

# Sound Document or How to Make Music with the Sega

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Revision 0.3 Nov 27, 1990

**! Caveat :** This information has been obtained by careful inspection of, and experimentation on the sound hardware. While the information is fairly complete with respect to the capacity of the sound hardware concerning the use of 'patches' and 'instruments', it is not complete with respect to the full capabilities of the Yamaha sound chip in the Sega. We know of specific capabilities that are not discussed in this document (Revision 0.2 specifically digitized sound output - Dan Thompson and Tim Wilson know how to do this.).

Some Definitions are in order.

Instrument: the basic high level unit of sound output or sound effect.

The sound chip has six instruments.

Voice: see waveform

Patch: the data that characterizes the sound quality of an instrument.

We might have a piano patch, or a phaser patch, or trumpet patch, etc.

Waveform: the lowest level of sound output. In our case each instrument is

capable of combing up to four waveforms.

## More Caveats -

This is very much a 'document in progress' and serves mostly to give an over view. Also, this is the document to refer to when trying to figure out patch register function and use. For actually making sounds, the better choice of document is the Z-80 code of the Sound Driver. After all, the final proof of accuracy in documentation is when you have the sounds coming out of the machine, and to do that I highly recommend using the Sound Driver. It's a way of life. Any way, by using the Sound Driver, we can use your feedback to keep the Sound Driver and this document in a current state. So, if you use the Sound Driver and go hacking it to your own obscure purposes, let us know what you have discovered so we can incorporate and share your knowledge.

- thanks,  
Andrew

**Outline:** ( not followed rigorously )

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## Sound Chip Register Access

There are two pages of registers that can be accessed in the sound chip. Each page is accessed by a pair of memory locations.

Page A:

SRegA at \$4000 (Z-80 relative)

SValA at \$4001

Page B:

SRegB at \$4002

SValB at \$4003

Setting the value of a register is accomplished as follows:

1. LD SVal and loop until bit 7 is clear.
2. LD SReg with the register number you want to access.
3. LD SVal with the data.

( for specific code example see the Sound Driver )

Register Map - stapled to back -

( see THINK C document - because this WP sucks )

## The Sound Chip - Over View of Capabilities

The main feature we are concerned with in this document is the ability to generate music and sound effects. The Sound Chip is capable of playing six independent instruments simultaneously. Each of the instruments is identical in format.

Instrument Description: ( here instrument and patch are used interchangeably )

An instrument is comprised of four wave forms labeled A, B, C and D.

Each wave form has associated the following variables:

Harmonic multiplier : the multiple of the root frequency.

Detune factor : a small 'delta' to pull the wave frequency away from tune.

Attenuation : control of the wave form amplitude.

Envelope scaler : an overall scaler for attack,decay,sustain and release timing:

Attack rate : control of the attack time of the envelope.

Decay rate : control of the decay time of the envelope.

Sustain Amplitude : control the volume sustain begins at.

Sustain : control of the sustain slope ( zero slope is common to other kind of sound chip ).

Release rate : control of the rate of envelope once release has been triggered.

Root frequency : control of the frequency that the harmonic multipliers use as a base from which to multiply.

Control A and Control B : control of how the wave forms are shaped, how they are combined and where they are output in terms of left/center/right channel.

The implementation of an instrument i.e. the above components are contained in a 32 byte patch:

	A	B	C	D
0	detune/ harmonic multiplier	...	...	...
1	attenuation	...	...	...
2	scaler/attack	...	...	...
3	decay	...	...	...
4	sustain	...	...	...
5	sustain level/ release	...	...	...
6	unknown function function	unknown function	unknown function	unknown
7	root frequency register L	root frequency register H		
8	Control register 1	Control register 2	<i>— L,R output &amp; resonance</i>	

Patch registers A0 thru A6, B0 thru B6, C0 thru C6, and D0 thru D6 have the same functional relations corresponding respectively to waveforms A, B, C, and D.

**Patch register x0 - detune/harmonic multiplier**

$x d_2 d_1 d_0 h_3 h_2 h_1 h_0$

$h$  bits 0,1,2,3 choose the factor the root frequency is multiplied by to obtain the waveform frequency. They form a number between 0h and Fh.

$h$ bits	multiplier	$h$ bits	multiplier
0h	1	8h	16
1h	2	9h	18
2h	4	Ah	20
3h	6	Bh	22
4h	8	Ch	24
5h	10	Dh	26
6h	12	Eh	28
7h	14	Fh	30

*d* bits 4,5,6 control the degree to which the waveform frequency is pulled out of tune. They form a 3 bit 'signed' number which we tabulate in an 'unsigned' manner.

