YAMAHA'L 5 I

NO.84-13

YM2203

FM Operator Type-N(OPN)

OUTLINE

OPN (FM OPERATOR TYPE-N) is a new type synthesizer which can produce all sounds required owing to the FM sound source system. It is provided with a built-in register which can store sound information and be connected easily with a microprocessor or microcomputer. It also comprises a square wave sound source different from the sound source according to the FM system and a noise generator.

FEATURES

- *The FM system sound source produce three different sounds simultaneously.
- *One of the above three sounds can be set to the mode by which specific sound effects and composite sine wave sound are synthesized.
- *Two programmable timers are incorporated.
- *8 bits general purpose input/output ports of two system are incorporated.
- *Three square wave sounds and white noise can be produced in addition to the FM system sounds.
- *Clock divider is built in so that wide operating frequency range is obtained.
- *Input and output are compatible with TTL.
- *Nch-Si gate MOS LSI is used.
- *Single phase power source of 5V is used.
- *This is compatible with software of YM2149 and AY-3-8910 and 8912 produced by GI.

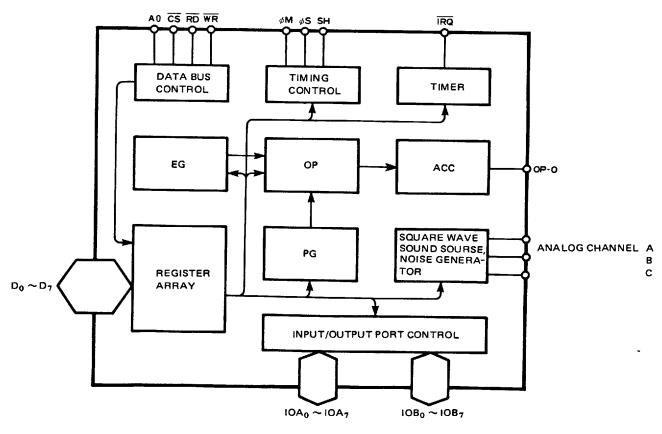
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YM2203 カタログ

■ TERMINAL DIAGRAM

GND		40	DQ
DI	2	39	φS
D2	3	38	φM
DЗ	4	37	A0
D4	5	36	RD
D5	6	35	WR
D6	7	34	cs
Đ7	8	33	1087
IOA7	9	32	1086
IOA6	10	31	1085
IOA5	11	30	10B4
IOA4	12	29	10B3
EAOI	13	28	1082
10A2	14	27	IOBI
I AOI	15	26	IOB0
IOA0	16	25	ĪRQ
AGND	17	24	īc
ANALOG CHANNEL C	18	23	0P-0
ANALOG CHANNEL B	19	22	SH
ANALOG CHANNEL A	20	21	V_{DD}

BLOCK DIAGRAM



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DESCRIPTION OF TERMINAL FUNCTION

1. ϕM

This is the master clock of the OPN. The FM sound source and square wave sound source operate, based on this clock. The maximum input frequency up to 4.2MHz can be input by using a built-in 1/6 divider.

2. $\phi S \cdot SH$

These are the clock (ϕ s) and the synchronous signal (SH). They drive a D/A converter which converts digital output of the FM sound source into analog output.

3. D_0 through D_7

These 8-bit bi-directional bus exchange the data and address between the OPN and the micro-processor.

4. $\overline{CS} \cdot \overline{RD} \cdot \overline{WR} \cdot A_0$

These signals control bi-directional bus of D_0 through D_7 .

CS RD WR A ₀	
0 1 0 0	Writes address into the register of the OPN.
0 1 0 1	Writes the resister content into the OPN.
0 0 1 0	Reads the OPN status.
0 0 1 1	Reads the content of the OPN register.
1 x x x	D ₀ through D ₇ bus line become high impedance.

^{*} Read enable register addresses 00 througe 0F (16 bits).

5. IRQ

This is an interrupt signal sent from two timers. It can be masked by the program.

6. IC

This signal resets the system at low level. All the content of register array become "0".

7. OP-O

This outputs the FM modulated audio signal as 13-bit serial data. Accordingly, an external D/A converter is necessary.

8. Analog channel A, B and C

They are analog square wave audio signals. They can be mixed by setting resistance because of source follower.

9. IOA₀ through IOA₇, IOB₀ through IOB₇

They are two sets of 8-bit input/output ports. Each terminal incorporates pull up resistance.

10. AGND

This is analog ground terminal for the D/A converter which is built in the square wave sound source section.

