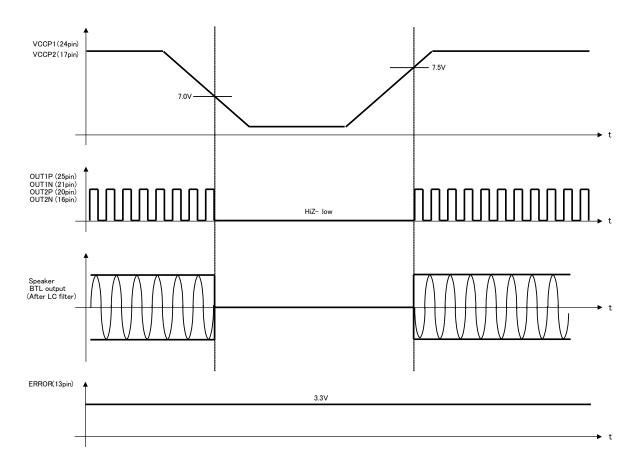


Figure 49.

- 6) Clock stop protection
 - This IC has the clock stop protection circuits that make the speaker output mute when the BCLK and LRCLK frequency of the digital sound input are decreased or low frequency.
 - Detecting condition BCLK frequency is low or stop, LRCLK frequency is stop. Speaker output is muted.
 - Releasing condition Mute is released when it passes more than 60ms after an internal error state was removed. (max).

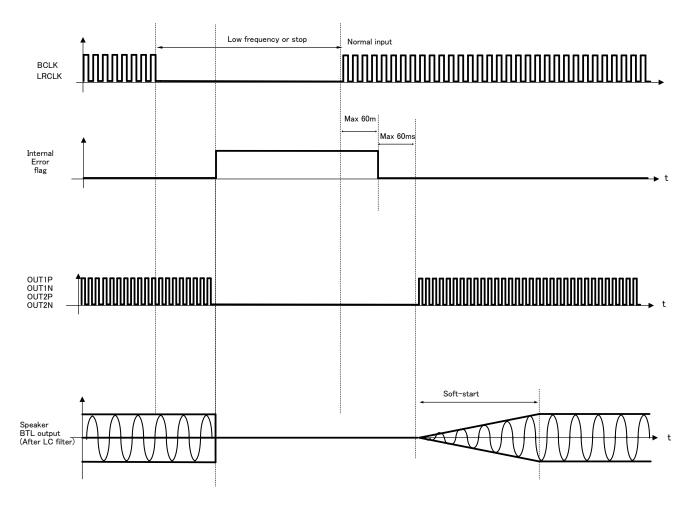


Figure 50.

Functional descriptions of DSP Block

1. Digital Sound Processing(DSP)

The digital sound processing (DSP) part of BM28720 is composed of the special hardware which is the optimal for FPD-TV, the Mini/Micro Compo. BM28720MUV does the following processing using this special DSP.

Pre-Scaler, Channel Mixer, 12 Band BQ, Fine Master Volume, 3 Band DRC, Fine Post-Scaler, DC Cut HPF, Hard Clipper

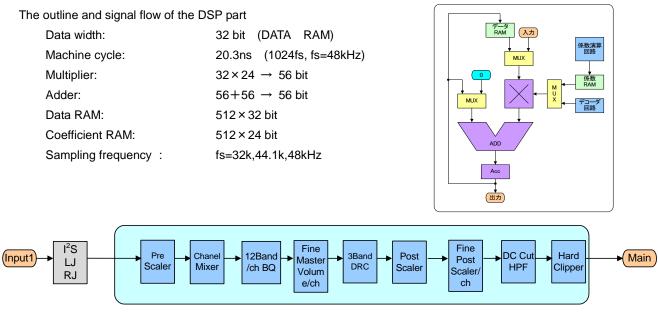


Figure 51.

The digital signal from 16 bits to 24 bits is inputted to the DSP but extends 8bit(+48dB) as the overflow margin to the upper side. When doing the processing which exceeds this range, it processes a clip in the DSP. Incidentally, in case of the 2nd IIR-type (BQ) filter which is often used generally as the digital filter, because it consumes a lot of overflow margins, the output of the multiplier and the adder inside needs note.

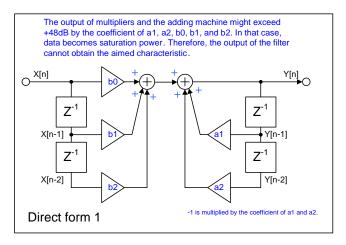
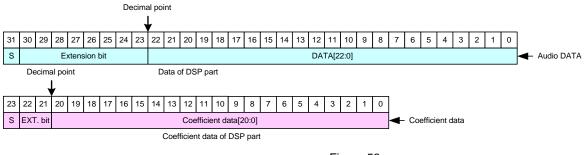


Figure 52.

The management of audio data is as follows by each block.

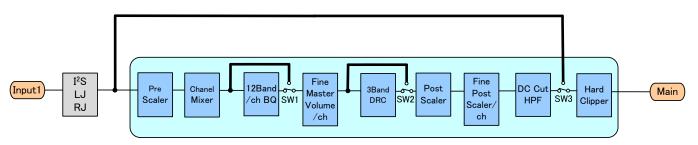




1-1. Bypass

It enable pass the some function of the DSP. By using it, it left the set value of the each function and can be passed that function. it is possible to make ON/OFF of the sound effect easily.

1) 12 Band BQ, 2) 3 Band DRC and the 3) whole DSP can be passed.





Default = 0x00 *Blue square means initial value.

Select Address	bit	Explanation of ope	eration
	2	Bypass of 12 Band BQ (SW1)	0:Normal 1:Bypass
0x02 [2:0]	1	Bypass of 3 Band DRC (SW2)	0:Normal 1:Bypass
	0	Bypass of DSP (SW3)	0:Normal 1:Bypass

1-2. Pre-Scaler

Pre-Scaler adjusts gain of input audio data. The adjustable-range can be set from +48 dB to -79 dB with the 0.5-dB step. (Lch/Rch use same value)

Pre-Scaler doesn't have a soft transfer feature.

Select Address	Explanation	on of operation
0x16 [7:0]	Command Value	Gain
	0x00	+48dB
	0x01	+47.5dB
	:	:
	0x60	0dB
	0x61	-0.5dB
	0x62	-1dB
	:	÷
	0xFE	-79dB
	0xFF	-∞

1-3. Channel setup with a phase inversion function (Channel Mixer 1)

It sets a mixing in the sound on the left channel and the right channel of the digital signal which was inputted to the DSP. It makes a stereo signal a monaural here. Also, the phase-inversion, the mute on each channel can be set.

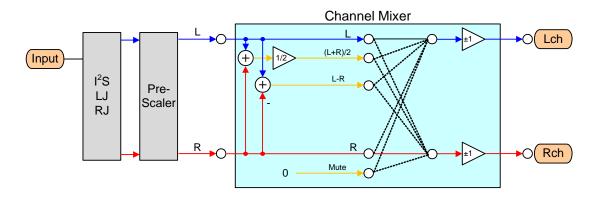


Figure 55.

DSP input: The data inputted into Lch of DSP is inverted.

Default = 0x0 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x17[7]	0x0	Normal
	0x1	Invert

DSP input: The data inputted into Lch of DSP is mixed.

Default = 0x1 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x17 [6:4]	0x0	Mute
	0x1	Lch data input
	0x2	Rch data input
	0x3	(Lch + Rch) / 2
	0x4	Lch-Rch

DSP input: The data inputted into Rch of DSP is inverted.

Default = 0x0 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x17 [3]	0x0	Normal
	0x1	Invert

DSP input: The data inputted into Rch of DSP is mixed.

Default = 0x2 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x17 [2:0]	0x0	Mute
	0x1	Lch data input
	0x2	Rch data input
	0x3	(Lch + Rch) / 2
	0x4	Lch-Rch

1-4. Biquad filter

In this IC, the following block has the biquad type filter(BQ).

There are 12 Band BQ, Crossover filter of 3 Band DRC block and BQ of the smooth transition.

The shape is used as peaking filter, low shelf filter, high shelf filter, low-pass filter, high-pass filter, notch filter and all path filters.

Setting the coefficient of the digital filter in the IC by transmit to the coefficient RAM via command. 12 Band BQ have the soft transfer feature. About the detailed order of the parameter setting, please refers to the following BQ setting method.

Select of BQ independence or synchronous setting

Default = 0x0 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x60 [4]*	0x0	L/R common setting
0.000[1]	0x1	L/R independence setting

*0x60[4] setting note.

Please re-set all BQ when you change the setting of 0x60[4]. Total 18 BQ needs to be set again. (BQ1-12, DRC1, DRC2 and DRC3).

BQ soft transition destination band

Default = 0x0 *Blue square means initial value.

Select Address	Exp	planation of c	peration	
0x51 [4:0]*	Value	Destination	Value	Destination
	0x00	12BAND(1)	0x0A	12BAND(11)
	0x01	12BAND(2)	0x0B	12BAND(12)
	0x02	12BAND(3)		
	0x03	12BAND(4)		
	0x04	12BAND(5)		
	0x05	12BAND(6)		
	0×06	12BAND(7)		
	0x07	12BAND(8)		
	0x08	12BAND(9)		
	0x09	12BAND(10)		

Select of smooth transition

Default = 0x0 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x53 [6]	0x0	Use smooth transition
	0x1	Not use smooth transition

Select the destination channel of smooth transition

Default =	0x0	*Blue	square	means	initial	value
Delault –	0.00	Dide	Square	means	muai	value.

Select Address	Value	Explanation of operation
0x53 [5:4]	0x0	Lch と Rch
	0x1	Lch
	0x2	Rch
	0x3	Don't use

Setting of smooth transition time

Default = 0x3 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x53 [3:2]	0x0	2.7ms
	0x1	5.3ms
	0x2	10.7ms
	0x3	21.3ms

Setting of smooth transition wait time

Default = 0x0 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x53 [1:0]	0x0	2.7ms
	0x1	5.3ms
	0x2	10.7ms
	0x3	21.3ms

Setting of smooth transition start

Default = 0x0 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x58 [0]	0x0	Stop the smooth transition operation
	0x1	Start the smooth transition operation (After the transition is completed, it becomes 0 automatically)

This register is write only.

Read-out smooth transition status

Read only

Select Address	Explanation of operation
0x59 [0]	1 : Executing soft transition
	0 : Not during soft transition

1-5. Volume

Volume is from+24dB to -103dB, and can be selected by the step of 0.125dB. And it is possible to be setting of $-\infty$ dB, too. At the time of switching of Volume, smooth transition is performed. Soft transition duration is optional with the command. L/R synchronous or L/R independent can be selected by 0x10[7].

It becomes the following formula at the transition from AdB to BdB. C is smooth transition duration selected by 0x15[7:6] command.

Transition time =
$$|(10^{20} - 10^{20})^*$$
 C ms|

Setting of soft transition time

Default = 0x0 *Blue square means initial value.

Select Address	Value	動作説明
0x15 [7:6]	0x0	21.3ms
	0x1	42.7ms
	0x2	85.3ms
	0x3	Don't use

Lch/common volume setting

Select Address	Explanation of operation			
0x11 [7:0]		Value	Gain	
		0x00	+24dB	
		0x01	+23.5dB	
			:	
		0x30	0dB	
		0x31	-0.5dB	
		0x32	-1dB	
		:	:	
		0xFE	-103dB	
		0xFF	-∞	

Setting of fine volume

This command becomes effective by sending the following command after setting. When using this command, it is possible to set a volume in 0.125dB carving. When L/R synchronous volume setting, 0x11[7:0] is enable. When L/R independent volume setting, 0x11[7:0] is the volume setting of Lch.

Lch/common fine volume setting

Select Address	Value	Explanation of operation
0x10 [1:0]	0x0	0dB
	0x1	-0.125dB
	0x2	-0.25dB
	0x3	-0.375dB

[Note1]

It is possible to use with the 0.5-dB step in changing only 0x11[7:0] when 0x10[1:0]=0x0.

The Lch/Rch independent volume setting and the synchronous volume setting can be selected by 0x10[7] command. When Lch/Rch independent volume set, the volume setting of Lch is the setting of 0x10[1:0] and 0x11, and the volume setting of Rch is the settings of 0x10[5:4] and 0x12.

Setting of Lch/Rch independent volume

Default = 0x0 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x10[7]	0x0	Lch/Rch common volume setting
	0x1	Lch/Rch independent volume setting

Setting of volume (Setting of Rch volume, It is enable only to set an independent volume.)

Default = 0xFF *Blue square means initial value.

Select Address	Explanatio	on of opera
0x12 [7:0]	Value	Gain
	0x00	+24dB
	0x01	+23.5dB
		:
	0x30	0dB
	0x31	-0.5dB
	0x32	-1dB
		:
	0xFE	-103dB
	0xFF	-00

Setting of fine volume

This command becomes effective by sending the following command after setting. When using this command, it is possible to set a volume in 0.125dB carving.

Setting of fine volume (Setting of Rch fine volume, It is enable only to set an independent volume.)

Select Address	Value	Explanation of operation
0x10 [5:4]	0x0	0dB
	0x1	-0.125dB
	0x2	-0.25dB
	0x3	-0.375dB

[Note2]

It is possible to use with the 0.5-dB step in changing only 0x12[7:0] when 0x10[5:4]=0.

[Note3]

It is possible to use with the 0.125-dB step in setting both 0x10[1:0] and 0x11[7:0].

In case of 0x10[1:0] = 0x0, it becomes the set value of 0x11[7:0].

In case of 0x10[1:0] = 0x1, it becomes the -0.125dB set value of 0x11[7:0].

In case of 0x10[1:0] = 0x2, it becomes the -0.25dB set value of 0x11[7:0].

In case of 0x10[1:0] = 0x3, it becomes the -0.375dB set value of 0x11[7:0].

Because it is fixed by the transfer of 0x11 in any case, the soft transfer can be beforehand begun in the set value for the direct following of the purpose in setting 0x11 after setting in 0x10.

0x10[5:4] is the same function as 0x10[1:0], 0x12 is the same function as 0x11 when Lch/Rch independently set for Rch.





1-6. 3 band DRC

This DRC is used in order to prevent clip output of a large audio signal. There are three kinds of DRC (DRC1, DRC2, and DRC3), and non clip output can be achieved at each band. DRC1, DRC2 and DRC3 can set up two threshold value levels. Moreover, it is possible to also change slope.

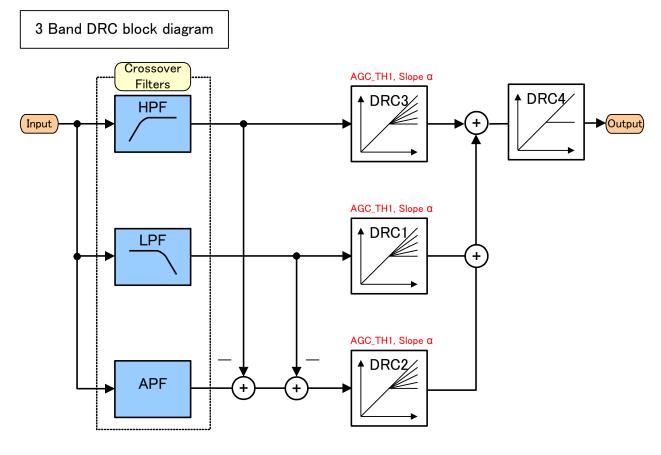
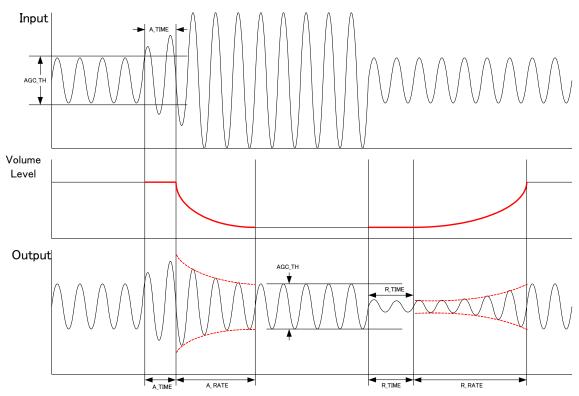


Figure 57.

DRC transition figure



In here A_TIME is the time for detecting time before starting gain down operation. And A_RATE decides the slope of gain compression.

On the other hand R_TIME is the time for detecting before starting to release gain operation. And R_RATE decides slope of release gain.

Figure 58.

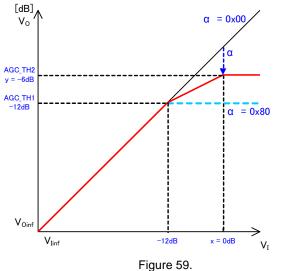
DRC1, DRC2, DRC3 can be set AGC_TH1 and AGC_TH2 as shown in below. If output is in between AGC_TH1 and AGC_TH2, a slant for output gain can be made. If input become bigger and output over AGC_TH2, output gain don't have slant and become constant level. Value for slant setting (α) is calculated by AGC_TH1, AGC_TH2 and the value of input gain for reaching AGC_TH2 (xdB).

The operation between AGC_TH1 and AGC_TH2 is decided as DRC1_{slope} and DRC2_{slope} and DRC3_{slope}.

And the operation over AGC_TH2 is decided as DRC1_{comp} and DRC2_{comp} and DRC3_{comp}.

Each operation can be set ON/OFF, A_TIME, A_RATE, R_TIME, R_RATE respectively.

DRC input-and-output gain characteristics



The formula which asks for Slope alpha is described below.

Alpha changes into 8bit Hex data of the complement of 2 the value calculated by calculation. v

$$\alpha = \frac{10^{\frac{1}{20}} - 10^{\frac{1}{20}}}{10^{\frac{1}{20}} - 10^{\frac{1}{20}}} \times 128$$

х

TH is AGC_TH1. x is input level. y is output level.

Ex) It asks for alpha at the time of AGC_TH1 = -12dB, x = 0dB y = -6dB

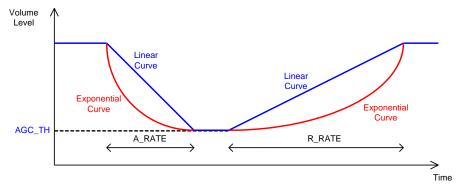
$$\frac{10^{-6}}{10^{20}} - \frac{0}{10^{20}}$$

$$\alpha = \frac{10^{-3} - 10^{-3}}{10^{-12} - 10^{-20}} \times 128$$

$$\alpha = 85.266 \rightarrow 0 \times 55$$

0x55 calculated is set as 0x29, 0x31 or 0x39.

Volume Curve





DRC1_{slope} ON/OFF setting

OFF is through output.

Default = 0x1 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x20 [7]	0x0	Not use
	0x1	Use

DRC1_{comp} ON/OFF setting

OFF is through output.

Default = 0x1 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x20 [6]	0x0	Not use
	0x1	Use

DRC2_{slope} ON/OFF setting

OFF is through output.

Default = 0x1 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x20 [5]	0x0	Not use
	0x1	Use

DRC2_{comp} ON/OFF setting

OFF is through output.

Default = 0x1 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x20 [4]	0x0	Not use
	0x1	Use

DRC3_{slope} ON/OFF setting

OFF is through output.

Default = 0x1 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x20 [3]	0x0	Not use
	0x1	Use

DRC3_{comp} ON/OFF setting of compressor

OFF is through output.

Default = 0x1 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x20 [2]	0x0	Not use
	0x1	Use

DRC4 ON/OFF setting

DRC4 have only compression function. DRC4 don't have slope function. OFF is through output.

Default = 0x0 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x3F [4]	0x0	Not use
	0x1	Use

The volume curve at the time of an attack (A_RATE) is selected.

Default = 0x1 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x21[7]	0x0	Linear curve
	0x1	Exponential curve

The volume curve at the time of a release (R_RATE) is selected.

Default = 0x1 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x21[6]	0x0	Linear curve
	0x1	Exponential curve

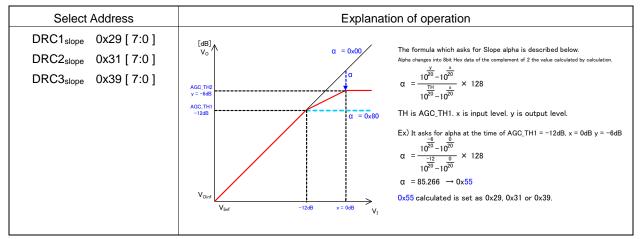
DRC is 1 Band setting at initial state.

To set the crossover filter (HPF, LPF and APF) which divides the frequency band of 3 Band DRC, therefore, it is referred to the 1-4.

Slope (a) setting of DRC1_{slope}, DRC2_{slope}, and DRC3_{slope}

DRC1_{slope}, DRC2_{slope} and DRC3_{slope} can be set individually.

Default = 0x80 *Blue square means initial value.



AGC_TH1 setting of DRC1_{slope}, DRC2_{slope}, and DRC3_{slope}

DRC1_{slope}, DRC2_{slope} and DRC3_{slope} can be set individually. Please set this value is smaller than the value of AGC_TH2.

Default = 0x40 *Blue square means initial value.