11. VDD

This is a power terminal of + 5V.

12. GND

This is a ground terminal.

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DESCRIPTION OF FUNCTIONS

The OPN is controlled based on the data written into the register. Accordingly, a microprocessor is free from the sound control operation except sending the data to the register.

The FM sound source determines a sound by the combination (modulation) of four sin waves. All the modulation systems such as feedback FM, simple FM and multiple FM are possible. In respect to the square wave sound source, this is compatible with YM2149 (SSG) and AY-3-8910 and 8912 (PSG.GI) in the use of the software. Therefore, function of the OPN can be improved by the exchange with the above LSI.

Each block of the OPN functions as follows.

*Envelope generator (EG): Determines the modulation index of the envelope and modulation

wave of the FM sound source.

*Phase generator (PG): Determines the sin wave phase at each time step of the FM sound

source.

*Operator (OP): Calculates the E sin θ value on the basis of the amplitude from the

envelope generator and the phase from the phase generator.

*Accumulator (ACC): Accumulates and adds operator output of each channel to mix

each sound of the FM sound source and matches with the D/A

converter.

*Square wave sound source/noise generator:

Generates three different frequency square waves and pseudowhite noise. It can also mix noise and square wave. As for sound volume, it is possible to select either fixed sound volume (programmed value) or 10 pattern envelope producing mode. In this block, one D/A converter is provided for each sound.

*Input/output port control: These are the general-purpose input/output ports to gets interface

with external equipment.

*Timer: Two types of timers are provided.

☆ Register content and address map

The OPN register is provided with the internal address as shown in the address map.

The content of each register (address) is as follows.

(1)	\$ 00 ~ \$ 05	Generates frequency of the square wave sound source.
(2)	\$ 06	Generates frequency of noise source.
(3)	\$ 07	Controls the input and output of the input and output ports and the output of musical sound and noise.
(4)	\$ 08 ~ \$ 0A	Controls sound volume. It is possible to select the fixed sound volume system (programmable) or the variable sound volume system.
(5)	\$ OB ~ \$ OC	Controls the envelope cycle in the variable sound volume sytem.
(6)	\$ 0D	Specifies the envelope shape in the variable sound volume system.
(7)	\$ OE ~ \$ OF	8 bit general-purpose input and output ports.
(8)	\$ 21	Test information, always set to "0".
(9)	\$ 24 ~ \$ 26	Gives the set time of Timers A and B.

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(10)	\$ 27	Controls the operation of Timers A and B, and sets the third channel mode of the FM sound source.
(11)	\$ 2D ~ \$ 2F	Specifies the dividing number of the input clock. The dividing numbers 2 through 6 are for the FM sound source, and the numbers 1 through 4 are for square wave sound source.
(12)	\$ 30 ~ \$ 3E	Controls Detune and Multiple. This is used to set tones. This controls the relationship between the fundamental wave and harmonic.
(13)	\$ 40 ~ \$ 4E	Gives the total level. This information becomes the modulation index of the sound volume and modulation wave of the modulated wave.
(14)	\$ 50 ~ \$ 5E	Key -Scale controls the rate of change of A · D · S and R according to the keyboard information. Attack rate gives the rate of change of the envelope at the time of attack.
(15)	\$ 60 ~ \$ 6E	Decay Rate shows the rate of change of the envelope at the time of decay.
(16)	\$ 70 ~ \$ 7E	Sustain Rate shows the rate of change of the envelope at the time of sustain.
(17)	\$ 80 ~ \$ 8E	Sustain level shows the level of the shift from decay to sustain. Release rate shows the rate of change of the envelope at the time of release.
(18)	\$ 90 ~ \$ 9E	Generates the envelope including the repeat pattern similar to that of square wave sound source.
(19)	\$ A0 ~ \$ A6	Gives key-code (F-number) of each channel.
(20)	\$ A8 ~ \$ AE	Gives the key-code (F-number) of three channels when set to the special mode.
(21)	\$ BO ~ \$ B2	Gives the modulation system (connection) of the FM modulation and the modulation factor of the feedback FM (self-feedback).

☆ FM system

In the FM system, musical sounds are synthesized by controlling various high harmonic waves by use of the frequency modulation.

The basic equation of the FM system is as follows.

$$F = A \sin (\omega Ct + I \sin \omega Mt)$$
 (1)

Where A is output amplitude, I is modulation index, and ωC and ωM are angular frequencies of carrier and modulator, respectively.