Select Address	Explanation of operation			
DRC1 _{slope} 0x28 [6:0]	Value Threshold			
DRC2 _{slope} 0x30 [6:0]	0x00	-32dB		
DRC3 _{slope} 0x38 [6:0]	:	:		
	0x3F	-0.5dB		
	- 0x40	0dB		
	- 0x41	+0.5dB		
	: · · · · · · · · · · · · · · · · · · ·	: · · · ·		
	0x58	+12dB		

AGC_TH2 setting of DRC1_comp, DRC2_comp, DRC3_comp, and DRC4

 $\mathsf{DRC1}_{\mathsf{comp}}, \mathsf{DRC2}_{\mathsf{comp}}, \mathsf{DRC3}_{\mathsf{comp}} \text{ and } \mathsf{DRC4} \text{ can be set individually}.$

Default = 0x40	*Blue squar	e means initial value.
Delault = 0.40	Dive Squar	

Select Address	Explanation of operation				
DRC1 _{comp} 0x2C [6:0]	Value	Threshold			
DRC2 _{comp} 0x34 [6:0]	0x00	-32dB			
DRC3 _{comp} 0x3C [6:0]		:			
DRC4 0x40 [6:0]	0x3F	-0.5dB			
	- 0x40	0dB			
	- 0x41	+0.5dB			
		÷			
	0x58	+12dB			

A_RATE setting of DRC1_{slope}, DRC2_{slope}, DRC3_{slope}, DRC1_{comp}, DRC2_{comp}, DRC3_{comp}, DRC4 DRC1_{slope}, DRC2_{slope}, DRC3_{slope}, DRC1_{comp}, DRC2_{comp}, DRC3_{comp}, DRC4 can be set individually.

Default = 0x3 *Blue square means initial value.

Select Address		Explanation of	of operation	
DRC1 _{slope} 0x2A [6:4] DRC2 _{slope} 0x32 [6:4]	Value	A_RATE	Value	A_RATE
DRC2 _{slope} 0x32 [6:4]	0x0	1ms	0x4	5ms
DRC1 _{comp} 0x2E [6:4]	0x1 0x2	2ms 3ms	0x5 0x6	10ms 20ms
DRC2 _{comp} 0x36 [6:4]	0x3	4ms	0x7	40ms
DRC3 _{comp} 0x3D [6:4] DRC4 0x41 [6:4]				

R_RATE setting of DRC1_{slope}, DRC2_{slope}, DRC3_{slope}, DRC1_{comp}, DRC2_{comp}, DRC3_{comp}, DRC4 DRC1_{slope}, DRC2_{slope}, DRC3_{slope}, DRC1_{comp}, DRC3_{comp}, DRC4 can be set individually.

Default = 0xB *Blue square means initial value.

Select Address		Explanation of	operation	
DRC1 _{slope} 0x2A [3:0]	Value	R_RATE	Value	R_RATE
DRC2 _{slope} 0x32 [3:0]	0x0	0.125s	0x8	2s
DRC3 _{slope} 0x3A [3:0]	0x1	0.1825s	0x9	2.5s
DRC1 _{comp} 0x2E [3:0]	0x2	0.25s	0xA	3s
DRC2 _{comp} 0x36 [3:0]	0x3	0.5s	0xB	4s
DRC3 _{comp} 0x3D [3:0]	0x4	0.75s	0xC	5s
DRC4 0x41 [3:0]	0x5	1s	0xD	6s
	0x6	1.25s	0xE	7s
	0x7	1.5s	0xF	8s

A_TIME setting of DRC1_{slope}, DRC2_{slope}, DRC3_{slope}, DRC1_{comp}, DRC3_{comp}, DRC4

 $\mathsf{DRC1}_{\mathsf{slope}}, \mathsf{DRC2}_{\mathsf{slope}}, \mathsf{DRC3}_{\mathsf{slope}}, \mathsf{DRC1}_{\mathsf{comp}}, \mathsf{DRC2}_{\mathsf{comp}}, \mathsf{DRC3}_{\mathsf{comp}}, \mathsf{DRC4} \text{ can be set individually}.$

Default = 0x1 *Blue square means initial value.

Select Address	E	Explanation of o	operation	
DRC1 _{slope} 0x2B [7:4]	Value	A_TIME	Value	A_TIME
DRC2 _{slope} 0x33 [7:4]	0x0	0ms	0x8	6ms
DRC3 _{slope} 0x3B [7:4]	0x1	0.5ms	0x9	7ms
DRC1 _{comp} 0x2F [7:4]	0x2	1ms	0xA	8ms
DRC2 _{comp} 0x37 [7:4]	0x3	1.5ms	0xB	9ms
DRC3 _{comp} 0x3E [7:4]	0x4	2ms	0xC	10ms
DRC4 0x42 [7:4]	0x5	3ms	0xD	20ms
	0x6	4ms	0xE	30ms
	0x7	5ms	0xF	40ms

R_TIME setting of DRC1, DRC2, DRC3, and DRC4 DRC1, DRC2, DRC3 and DRC4 are individually setting.

Default = 0x3 *Blue sq	uare means initial value.
Boldan - one Blac og	

Select Address		Explanation	of operation	
DRC1 _{slope} 0x2B [2:0]				
DRC2 _{slope} 0x33 [2:0]	Value	R_TIME	Value	R_TIME
DRC3 _{slope} 0x3B [2:0]	0x0	5ms	0x4	100ms
DRC1 _{comp} 0x2F [2:0]	0x1	10ms	0x5	200ms
DRC2 _{comp} 0x37 [2:0]	0x2	25ms	0x6	300ms
	0x3	50ms	0x7	400ms
DRC3 _{comp} 0x3E [2:0]				
DRC4 0x42 [2:0]				

1-7. Post-Scaler

Post-Scaler adjusts gain of output audio data. The adjustable-range can be set from +48 dB to -79 dB with the 0.5-dB step. (Lch/Rch use same value)

Pre-Scaler doesn't have a soft transfer feature.

Select Address	Explanation of operation			
0x13 [7:0]		Value	Gain	
		0x00	+48dB	
		0x01	+47.5dB	
		:	÷	
		0x60	0dB	
		0x61	-0.5dB	
		0x62	-1dB	
		:	:	
		0xFE	-79dB	
		0xFF	-∞	
			I	

1-8. Fine Post-Scaler

An adjustable range can be set up at a 0.1dB step from +0.7dB to -0.8dB. Fine Post-Scaler does not have a smooth transition function.

(Lch and Rch can be set different value.)

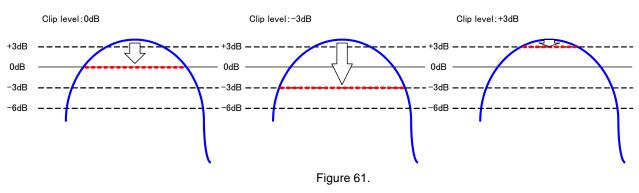
Default= 0x8 *Blue square means initial value.
--

Select Address	Explanation of operation				
Lch 0x14 [7:4]	Command	Gain	Command	Gain	
Rch 0x14 [3:0]	0x0	-0.8dB	0x8	0dB	
	0x1	-0.7dB	0x9	+0.1dB	
	0x2	-0.6dB	0xA	+0.2dB	
	0x3	-0.5dB	0xB	+0.3dB	
	0x4	-0.4dB	0xC	+0.4dB	
	0x5	-0.3dB	0xD	+0.5dB	
	0x6	-0.2dB	0xE	+0.6dB	
	0x7	-0.1dB	0xF	+0.7dB	

1-9. Hard Clipper

Measuring output of the television, output power is measured in 10% THD+N condition. It can be made to clip with any output amplitude by using a clipper function. For example, the output of 10W or 5W can be gained using the amplifier of 15W output.

Hard clip



Clipper setting

Default = 0x1 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x1A[0]	0x0	Clipper function is not used.
	0x1	Hard clipper function is used.

Clip level selection

Default = 0xE1 *Blue square means initial value.

0x1B [7:0] Value	Gain
0x00	-22.5dB
0x01	-22.4dB
	:
0xE0	-0.1dB
0xE1	0dB
0xE2	+0.1dB
	:
0xFE	+2.9dB
0xFF	+3dB

1-10. DC Cut HPF

DC offset element of the digital signal outputted from audio DSP is cut by this HPF. The cutoff frequency fc of HPF uses the 1Hz filter, and the degree uses the first-order filter.

Default = 0x1 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x18[0]	0x0	Not use
	0x1	Use

1-11. RAM clear

The data RAM of DSP and coefficient RAM are cleared. $40\mu s$ or more is required until all the data is cleared.

Clear of the data RAM

Default = 0x1 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x01 [7]	0x0	Normal
	0x1	Clear

Clear of coefficient RAM

Default = 0x1 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x01 [6]	0x0	Normal
	0x1	Clear

1-12. Audio Output Level Meter

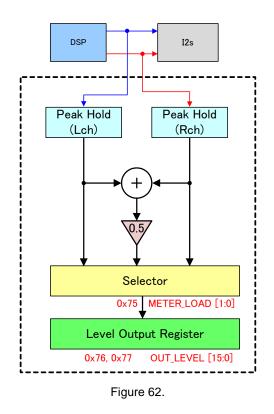
It is possible to output the peak level of the PCM data inputted into a PWM processor.

A peak value can be read using the 2-wire command interface as 16 bit data of an absolute value.

The interval holding a peak value can be selected from six steps (50ms step) from 50ms to 300ms.

A peak hold result can be selected from L channel, R channel, and a monophonic channel {(Lch+Rch) /2}.

Audio Output Level Meter block diagram



Setting of the peak level hold time interval of Audio Output Level Meter

Default = 0x0 *Blue square means initial value.

Select Address	Explanation of operation			
0x74 [2:0]	Value	Hold time		
	0x0	50ms		
	0x1	100ms		
	0x2	150ms		
	0x3	200ms		
	0x4	250ms		
	0x5	300ms		

The signal of Audio Level Meter read-back is selected.

A value will be taken into a read-only register if a setting value is written in.

In order to update this register value, it is necessary to write in a setting value again.

Write only

Select Address	Value	Explanation of operation
0x75 [1:0]	0x0	The peak level of L channel
	0x1	The peak level of R channel
	0x2	The peak level of monophonic channel {(Lch+Rch) /2}

Read-back of Audio Output Level

0x76 (upper 8 bits) and a 0x77 (lower 8 bits) commands are read for the maximum within the period appointed by the command 0x74 using the 2-wire interface.

(Example)

When 0xFFFF is read, mean 1.0 (0dBFs). When 0x8000 is read, mean 0.5 (-6dBFs).

2. Setting and reading method of BQ

It explains a detailed sequence of the setting method and the reading method of the parametric equalizer separately for usage.

2-1 BQ coefficient setting

The parametric equalizer consists of biquad filter as follows. Each coefficient of biquad filter can be written directly. It is S2.21 format, and setting range is $-4 \le x < +4$.

Moreover, the coefficient address is shown in Table 1.

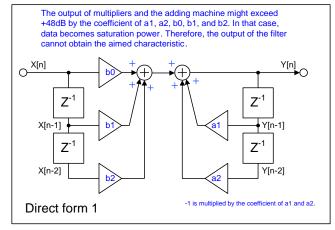


Figure 63.

2-1-1 Writing sequence (It sets up in number order)

- 1. Address setting (0x61) (*1)Table 1 is referred to.
- 2. 24bit coefficient upper [23:16] bit setting (0x62[7:0])
- 3. 24bit coefficient middle [15:8] bit setting (0x63[7:0])
- 4. 24bit coefficient lower [7:0] bit setting (0x64[7:0])
- 5. The writing of coefficients is performed. (0x65[0] = 0x1) *

* After completion of writing coefficients this register is cleared automatically. It is not necessary to transmit 0x65[0] = 0x0. Coefficient writing takes about 100μ s. 100μ s should not change an address setup and several 24-bit setup after coefficient write-in execution.

(ex) When 0x3DEDE7 is written, same L/Rch, 12band BQ1 b0

- 1. 0x61 = 0x00 (12band BQ1 b0 is appointed)
- 2. 0x62 = 0x3D (Upper [23:16] is setting)
- 3. 0x63 = 0xED (Middle [15:8] is setting)
- 4. 0x64 = 0xE7 (Lower [7:0] is setting)
- 5. 0x65 = 0x01 (Coefficient transfer)*

*After completion of writing coefficients this register is cleared automatically.

6. 100µs or more wait

2-1-2 Read-back sequence (It sets up in number order)

- 1. Address setting (0x61) . Please refer to Table 1.
- 2. Setting of a read-back register address (0xD0)
- 3. Read-back of the 24bit coefficient upper[23:16]bit (0x66[7:0])
- 4. Read-back of the 24bit coefficient middle[15:8]bit (0x67[7:0])
- 5. Read-back of the 24bit coefficient lower[7:0]bit (0x68[7:0])

2-1-3 When the coefficient of PEQ is set up directly and a soft transition is performed

- Set PEQ coefficient to soft transition address whose address is 0x50-0x54. Please refer to Table1. Since in the case of 0x60[4] = 0x1(Enable L/R independent setting) and 0x53 [5:4] = 0x0 a soft transition is carried out and it is set to LR simultaneously, please write a coefficient in both LR address. In the case of 0x53[5:4] = 0x1, coefficient is set to only Lch address. In the case of 0x53[5:4] = 0x2, coefficient is set to only Rch address.
- 2. Select PEQ Band that is performed soft transition by setting 0x51[4:0] address.(Refer to chapter 1-4)
- 3. 0x58[0] = 0x1 : Start soft transition (After the completion of soft transition this register is automatically cleared.)
- 4. Wait soft transition completion (It depend on soft transition time setting), or read command 0x59 [0], and stand by until it is cleared.

0x61[6:0]	Destination	0x61[6:0]	Destination	0x61[6:0]	Destination
0x00	12BandBQ1 b0	0x23	12BandBQ8 b0	0x46	DRC2_1 b0
0x01	12BandBQ1 b1	0x24	12BandBQ8 b1	0x47	DRC2_1 b1
0x02	12BandBQ1 b2	0x25	12BandBQ8 b2	0x48	DRC2_1 b2
0x03	12BandBQ1 a1	0x26	12BandBQ8 a1	0x49	DRC2_1 a1
0x04	12BandBQ1 a2	0x27	12BandBQ8 a2	0x4A	DRC2_1 a2
0x05	12BandBQ2 b0	0x28	12BandBQ9 b0	0x4B	DRC2_2 b0
0x06	12BandBQ2 b1	0x29	12BandBQ9 b1	0x4C	DRC2_2 b1
0x07	12BandBQ2 b2	0x2A	12BandBQ9 b2	0x4D	DRC2_2 b2
0x08	12BandBQ2 a1	0x2B	12BandBQ9 a1	0x4E	DRC2_2 a1
0x09	12BandBQ2 a2	0x2C	12BandBQ9 a2	0x4F	DRC2_2 a2
0x0A	12BandBQ3 b0	0x2D	12BandBQ10 b0	0x50	Smooth BQ b0
0x0B	12BandBQ3 b1	0x2E	12BandBQ10 b1	0x51	Smooth BQ b1
0x0C	12BandBQ3 b2	0x2F	12BandBQ10 b2	0x52	Smooth BQ b2
0x0D	12BandBQ3 a1	0x30	12BandBQ10 a1	0x53	Smooth BQ a1
0x0E	12BandBQ3 a2	0x31	12BandBQ10 a2	0x54	Smooth BQ a2
0x0F	12BandBQ4 b0	0x32	12BandBQ11 b0	0x55	DRC3_1 b0
0x10	12BandBQ4 b1	0x33	12BandBQ11 b1	0x56	DRC3_1 b1
0x11	12BandBQ4 b2	0x34	12BandBQ11 b2	0x57	DRC3_1 b2
0x12	12BandBQ4 a1	0x35	12BandBQ11 a1	0x58	DRC3_1 a1
0x13	12BandBQ4 a2	0x36	12BandBQ11 a2	0x59	DRC3_1 a2
0x14	12BandBQ5 b0	0x37	12BandBQ12 b0	0x5A	DRC3_2 b0
0x15	12BandBQ5 b1	0x38	12BandBQ12 b1	0x5B	DRC3_2 b1
0x16	12BandBQ5 b2	0x39	12BandBQ12 b2	0x5C	DRC3_2 b2
0x17	12BandBQ5 a1	0x3A	12BandBQ12 a1	0x5D	DRC3_2 a1
0x18	12BandBQ5 a2	0x3B	12BandBQ12 a2	0x5E	DRC3_2 a2
0x19	12BandBQ6 b0	0x3C	DRC1_1 b0		
0x1A	12BandBQ6 b1	0x3D	DRC1_1 b1		
0x1B	12BandBQ6 b2	0x3E	DRC1_1 b2		
0x1C	12BandBQ6 a1	0x3F	DRC1_1 a1		
0x1D	12BandBQ6 a2	0x40	DRC1_1 a2		
0x1E	12BandBQ7 b0	0x41	DRC1_2 b0		
0x1F	12BandBQ7 b1	0x42	DRC1_2 b1		
0x20	12BandBQ7 b2	0x43	DRC1_2 b2		
0x21	12BandBQ7 a1	0x44	DRC1_2 a1		
0x22	12BandBQ7 a2	0x45	DRC1_2 a2		

Table1. Specified coefficient

When L/R independent, Lch:0x61[7]=0x0, Rch:0x61[7]=0x1. When L/R is same, 0x61[7] is not effective.

3. The mute function by a terminal

BM28720MUV has a mute function of audio DSP by a terminal. It is possible to perform mute of the output from Audio DSP by setting a MUTEX terminal to low.

Transition time setting at the time of mute is as follows.

Smooth transition mute time setting

The transition time when changing to a mute state is selected.

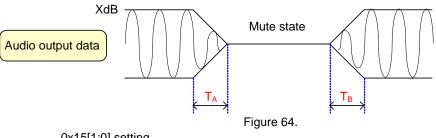
The soft transition time at the time of mute release is 10.7ms fixed.

Default = 0x3 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x15 [1:0]	0x0	10.7ms
	0x1	21.4ms
	0x2	42.7ms
	0x3	85.4ms

0x15[1:0] Mute time setting

It is only operated by mute terminal.



0x15[1:0] setting

	0	
Command	T _A	T _Β
0x0	10.7ms	10.7ms
0x1	21.4ms	10.7ms
0x2	42.7ms	10.7ms
0x3	85.4ms	10.7ms

Smooth transition mute release time setting Time after detecting mute release until it actually begins mute release operation is set up.

Default = 0x0 *Blue square means initial value	
--	--

Select Address	Value	Explanation of operation
0x15 [5:4]	0x0	0ms
	0x1	100ms
	0x2	200ms
	0x3	300ms

Operation of mute delay 0x15[5:4]

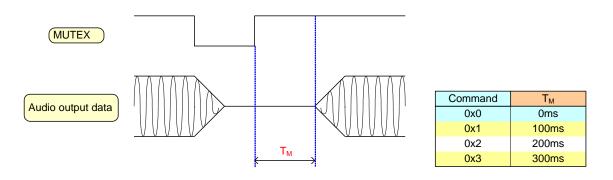
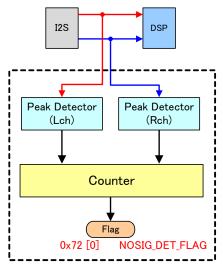


Figure 65.

4. Small signal input detection function

There is a function which detects the audio data input of a non-signal or a small signal. This function is used in order to reduce the standby power consumption of an audio set. Setting of a detection level and detection time can be performed. If the signal below a setting detection level continues in both low channel and R channel, a small signal detection flag will become "high". A detection result can be read from command 0x72 [0].

The point which acts as a monitor of the small signal becomes input data of audio DSP block.





Detection level setting

Default = 0x00 *Blue square means initial value.

Select Address	Explanation of operation						
0x70 [4:0]	Value	Level	Value	Level	Value	Level	
	0x00	-103dB	0x08	-77dB	0x10	-69dB	
	0x01	-93dB	0x09	-76dB	0x11	-68dB	
	0x02	-91dB	0x0A	-75dB	0x12	-67dB	
	0x03	-87dB	0x0B	-74dB	0x13	-66dB	
	0x04	-84dB	0x0C	-73dB	0x14	-65dB	
	0x05	-80dB	0x0D	-72dB	0x15	-64dB	
	0x06	-79dB	0x0E	-71dB	0x16	-62dB	
	0x07	-78dB	0x0F	-70dB	0x17	-60dB	

Detection time setting

Default = 0x0 *Blue square means initial value.

Select Address	Value	Explanation of operation
0x71 [1:0]	0x0	42.7ms
	0x1	85.4ms
	0x2	170.7ms
	0x3	341.4ms

* Sampling frequency is value of Fs = 48kHz. In the case of Fs = 44.1kHz, it will be about 1.09 times the setting value.

Detection flag read-back

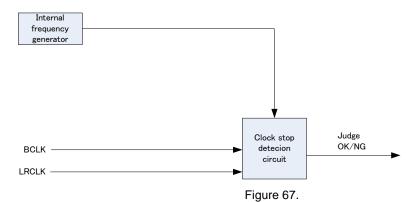
Read Only

Select Address	Value	Explanation of operation
0x72 [0]	0x0	Undetected
	0x1	Detecting

5. Clock stop detection and detection of BCLK frequency begin too low or too high or asynchronous state detection

5-1. Clock stops detection

BM28720MUV needs some clock source for generating proper clock to process Audio data. By stopping these cock sources, these clocks to process Audio data also stop. To prevent noise sounds, we need to detect BCLK or LRCLK stop condition. As we detect stop flag that is to be valid, output is gone to mute state (mute instantly).