This equation can also be expressed as follows.

$$F = A [J_0 (I) \sin \omega Ct + J_1 (I) (\sin (\omega C + \omega M)t - \sin (\omega C - \omega M)t) + J_2 (I) (\sin (\omega C + 2\omega M)t - \sin (\omega C - 2\omega M)t) + \cdot \cdot \cdot \cdot \cdot$$

$$] -(2)$$

Where, Jn (I) is the first class Bessel function of nth. As shown in the above equation, the FM system contains various harmonics and can control them.

The OPN provides the multiple FM modulation and feedback FM modulation shown in (3) and (4) in addition to the above FM modulation to produce every possible sound.

$$F = A \sin \left[\omega Ct + I_1 \sin \left(\omega M_1 t + I_2 \sin \omega M_2 t\right)\right] - (3)$$

$$F = A \sin \left(\omega Ct + \beta F\right) - (4)$$

*** WRITE DATA**

ADDRESS	Γ					
	1					
21	TEST					
24	TIMER-A TIME					
25						-A
26		- T	TIM	ER-I		LOAI
27	MOI	JE	BA		A	B A
28		SLO	Γ			СН
2D	-					
2E	-					
2F	1					
30		D	Т		MUI	ЛI
3E	$V \downarrow$					
40 4E				TL		
4E		т-	1			
50	KS			4	AR	
5E	<u> </u>	<u>V_</u>				
60		DR				
6E	1					
70 2F				,	SR	
7E	\vdash					
80		SL			RF	₹
8E				\perp		
90 9E		/			SSG-	EG
A0 A1 A2			F-N	um.	1	
A4 A5 A6			BLOG	CK	F-N	lum. 2
A8 A9 AA		3	СН *	F-Nu	ım. 1	
AC AC		/:	BCH *		3CH	(*
AD AE			BLC	ск	F-N	Num. 2
B0 B1 B2			FB		CO	NNECT

LSI TEST DATA	
8 most significant bits of TIMER-A	
2 least significant bits of TIMER-A	
TIMER-B DATA	
TIMER-A/B control and 3 channel mode	
Key-ON/OFF	
Set pre-scaler.	
Selection of the dividing numbers of 1/3 and 1/6.	
Set a divider to the dividing number of 1/2.	
Detune / Multiple	
(Addresses at 33, 37 and 3B are empty.)	
(Addresses at 35, 57 and 35 are empty.)	
Total Level	
(Addresses at 43, 47 and 4B are empty.)	
Key Scale / Attack Rate	
(Addresses at 53, 57 and 5B are empty.)	
Decay Rate	
(Addresses at 63, 67 and 6B are empty.)	
Sustain Rate	
(Addresses at 73, 77 and 7B are empty.)	
(Addresses at 75, 77 and 75 are empty.)	
Sustain Level / Release Rate	,
(Addresses at 83, 87 and 8B are empty.)	
SSG-Type Envelop Control	
(Addresses at 93, 97 and 9B are empty.)	
F-Numbers / BLOCK	
3CH-3slot	
F-Numbers / BLOCK	
·	
Self-Feedback / Connection	

COMMENT

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*** READ / WRITE DATA**

ADDRESS

ADDRESS
00
01
02
03
04
05
06
07
08
09
0A
0 B
0C
0 D
0E
0F

Fine Tune					
	_	Coarse Tune			
	Fir	ne Tune	•		
		Co	arse T	une	
	Fir	ie Tun	e		
	_	Co	oarse T	une	
		Perio	d Conti	ol	
IN/OUT IOB IOA	/Noise /Tone			one	
	M Level				
	M		Level		
	М		Level		
	Fir	ne Tun	e		
Coarse Tune					
	ATT	ALT	HLD		
I/O Port-A					
I/O Port-B					

COMMENT

Channel-A Tone Period
Channel-B Tone Period
Channel-C Tone Deriod
Noise Period
/ENABLE
Channel-A Amplitude
Channel-B Amplitude
Channel-C Amplitude
Envelop Period

*READ DATA

ADDRESS

xx

BUSY	FLAG B A

COMMENT

Status

Envelop Shape/Cycle

I/O Port Date

■ ELECTRICAL CHARACTERISTICS

1. Absolute Maximum Rating

ITEM	RATING	UNIT
Terminal voltage	$-0.3 \sim 7.0$	V
Ambient operating temperature	0 ~ 70	°C
Storage temperature	−50 ~ 125	°C

2. Recommended Operation Conditions

ITEM	SYMBOL	MIN.	STD.	MAX.	UNIT
6 1 1	Vdd	4.75	5.0	5.25	V
Supply voltage	Vss	0	0	0	v

3. DC Characteristics

ITEM		SYMBOL	CONDITIONS	MIN.	STD.	MAX.	UNIT
Input high level voltage	Total input	ViH		2.0		VDD	v
Input low level voltage	Total input	VIL		- 0.3		0.8	V
Input leakage current	ϕ M, \overline{WR} , \overline{RD} , A_0	IL	Vin = 0 ~ 5V	- 10		10	μΑ
Three-State (off) input current	$D_0 \sim D_7$	I TSL	Vin = 0 ~ 5V	- 10		10	μΑ
Output higl level voltage	Output except IRQ	Vo _{H1}	$IOH_1 = 0.4mA$	2.4			v
		VOH ₂	$IOM_2 = 40\mu A$	3,3			V
Output low level voltage	Total output	Vol	IOL = 2mA			0.4	v
Output leakage current (off)	ĪRQ	IOL	$V_{OH} = 0 \sim 5V$	– 10		10	μΑ
Analog output voltage	ANALOG-CHA, B, C	Voa	Max. Sound volume, no mixing	0.95		1.35	Vpp
Power current		IDD				120	mA
Pull-up resistance	$IOA_0 \sim IOA_7$, $IOB_0 \sim IOB_7$, \overline{IC} , \overline{CS}	RPU		60		600	kΩ
Input capacitance	Total input	C ₁	f = 1MHz			10	pF
Output capacitance	Total output	Co	1 – 1141117			10	pF

■ AC CHARACTERISTICS

4. AC Characteristics

ITEM		SYMBOL	CONDITIONS	MIN.	STD.	MAX.	UNIT
Input clock frequency	φМ	fC	Pre-scaler function (Fig. A-1)	0.7		4.2	MHz
Input clock duty	φM			40	50	60	%
Input clock rise time	φM	TR	(Fig. A-1)			50	ns
Input clock breaking time	φM	TF	(Fig. A-1)			50	ns

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Access to FM sound source

ITEM		SYMBOL	CONDITIONS	MIN.	STD.	MAX.	UNIT
Address set-up time	A ₀	TAS	(Figs. A-2 and A-3)	10			ns
Address hold time	A_0	ТАН	(Figs. A-2 and A-3)	10			ns
Chip select write width	CS	TCSW	(Fig. A-2)	200			ns
Chip select read width	CS	TCSR	(Fig. A-3)	250			ns
Write pulse write width	WR	Tww	(Fig. A-2)	200			ns
Write data set-up time	$D_0 \sim D_7$	Twds	(Fig. A-2)	100			ns
Write data hold time	$D_0 \sim D_7$	TWDH	(Fig. A-3)	20			ns
Read pulse width	RD	TRW	(Fig. A-3)	250			ns
Darid data access times	D - D	TACC	CL = 100pF			250	ns
Read data access time	$D_0 \sim D_7$	TACC	(Fig. A-3)				113
Read data hold time	$D_0 \sim D_7$	TRDH	(Fig. A-3)	10			ns
	φs	TOR ₁	CL = 100pF			200	ns
Output rise time	Ψ	TOR	(Fig. A-4)				113
Output lise time	OP-O, SH	TOR ₂	CL = 100 pF			300	ns
		1002	(Fig. A-5)				
Output rise time	φs	TOF ₁	CL = 100pF			200	ns
	70	10.1	(Fig. A-4)				
Output the time	OP-O, SH	TOF ₂	CL = 100pF			300	ns
	0, 0,	1012	(Fig. A-5)				

Access to SSG sound sourse

ITEN	Mi	SYMBOL	CONDITIONS	MIN.	STD.	MAX.	UNIT
Address set-up time	Ao	TSAS	(Figs. A-7 and A-8)	10			ns
Address hold time	Ao	TSAH	(Figs. A-7 and A-8)	10			ns
Chip select writh width	CS	TSCSW	(Fig. A-7)	250			ns
Chip select read width	CS	TSCSR	(Fig. A-8)	400			ns
Write pulse write width	WR	Tsww	(Fig. A-7)	250			ns
Write data set-up time	$D_0 \sim D_7$	TSWDS	(Fig. A-7)	0			ns
Write data hold time	$D_0 \sim D_7$	TSWDH	(Fig. A-7)	20			ns
Read pulse width	RD	TSRW	(Fig. A-8)	400			ns
Read data access time	$D_0 \sim D_7$	TSACC	CL = 100pF (Fig. A-8)			400	ns
Read data hold time	$D_0 \sim D_7$	TSRDH	(Fig. A-8)	10			ns

ſ	ITEM		SYMBOL	CONDITIONS	MIN.	STD.	MAX.	UNIT
	Reset pulse width	ĪC	TICW	(Fig. A-9)	72*			cycle

^{*} Depends on the dividing number of prescaler.
Pulse width = (dividing number) x 12

■ TIMING DIAGRAM (Timing is set on the basis of the values: VIH = 2.0V and VIL = 0.8V.)