Each detect condition is set by below command. We can check detected result by reading back flag register. These flags are cleared only by sending specified commands.

LRCLK stop detection time

Default = 0x2 *Blue square means initial value.

Select Address	Value	Operation
LRCLK 0x07 [2:0]	0x0	10µs to 20µs
	0x1	20µs to 40µs
	0x2	50µs to 100µs
	0x3	100µs to 200µs
	0x4	200µs to 400µs
	0x5	300µs to 600µs
	0x6	400µs to 800µs
	0x7	500µs to 1000µs

*Detection time has the above-mentioned variation within the limits.

BCLK stop detection time

Default = 0x0 *Blue square means initial value.

Select Address	Value	Operation
BCLK 0x08 [6:4]	0x0	10µs to 20µs
	0x1	20µs to 40µs
	0x2	50µs to 100µs
	0x3	100µs to 200µs
	0x4	200µs to 400µs
	0x5	300µs to 600µs
	0x6	400µs to 800µs
	0x7	500µs to 1000µs

*Detection time has the above-mentioned variation within the limits.

Stop detection flag read back register

Read Only

Select Address	Value	Operation
0x09 [5]	0x0	Normal
	0x1	Detection of LRCLK stop flag
0x09 [4]	0x0	Normal
	0x1	Detection of BCLK stop flag

Stop detection flag clear register

Write Only

Select Address	Operation
0x09 [1]	LRCLK stop detection flag is cleared by writing 1.
0x09 [0]	BCLK stop detection flag is cleared by writing 1.

% When using a clock error auto return function (P.59), this flag is cleared automatically.

LRCLK stop flag valid or invalid selection

Default = 0x1 *Blue square means initial value.

Select Address	Value	Operation
0x07 [3]	0x0	Valid
	0x1	Invalid

BCLK stop flag valid or invalid selection

Default = 0x0 *Blue square means initial value.

Select Address	Value	Operation
0x08 [7]	0x0	Valid
	0x1	Invalid

5-2. Synchronous blank detection

As for synchronous blank detecting function, it detects as synchronous blank error when it counts between the rising edges of LRCK with internal clock (49.152MHz), and it shifts more than the definite value, and whether PLL is normally locked is judged.

Count value 1023 (Start of counting from 0)	Input sampling frequency	32kHz,44.1kHz,48kHz
		1023

As for the detection result, reading from the register is possible. Register value is not cleared until a clear command is transmitted even if the state of the clock returns normally. Moreover, the setting of the detection approval frequency is also possible, and if the error counts more than the predetermined number is detected, the flag (0x06[1]) becomes 1.

Synchronous blank flag reading register

Read Only

Select Address	Value	Explanation of operation
0x06 [1]	0x0	Normal
	0x1	Detect synchronous error

Synchronous blank flag clear register

Write Only

Select Address	Explanation of operation		
0x06 [0]	When 1 is written, the synchronous blank flag is cleared.		
* When using a clock error auto return function (P59), this flag is cleared automatically			

* When using a clock error auto return function (P.59), this flag is cleared automatically.

Synchronous blank count setting

Default = 0x2 *Blue square means initial value.

Select Address	Explanation of operation
0x06 [6:4]	Please set more than 1.

5-3. BCLK high or low speed detection

BCLK high or low speed detection function is to judge BCLK speed being too high or low by measuring by using internal clock (12MHz to 25MHz).

When using BCLK speed detection, speed failure detection can be more correctly performed by setting a command according to inputted sample rate.

When you use error detection function, please set up the sample rate by setting address 0x0C[1:0]. A high speed and the low-speed detection flag can be set up valid or invalid respectively, and if the error is detected, mute (mute instantly) will be carried out.

Valid or invalid frequency value setting up by 0x0C[1:0] command.

Default = 0x0 *Blue square means initial value.

Select Address	Value	Operation
0x0A [3]	0x0	Valid
	0x1	Invalid

Setting of sampling rate

Default = 0x0 *Blue square means initial value.

Select Address	Value	Operation
0x0C [1:0]	0x0	48kHz
	0x1	44.1kHz
	0x2	32kHz

The constraints of a high speed or a low-speed condition

Default = 0x0 *Blue square means initial value.

Select Address	Value	Operation
0x0A [2]	0x0	±10%

We can check detection result by reading back.

The result judged that is once unusual is not cleared until it transmits a clear command, even if the condition of a clock returns to normal. We can set up

We can set up the constraints of the count of formation, and it does not set a flag until it detects it by count continuation.

BCLK high speed flag

Read Only

Select Address	Value	Operation
0x0A [1]	0x0	Normal
	0x1	High speed detection flag

BCLK low speed flag

Read Only

Select Address	Value	Operation
0x0B [1]	0x0	Normal
	0x1	Low speed detection flag

High speed detection clear register

Write Only

Select Address	Operation			
0x0A [0]	If 1 is written in, a high speed detection flag will be cleared.			
When using a clock error auto return function (P.59), this flag is cleared automatically.				

Low speed detection clear register

Write Only

Select Address	Operation		
0x0B [0]	If 1 is written in, a high speed detection flag will be cleared.		
When using a clock error auto return function (P.59), this flag is cleared automatically.			

A constraint of the count of judging with high speed flag detection

Default = 0x2 *Blue square means initial value.

Select Address	Operation
0x0A [6:4]	Please set more than 1.

A constraint of the count of judging with low speed flag detection

Default = 0x2 *Blue square means initial value.

Select Address	Operation
0x0B [6:4]	Please set more than 1.

High speed detection flag valid or invalid

Select Address	Value	Operation
0x0A [7]	0x0	Valid
	0x1	Invalid

Low speed detection flag valid or invalid

Default = 0x0 *Blue square means initial value.

Select Address	Value	Operation
0x0B [7]	0x0	Valid
	0x1	Invalid

The frequency range of BCLK by which high speed detection or low speed detection is carried out becomes below.

Setting1	Settin2	Low detection lowest frequency (MHz)	High detection highest frequency (MHz)
	48kHz(0x0C[1:0]=0x0)	1.28	7.13
64fs BCLK(0x03[5:4]=0x0)	44.1kHz(0x0C[1:0]=0x1)	1.21	6.55
	32kHz(0x0C[1:0]=0x2)	0.88	4.76
	48kHz(0x0C[1:0]=0x0)	0.96	5.35
48fs BCLK(0x03[5:4]=0x1)	44.1kHz(0x0C[1:0]=0x1)	0.91	4.92
	32kHz(0x0C[1:0]=0x2)	0.66	3.57
	48kHz(0x0C[1:0]=0x0)	0.64	3.56
32fs BCLK(0x03[5:4]=0x2)	44.1kHz(0x0C[1:0]=0x1)	0.60	3.28
	32kHz(0x0C[1:0]=0x2)	0.44	2.38

6. Auto recovery from clock error function

If clock error is detected, immediately output is muted.

In that case, if the clock error auto return function is enabled, when it returns to a normal input, a mute condition will be canceled automatically.

If the clock error auto return function is OFF, it is necessary to clear these error flag manually.

Since it is invalid at initial state, we recommend to send 0x0D[6] = 0x1 and to enable auto recovery function.

Valid or invalid auto recover from clock error

Default = 0x0 *Blue square means initial value.

Select Address	Value	Operation
0x0D [6]	0x0	Invalid
	0x1	Valid

Each error flag can be read from the following addresses. When 1 is read from a read address, the error flag stands. Error flag is not cleared automatically until this register is cleared manually.

Error flag read register

Select Address	Operation	
0x0E [6]	Asynchronous flag	
0x0E [4]	LRCLK stop flag	
0x0E [3]	BCLK stop flag	
0x0E [2]	BCLK high speed detection flag	
0x0E [1]	BCLK low speed detection flag	

7. The wake-up Procedure of power-up

It recommends starting power-up in the following Procedures.

1. Power up

Wait over 10ms

2. Release reset(RSTX=H)

Wait over 1ms

3. 0x0C[1:0]=0x* : Sampling rate(Please set up 0x0 in the case of 48kHz, set up 0x1 in the case of 44.1kHz and 0x02 in 32kHz.)

Please input BCLK and LRCLK

4. 0xE9=0x10 : changing clock to normal state

Wait over 100ms

- 5. 0x01=0x00 : Set RAM clear OFF
- 6. 0x0D=0x40 : Valid auto recover from clock error
- 7. 0x0E=0x00 : Clear error flag
- 8. 0x92=0x11 : PWM setting1
- 9. 0x93=0x1C : PWM setting2
- 10.0x94=0x15 : PWM setting3
- 11.0x95=0x04 : PWM setting4

12. Please set up DSP function such as volume, PEQ, DRC, and Scaler etc.

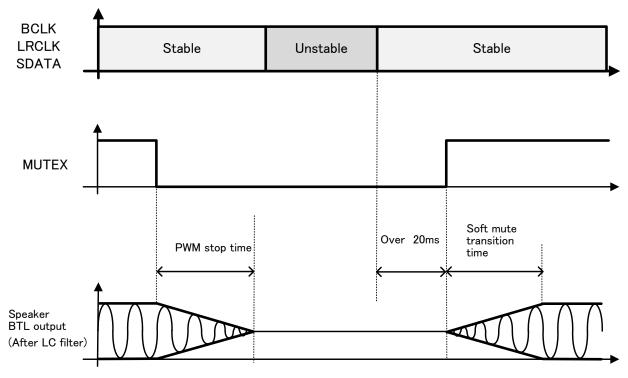
13.MUTEX=high : Release mute

* The wait time, Twait, is necessary between 2 and 13.(Please refer to P15 for Twait)

8. The operating procedure in a status with an unstable clock

In the period which I²S input may become unstable, please set to MUTEX=low and carry out mute.

- 1. Set MUTEX=low
- 2. Please wait over 20ms after stabilizing I²S input.
- 2. Set MUTEX=high



(Note 1)When clock stop was detected, mute release procedure world follow clock stop error release condition.

Figure 68.

9. I²S Data output select

Output I²S data signal From SDATAO (pin12). That SDATAO synchronize to inputted LRCK and BCK signal. And enable to select output SDATA signal as shown below. Whatever output is selected, hard clip is processed.

SDATAO output select

Default = 0x0 *Blue square means initial value.

Select Address	Value	Explanation of operation	
0x78 [6:4]	0x0	DSP output (point1)	
	0x1	DSP input (pont2)	
	0x2	Pre Scaler output (point3)	
	0x3	Mixer output (point4)	
	0x4	12 Band PEQ output (point5)	
	0x5	Fine Master Volume output (point6)	
	0x6	No use	
	0x7	Fine Post Scaler output (point7)	

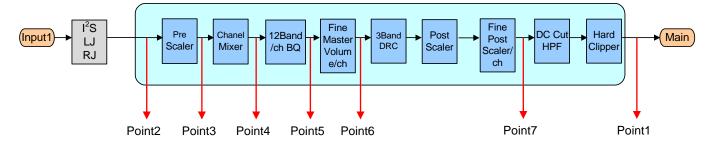


Figure 69.

Application Circuit Example1 (Stereo BTL output, R_L=8Ω, Vcc=10V to 18V)

• When using at Vcc>18V, fc of the LC filter should be lowered to about 60kHz and decrease the influence of LCR resonance using application circuit. Please refer 5) Output LC Filter Circuit (Vcc>18V)

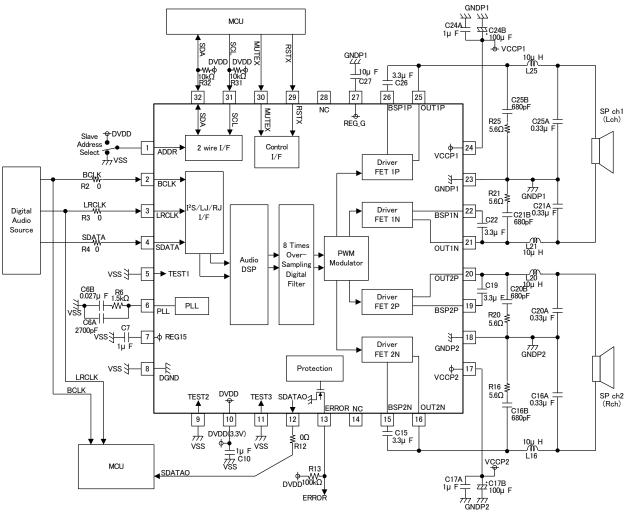


Figure 70.

BOM list1(Stereo BTL output, R_L=8Ω, Vcc=10V to 18V)

Parts	Qty	Parts No.	Description	Company	Product No.
Inductor	4	L16, L20, L21, L25	10µH / (±20%) / 7.6mm×7.6mm	ТОКО	B1047AS-100M
	4	R16, R20 R21, R25	5.6Ω / 1/10W / J(±5%) / 1.6mm×0.8mm		MCR03EZPJ5R6
	1	R6	1.5k Ω / 1/16W / F(±1%) / 1.0mm×0.5mm		MCR01MZPF1501
Resistor	2	R31, R32	10k Ω / 1/16W / J(±5%) / 1.0mm×0.5mm	ROHM	MCR01MZPJ103
	4	R2, R3, R4, R12	0Ω / 1/10W / J(±5%) / 1.6mm×0.8mm		MCR03EZPJ000
	1	R13	100k Ω / 1/16W / J(±5%) / 1.0mm×0.5mm		MCR01MZPJ104
	4	C16B, C20B C21B, C25B	680pF / 50V / B(±10%) / 1.6mm×0.8mm		GRM188B11H681KA01
	1	C6A	2700pF / 6.3V / B(±10%) / 0.6mm×0.3mm		GRM033B10J272KA01
	1	C6B	$0.027 \mu F$ / $6.3 V$ / $B(\pm 10\%)$ / $0.6 mm \times 0.3 mm$	MURATA	GRM033B10J273KE01
	4	C16A, C20A, C21A, C25A,	0.33µF / 50V / B(±10%) / 2.0mm×1.25mm		GRM219B31H334KA87
Capacitor	2	C17A, C24A (Note 1)	1µF / 50V / B(±10%) / 2.0mm×1.25mm		GRM21BB31H105KA12
	4	C15, C19, C22, C26	3.3µF / 16V / B(±10%) / 1.6mm×0.8mm		GRM188B31A335KE15
	2	C7, C10	1µF / 10V / B(±10%) / 1.6mm×0.8mm		GRM185B31A105KE25
	1	C27	10µF / 10V / B(±10%) / 1.6mm×0.8mm		GRM188B31A106KE69
	2	C17B, C24B	100μF / 35V / (±20%) / φ6.3mm×11.2mm	PANASONIC	ECA1VMH101

(Note 1) Please put the C17A and C24A near the VCCP1 and VCCP2 pins on the board.

Application Circuit Example2 (Monaural BTL output, R_L=8Ω, Vcc=10V to 18V)

• When using at Vcc>18V, fc of the LC filter should be lowered to about 60kHz and decrease the influence of LCR resonance using application circuit. Please refer 5) Output LC Filter Circuit (Vcc>18V)

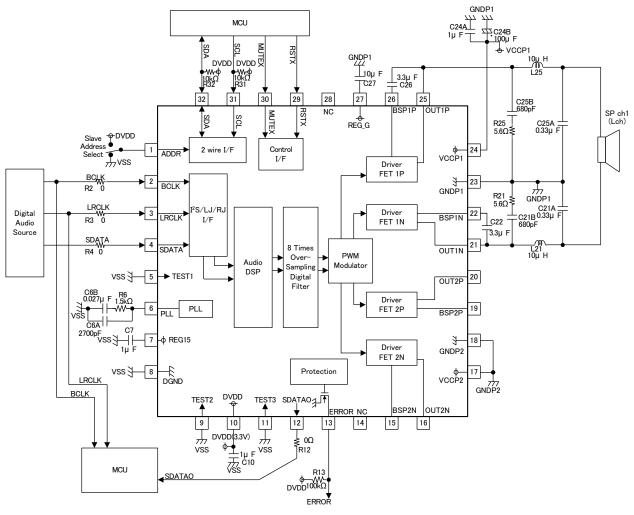


Figure 71.

BOM list2(Monaural BTL output, R_L=8Ω, Vcc=10V to 18V)

Parts	Qty	Parts No.	Description	Company	Product No.
Inductor	2	L21, L25	10µH / (±20%) / 7.6mm×7.6mm	ТОКО	B1047AS-100M
	2	R21, R25	5.6Ω / 1/10W / J(±5%) / 1.6mm×0.8mm		MCR03EZPJ5R6
	1	R6	1.5kΩ / 1/16W / F(±1%) / 1.0mm×0.5mm		MCR01MZPF1501
Resister	2	R31, R32	10kΩ / 1/16W / J(±5%) / 1.0mm×0.5mm	ROHM	MCR01MZPJ103
	4	R2, R3, R4, R12	0Ω / 1/10W / J(±5%) / 1.6mm×0.8mm		MCR03EZPJ000
	1	R13	100kΩ / 1/16W / J(±5%) / 1.0mm×0.5mm		MCR01MZPJ104
	2	C21B, C25B	680pF / 50V / B(±10%) / 1.6mm×0.8mm		GRM188B11H681KA01
	1	C6A	2700pF / 6.3V / B(±10%) / 0.6mm×0.3mm		GRM033B10J272KA01
	1	C6B	$0.027\mu F$ / $6.3V$ / $B(\pm 10\%)$ / $0.6mm \times 0.3mm$	MURATA	GRM033B10J273KE01
	2	C21A, C25A	0.33µF / 50V / B(±10%) / 2.0mm×1.25mm		GRM219B31H334KA87
Capacitor	1	C24A (Note 1)	1µF / 50V / B(±10%) / 2.0mm×1.25mm	MORATA	GRM21BB31H105KA12
	2	C22, C26	3.3µF / 16V / B(±10%) / 1.6mm×0.8mm		GRM188B31A335KE15
	2	C7, C10	1µF / 10V / B(±10%) / 1.6mm×0.8mm		GRM185B31A105KE25
	1	C27	10µF / 10V / B(±10%) / 1.6mm×0.8mm		GRM188B31A106KE69
	1	C24B	100μF / 35V / ±20% / φ6.3mm×11.2mm	PANASONIC	ECA1VMH101

(Note 1) Please put the C24A near the VCCP1 pins on the board.

Selection of Components Externally Connected (Vcc=10V to 18V)

1) Output LC Filter Circuit

An output filter is required to eliminate radio-frequency components exceeding the audio-frequency region supplied to a load (speaker). Because this IC uses sampling clock frequency 384kHz(fs=48kHz) in the output PWM signals, the high-frequency components must be appropriately removed.

This section takes an example of an LC type LPF shown below, in which coil L and capacitor C compose a differential filter with an attenuation property of -12dB/oct. A large part of switching currents flow to capacitor C, and only a small part of the currents flow to speaker R_{L1} . This filter reduces unwanted emission this way. In addition, coil L and capacitor Cg composes a filter against in-phase components, reducing unwanted emission further.

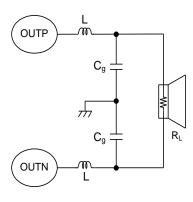


Figure 72.

Following presents output LC filter constants with typical load impedances.

RL	L	Cg
4Ω	10µH	1µF
6Ω	10µH	0.47µF
8Ω	10µH	0.33µF

Use coils with a low direct-current resistance and with a sufficient margin of allowable currents. A high direct-current resistance causes power losses. In addition, select a closed magnetic circuit type product in normal cases to prevent unwanted emission.

Use capacitors with a low equivalent series resistance, and good impedance characteristics at high frequency ranges (100kHz or higher). Also, select an item with sufficient withstand voltage because flowing massive amount of high-frequency currents is expected.

2) The value of the LC filter circuit computed equation

The output LC filter circuit of BM28720MUV is as it is shown in Figure 73. The LC filter circuit of Figure 73 is thought to substitute it like Figure 74 on the occasion of the computation of the value of the LC filter circuit.

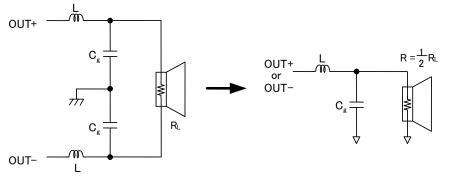


Figure 73. Output LC filter 1

Figure 74. Output LC filter 2

The transfer function H(s) of the LC filter circuit of Figure 74 becomes the following.

$$H(s) = \frac{\frac{1}{LC_g}}{s^2 + \frac{1}{C_g R}s + \frac{1}{LC_g}} = \frac{\omega^2}{s^2 + \frac{\omega}{Q}s + \omega^2}$$

The ω and Q become the followings here.

$$\omega^{2} = \frac{1}{LC_{g}} \qquad \omega = 2\pi f_{C} \qquad f_{c} = \frac{1}{2\pi\sqrt{LC_{g}}}$$
$$Q = R\sqrt{\frac{C_{g}}{L}} = \frac{1}{2}R_{L}\sqrt{\frac{C_{g}}{L}}$$

Therefore, L and C_g become the followings.

$$L = \frac{1}{\omega^2 C_g} = \frac{R_L}{4\pi f_C Q} \qquad C_g = \frac{Q}{\omega R} = \frac{Q}{\pi f_C R_L}$$

The R_L and L should be made known, and f_c is set up, and C_g is decided.