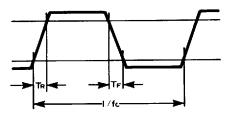


Fig. A-1 Clock timing

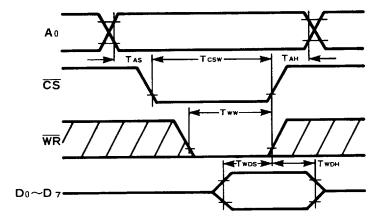


Fig. A-2 FM section write timing

Note.

TCSW , Tww and Twdh are determined based on the time when either \overline{CS} or \overline{WR} goes to the high level.

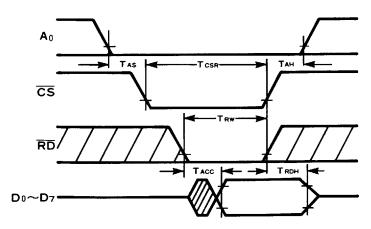


Fig. A-3 FM section read timing

Note.

TAAC is determined based on the time when either \overline{CS} or \overline{RD} goes to the low level. TCSR, TRW and TRDH are determined based on the time when either \overline{CS} or \overline{RD} goes to the high level.

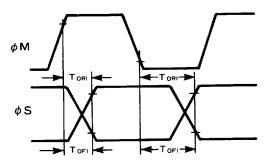


Fig. A-4-a ϕ M and ϕ S (dividing numbers: 2 and 3)

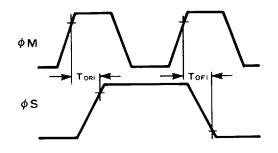


Fig. A-4-b ϕ M and ϕ S (dividing number: 6)

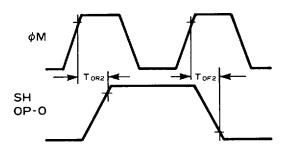


Fig. A-5 ϕ M and SH · OP-O

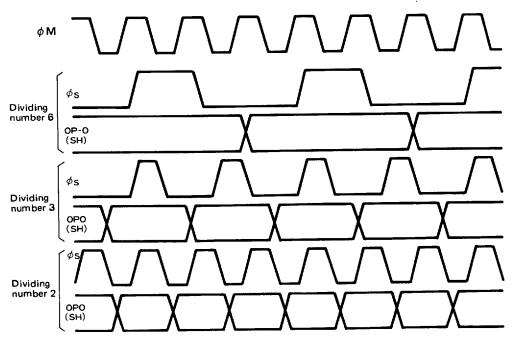


Fig. A-6 Timing of ϕ s and OP-O/CH at each dividing number

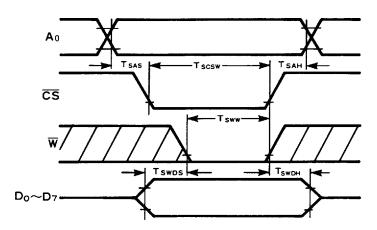


Fig. 6-7 SSG section write timing

Note.

Tswps is determined based on the time when either \overline{CS} or \overline{WR} goes to the low level.

TSCW, TSWW and TSWDH are determined based on the time when either \overline{CS} or \overline{WR} goes to the High level.

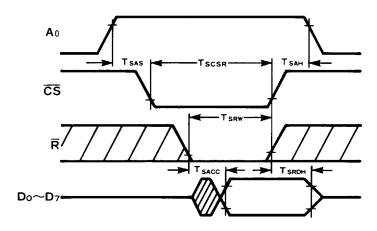


Fig. A-8 SSG section read timing

Note.

TSACC is determined based on the time when either \overline{CS} or \overline{RD} goes to the low level. TSCSR, TSRW and TSRDH are determined based on the time when either \overline{CS} or \overline{RD} goes to the High level.

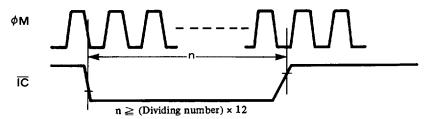


Fig. A-9 Reset pulse

OUTER DIMENSION DRAWING