3) The settlement of the L value of the coil

For selection of the L value of a coil, please consider side effect as shown below.

①When L value was made small.

- (1) Circuit electric currents increase without a signal. And, efficiency in the low output gets bad.
- (2) Direct current resistance value is restrained small when the coil of other L value and size are made the same. Therefore, maximum output is easy to take out. And, it can be used in the low power supply voltage because DC electric current (allowable electric current) value can be taken greatly.

2When L value was made large.

- (1) Circuit electric current is restrained low without a signal. Efficiency in the low output improves.
- (2) Direct current resistance value grows big when the coil of other L value and size are made the same. Therefore, maximum output is hard to take out. And, because it becomes small, use becomes difficult [the DC electric current (allowable electric current) value] in the low power supply voltage, too.
- 4) The settlement of the f_c

As for the settlement of the fixed number of the LC filter circuit, it is taken into consideration about two points of the following, and set up.

(1) The PWM sampling frequency fPWM (=8fs) of BM28720MUV is set up in 384kHz (@fS=48kHz).

It is set up with $f_c < fPWM$ to restrain career frequency omission after the LC filter circuit. (2)When fc is lowered too much, the voltage profit of the voice obi stage (especially, the neighborhood of 20kHz) declines in the speaker output frequency character of the difference movement mode.

And, the speaker output frequency character of the difference movement mode becomes the following.

R _L =8Ω			R _L =6Ω			$R_L=4\Omega$					
L[µH]	C _g [µF]	f _c [kHz]	L[µH]	C _g [µF]	f _c [kHz]	L[µH]	C _g [µF]	f _c [kHz]	L[µH]	C _g [µF]	f _c [kHz]
	0.1	75.32	0.40		0.1	51.01	0.30		0.1	32.19	0.20
	0.15	80.85	0.49		0.15	54.76	0.37		0.15	33.35	0.24
10	0.22	86.79	0.59	10	0.22	56.73	0.44	10	0.22	34.55	0.30
10	0.33	89.92	0.73	10	0.33	63.1	0.54	10	0.33	35.8	0.36
	0.47	86.79	0.87		0.47	66.68	0.65		0.47	38.37	0.43
	1.0	69.01	1.26		1.0	62.29	0.95		1.0	44.1	0.63
	0.1 46.99 0.33		0.1	33.11	0.24		0.1	21.68	0.16		
	0.15	49.66	0.40	15	0.15	34.36	0.30	15	0.15	22.08	0.20
15	0.22	53.46	0.48		0.22	35.65	0.36		0.22	22.49	0.24
15	0.33	57.54	0.59		0.33	38.37	0.44		0.33	22.91	0.30
	0.47	59.7	0.71		0.47	41.3	0.53		0.47	23.77	0.35
	1.0	52.75 1.03		1.0	44.67	0.77		1.0	27.47	0.52	
	0.1	30.76	0.27		0.1	22.49	0.20		0.1	14.72	0.13
	0.15	31.92	0.33		0.15	22.91	0.25		0.15	14.72	0.17
22	0.22	33.73	0.40	22	0.22	23.77	0.30	22	0.22	15	0.20
22	0.33	36.31	0.49		0.33	24.66	0.37	22	0.33	15.28	0.24
	0.47	39.08	0.58		0.47	26.06	0.44		0.47	15.56	0.29
	1.0	39.30	0.85		1.0	30.05	0.64		1.0	17.33	0.43

5) Output LC Filter Circuit (Vcc>18V)

When using at Vcc>18V, fc of the LC filter should be lowered to about 60kHz or less and decrease the influence of LCR resonance using application circuit.

R_{L1}	L	С
4Ω	10µH	1µF
6Ω	15µH	0.47µF
00	15µH	0.33µF
8Ω	22µH	0.47µF

Use coils with a low direct current resistance and with a sufficient margin of allowable currents. A high direct current resistance causes power losses. In addition, select a closed magnetic circuit type product in normal cases to prevent unwanted emission.

Use capacitors with a low equivalent series resistance and good impedance characteristics at high frequency ranges (100kHz or higher). Also, select an item with sufficient voltage rating because massive amount of high frequency currents flow is expected.

In addition, please do not place the C_{BTL} shown in Figure 75.

When considering only common mode signals, OUTP and OUTN are equal to each other and this circuit is equivalent to the circuit shown in Figure 76.

Therefore, LC resonance may occur depending on the constant of a LC filter.

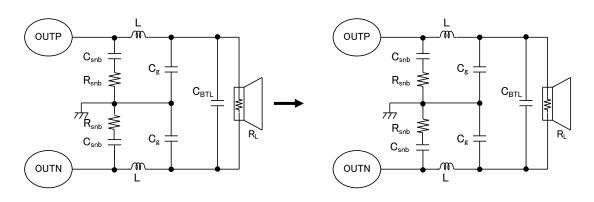


Figure 78.

Figure 79.

6) The settlement of the snubber

The snubber circuit must be optimized for application circuit to reduce the overshoot and undershoot of output PWM.

1 Measure the ringing resonance frequency f1 of the PWM output wave shape (When it stands up.) by using FET probe in the OUT terminal. (Figure 77) The FET probe is to monitor very near pin and shorten ground lead at the time of that.

② Measure resonance frequency f2 of the ringing as a snubber circuit fixed number R=0 Ω . (Only with connecting the capacitor C to GND) At this time, the value of the capacitor C is adjusted until it becomes half of the frequency (2f2=f1) of the resonance frequency f1 of ①. The value of C which it could get here is three times of the parasitic capacity C_p that a ringing is formed. (C=3C_p)

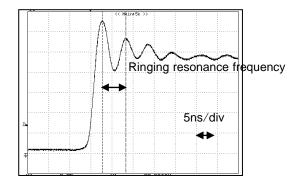
(3) Parasitic inductance L_p is looked for at the next formula.

$$\mathbf{L}_{\mathbf{p}} = \frac{1}{\left(2\pi f_{1}\right)^{2} C_{p}}$$

(4) The character impedance Z of resonance is looked for from the parasitic capacity C_p and the parasitism inductance L_p at the next formula.

$$Z = \sqrt{\frac{L_p}{C_p}}$$

(5) A snubber circuit fixed number R is set up in the value which is the same as the character impedance Z. A snubber circuit fixed number C is set up in the value of 4-10 times of the parasitic capacity C_p . (C=4C_p to 10C_p) Decide it with trade-off with the character because switching electric currents increase when the value of C is enlarged too much.



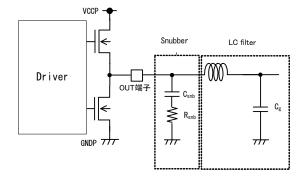


Figure 77. PWM Output waveform (Measure of ringing resonance frequency

Figure 78. Snubber schematic

Following presents snubber filter constants with the recommendation value at ROHM 4 layer board.

RL	C _{snb}	R _{snb}
4Ω	680pF ~ 1200pF, 50V B(±10%)	5.6Ω, 1/10W J(±5%)
6Ω	680pF ~ 1200pF, 50V B(±10%)	5.6Ω, 1/10W J(±5%)
8Ω	680pF ~ 1200pF, 50V B(±10%)	5.6Ω, 1/10W J(±5%)

7) Operating condition with the application component

Devementer	Darta Na	Limit			Unit	
Parameter	Parts No.	Min	Тур	Max	Min	Conditions
Conscitor for DCD	C15, C19,	2.0 ^(Note 1)	3.3	4.5 (Note 2)	μF	Recommend characteristics,
Capacitor for BSP	C22, C26	2.0 ^(Note 1)	4.7	6.3 (Note 2)	μF	16V ceramic type capacitor

(Note1) Capacitor value should not be less than a minimum in consideration of temperature characteristics and dc-bias characteristics. (Note2) It is value in consideration of +/-10% of capacity unevenness, capacity rate of change 22%. Please use the capacitor within this limit.

Level Diagram of Audio Signal

Level diagram of audio signal is shown the below figure. Speaker output level is depended on I²S digital audio input level, DSP gain, PWM gain, BTL gain and Loss of power stage and low pass filter.

I2S input level is full-scale signal, the supply voltage of the block is DVDD, and therefore, 0dBFS is equal to DVDD voltage [Vpp]. DSP gain is set by 2 wire control variably, and -0.3dB is set at PWM Modulator block usually. At the Power stage, the PWM Modulator output is shifted PWM signal level from DVDD to VCC, and added loss of the output transistor resistance r_{DS} and DC resistance of coil r_{DC} .

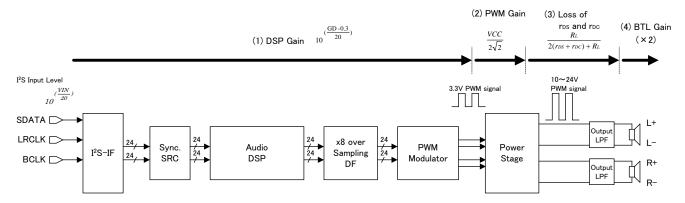


Figure 79. Level Diagram of Audio Signal

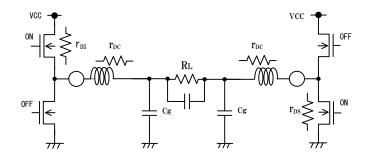


Figure 80. Output LPF circuit

In Bridge-Tied-Load (BTL) connection, the following formula gives an approximate value of output power *Po at non-*clipping output waveform:

$$P_{o} = \frac{(10^{VIN}/_{20} \times 10^{(GD-0.3)/20} \times \frac{VCC}{2\sqrt{2}} \times 2 \times \frac{R_{L}}{2 \times (r_{DS} + r_{DC}) + R_{L}})^{2}}{R_{L}}$$

$$VIN : I^{2}S \text{ Input level [dBFS]}$$

$$GD : DSP \text{ gain [dB]}$$

$$VCC : Power \text{ supply voltage of Power stage [V]}$$

$$DVDD : Power \text{ supply voltage of DSP block [V]}$$

$$RL : \text{Load impedance } [\Omega]$$

$$r_{DS} : \text{Turn-on resistance of output MOS Tr. } [\Omega]$$

$$(typ.=160m \Omega)$$

 r_{DC} : DC resistance of output LPF coil [Ω]

If the circuit is driven further until an output waveform is clipped, an output power higher than that without distortion is obtained. In general a clipped output is quantified where "THD+N = 1% and 10%," and a maximum output power under that status is calculated by the following formula:

$$P_{O(1\%)} = \frac{(10^{(-0.3/20)} \times \frac{\text{VCC}}{\sqrt{2}} \times \frac{R_L}{2(r_{DS} + r_{DC}) + R_L})^2}{R_L} [W]$$

$$P_{O(10\%)} = P_{O(1\%)} \times 1.25[W]$$

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Power Dissipation

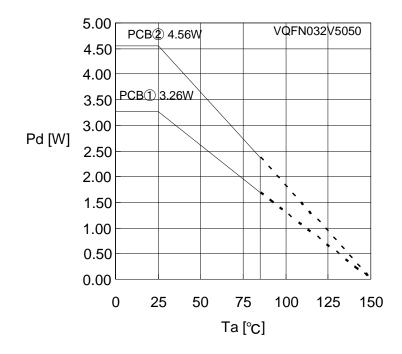


Figure 81. Allowable Power Dissipation

Measuring instrument : TH-156(Shibukawa Kuwano Electrical Instruments Co, Ltd.) Measuring conditions : Installation on ROHM' s board Board size : 74.2mm × 74.2mm × 1.6mm(with thermal via on board) Material : FR4

• The board on exposed heat sink on the back of package are connected by soldering.

PCB1 : 4- layer board (Top and bottom layer back copper foil size: 20.2mm², 2nd and 3rd layer back copper foil size: 5505mm²), Øja = 38.3°C/W

PCB2 : 4-layer board(back copper foil size: 5505mm²), $\theta_{ja} = 27.4^{\circ}$ C/W

Use a thermal design that allows for a sufficient margin in light of the power dissipation (Pd) in actual operating conditions. This IC exposes its frame of the backside of package. Note that this part is assumed to use after providing heat dissipation treatment to improve heat dissipation efficiency. Try to occupy as wide as possible with heat dissipation pattern not only on the board surface but also the backside.

Class D speaker amplifier is high efficiency and low heat generation by comparison with conventional Analog power amplifier. However, In case it is operated continuously by maximum output power, Power dissipation (Pdiss) may exceed package dissipation. Please consider about heat design that Power dissipation (Pdiss) does not exceed Package dissipation (Pd) in average power (Poav). (Tjmax : Maximum junction temperature=150°C, Ta : Peripheral temperature[°C], θ_{ja} : Thermal resistance of package[°C/W], Poav : Average power[W], η : Efficiency)

Package dissipation : $Pd(W)=(Tjmax - Ta) \neq \theta ja$ Power dissipation : $Pdiss(W)=Poav \times (1 \neq \eta - 1)$

I/O equivalence circuit (Provided pin voltages are typ. Values)

Pin				
No.	Pin name	Pin voltage	Pin explanation	Internal equivalence circuit
29	RSTX	OV	Reset pin for digital circuit High : Reset OFF Low : Reset ON	
30	MUTEX	OV	Speaker output mute control pin High : Mute OFF Low : Mute ON	
8	DGND	0V	GND pin for Digital I/O	_
31	SCL	_	 2 wire Bus control transmit clock input pin Please notice. Absolute Maximum Voltage is 4.5V. 	
32	SDA	_	 2 wire Bus control data input/output pin Please notice. Absolute Maximum Voltage is 4.5V. 	
1	ADDR	0V	2 wire Bus control Slave address select pin Select LSB data of slave address for 2 wire Bus control.	
2	BCLK	3.3V	Digital sound bit clock input pin Input bit clock of digital audio signal.	
4 3	SDATA LRCLK	3.3V	Digital sound signal input pin Input LR clock of digital audio signal to LRCLK terminal. Input data of digital audio signal to SDATA terminal.	
12	SDATAO	3.3V	Digital sound signal output pin Output data of digital audio signal.	

70/75

I/O equivalence circuit (Provided pin voltages are typ. Values)

Pin No.	Pin name	Pin voltage	Pin explanation	Internal equivalence circuit
6	PLL	1V	PLL's filter pin Connect filter circuit for PLL.	
10	DVDD	3.3V	Power supply pin for Digital I/O.	—
5 9 11	TEST1 TEST2 TEST3	_ _ _	Test pin Connect VSS.	
7	REG15	1.5V	Internal power supply pin for Digital circuit Connect capacitor.	
13	ERROR	3.3V	Error flag pin H: Normal operation L: Error	
14 28	NC	_	No connection Pin	_

I/O equivalence circuit (Provided pin voltages are typ. Values)

Pin No.	Pin name	Pin voltage	Pin explanation	Internal equivalence circuit
17	VCCP2	VCC	Power supply pin for ch2 PWM signal	
20	OUT2P	VCC to 0V	Output pin of ch2 positive PWM Connect output LPF.	
19	BSP2P	_	Boot-strap pin of ch2 positive Connect capacitor.	
18	GNDP2	0V	GND pin for ch2 PWM signal	
16	OUT2N	VCC to 0V	Output pin of ch2 negative PWM	│ ⊣€ 本 本
			Please connect to Output LPF.	
15	BSP2N	_	Boot-strap pin of ch2 negative Connect capacitor.	
22	BSP1N	_	Boot-strap pin of ch1 negative Connect capacitor.	
21	OUT1N	VCC to 0V	Output pin of ch1 negative PWM Connect output LPF.	
23	GNDP1	0V	GND pin for ch1 PWM signal	
26	BSP1P	_	Boot-strap pin of ch1 positive Connect capacitor.	
25	OUT1P	VCC to 0V	Output pin of ch1 positive PWM Connect output LPF.	│ →≤★
24	VCCP1	VCC	Power supply pin for ch1 PWM signal	
27	REG_G	5.7V	Internal power supply pin for gate driver Connect capacitor.	24 27 ₹370K 23

Operational Notes

1) Absolute maximum ratings

Use of the IC in excess of absolute maximum ratings such as the applied voltage or operating temperature range may result in IC damage. Assumptions should not be made regarding the state of the IC (short mode or open mode) when such damage is suffered. A physical safety measure such as a fuse should be implemented when use of the IC in a special mode where the absolute maximum ratings may be exceeded is anticipated.

2) Power supply lines

As return of current regenerated by back EMF of output coil happens, take steps such as putting capacitor between power supply and GND as an electric pathway for the regenerated current. Be sure that there is no problem with each property such as emptied capacity at lower temperature regarding electrolytic capacitor to decide capacity value. If the connected power supply does not have sufficient current absorption capacity, regenerative current will cause the voltage on the power supply line to rise, which combined with the product and its peripheral circuitry may exceed the absolute maximum ratings. It is recommended to implement a physical safety measure such as the insertion of a voltage clamp diode between the power supply and GND pins.

3) GND potential (Pin 8, 18, 23)

Any state must become the lowest voltage about DGND, GNDP1 and GNDP2 terminal.

4) Input terminal

The parasitic elements are formed in the IC because of the voltage relation. The parasitic element operating causes the wrong operation and destruction. Therefore, please be careful so as not to operate the parasitic elements by impressing to input terminals lower voltage than DGND and VSS. Please do not apply the voltage to the input terminal when the power-supply voltage is not impressed.

5) Actions in strong magnetic field

Use caution when using the IC in the presence of a strong magnetic field as doing so may cause the IC to malfunction. 6) Thermal shutdown circuit

This product is provided with a built-in thermal shutdown circuit. When the thermal shutdown circuit operates, the output transistors are placed under open status. The thermal shutdown circuit is primarily intended to shut down the IC avoiding thermal runaway under abnormal conditions with a chip temperature exceeding Tjmax = 150°C.

7) Shorts between pins and miss-installation

When mounting the IC on a board, pay adequate attention to orientation and placement discrepancies of the IC. If it is miss-installed and the power is turned on, the IC may be damaged. It also may be damaged if it is shorted by a foreign substance coming between pins of the IC or between a pin and a power supply or a pin and a GND.

8) Power supply on/off (Pin 10, 17, 24) In case power supply is started up, RSTX(Pin 29) and MUTEX(Pin 30) always should be set Low. And in case power supply is shut down, it should be set Low likewise. Then it is possible to eliminate pop noise when power supply is turned on/off. And also, all power supply terminals should start up and shut down together.

9) ERROR terminal (Pin 13)

An error flag is outputted when Output short protection or DC voltage protection. This flag is the function which the condition of this product is shown in.

10) N.C. terminal (Pin 14, 28)

N.C. terminal (Non Connection Pin) does not connect to the inside circuit. Therefore, possible to use open.

11) TEST terminal (Pin 5, 9, 11)

TEST terminal connects with ground to prevent the malfunction by external noise.

12) Precautions for Speaker-setting

If the impedance characteristics of the speakers at high-frequency range while increase rapidly, the IC might not have stable-operation in the resonance frequency range of the LC-filter. Therefore, consider adding damping-circuit, etc., depending on the impedance of the speaker.

13) External capacitor

In order to use a ceramic capacitor as the external capacitor, determine the constant with consideration given to a degradation in the nominal capacitance due to DC bias and changes in the capacitance due to temperature, etc. 14) About the rush current

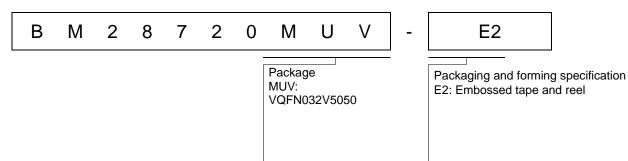
For ICs with more than one power supply, it is possible that rush current may flow instantaneously due to the internal powering sequence and delays. Therefore, give special consideration to power coupling capacitance, power wiring, width of GND wiring, and routing of wiring.

Status of this document

The Japanese version of this document is formal specification. A customer may use this translation version only for a reference to help reading the formal version.

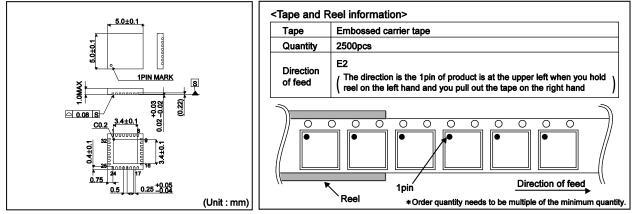
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Ordering Information

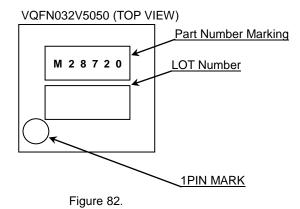


Physical Dimensions Tape and Reel Information

VQFN032V5050



Marking Diagram



Revision History

Date	Revision	Changes
21.Oct.2013	001	First version
10.Dec.2013	002	Corrected (P.2) Corrected performance curves (P.5-7) Modified recommend constant (P.63-64, 67, 68) Deleted application circuit (P.63-64)
10.Jun.2016	003	Modify electric characteristics, condition, explanation of command Correction of errors Modify power up sequence(P.22) Modify procedure for power up(P.59)

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CLASSⅣ	CLASSII	CLASSⅢ	CLASSII	

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- 2. In principle, the reflow soldering method must be used on a surface-mount products, the flow soldering method must be used on a through hole mount products. If the flow soldering method is preferred on a surface-mount products, please consult with the ROHM representative in advance.

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 - [c] the Products are exposed to direct sunshine or condensation
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