<u><i>d</i> bits</u>	<u>detune factor</u>
%000	0.00%
%001	+0.10%
%010	+0.20%
%011	+0.25%
%100	0.00%
%101	-0.11%
%110	-0.20%
%111	-0.31%

( One use of the detune - by selecting two waves of the same frequency and pulling one of them off tune, you can achieve amplitude modulation in the form of a 'beat' between the two waves. )

#### Patch register x1 - attenuation

*x v6 v5 v4 v3 v2 v1 v0*

*v* bits choose the amount of attenuation applied to the waveform. 0 being no attenuation (full amplitude waveform) thru 0x7F being full attenuation (no waveform).

*x* bit - always observed to be 0.

<u><i>v</i> bits</u>	<u>amplitude</u>
00h	100%
01h	91.2%
02h	83.3%
04h	69.8%
08h	49.3%
10h	25.6%
20h	7.9%
40h	2.8%

#### Patch register x2 - envelope scale and attack rate

*s1 s0 x a4 a3 a2 a1 a0*

*s* bits control the scale factor applied to attack/decay/sustain/release.

%00 being the slowest factor and %11 being the quickest.

*x* bit has unknown function.

a bits control attack rate from never (0) to very fast (0x1F).  
the longest attack can be nearly 150 seconds, the shortest to fast to hear.

**Patch register x3 - decay rate**

x x x d<sub>4</sub> d<sub>3</sub> d<sub>2</sub> d<sub>1</sub> d<sub>0</sub>

d bits control decay rate like the a bits in instrument register x2.

**Patch register x4 - sustain rate**

x x x s<sub>4</sub> s<sub>3</sub> s<sub>2</sub> s<sub>1</sub> s<sub>0</sub>

s bits control sustain rate like the a bits in instrument register x2.

**Patch register x5 - sustain level and release rate**

v<sub>3</sub> v<sub>2</sub> v<sub>1</sub> v<sub>0</sub> r<sub>3</sub> r<sub>2</sub> r<sub>1</sub> r<sub>0</sub>

v bits control the level at which sustain is initiated - full volume (0)

no sound (0xF).

r bits control the rate at which a note is released once release is activated.

/\* make sure to check what happens if 0 - not done yet in version 0.2 \*/

**Patch register x6 - unknown function**

*filter (low pass?)*

~~These registers are almost always 0 in the patches observed except for one case where they are 0x20, but I have not been able to detect any effect that they might produce. (when set to FFh there is no sound out.)~~

It is likely that their function is selected by some other bits, possibly in the control registers - I have a feeling they affect phase modulation, but I haven't verified this.

**Patch registers A7 and B7 - root frequency**

A7 - f<sub>7</sub> f<sub>6</sub> f<sub>5</sub> f<sub>4</sub> f<sub>3</sub> f<sub>2</sub> f<sub>1</sub> f<sub>0</sub>

least significant bits of the root frequency

B7 - x x o<sub>2</sub> o<sub>1</sub> o<sub>0</sub> f<sub>10</sub> f<sub>9</sub> f<sub>8</sub>

most significant bits of the root frequency, and the octave bits

x bits unknown function

o bits control the octave of the f bits

f bits are the most significant bits of the root frequency

the functional relation has been measured to be:

30 9 1F 01 00 D5  
1E 15 1F 17 14 04  
01 00 1F 17 14 98  
10 00 1F 17 14 05  
10 00

$$\text{Frequency}(f,o) = (0.0126990 \pm 0.0000005) f * 2^o \text{ [Hz]}$$

( Under some conditions - not specifically clear - the frequency or at least the octave bits have an affect on the time scale of the envelope i.e. higher frequency makes for shorter envelope. )

**Patch register A8 - waveform control 1**

x w w w s s s

~~Qualitatively this controls which waveforms are added to the composite waveform and the waveform shapes e.g. sine, sawtooth, or noise.~~




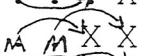
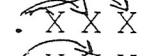

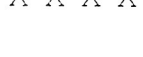
~~So far it appears that waveforms B, C, and D are restricted to being sinewaves and that waveform A can be selected to be sine/sawtooth/noise.~~

~~I believe that waveform B can also have a selected form but this is not yet verified.~~

~~A bitwise functional description is forthcoming.~~

This has proved to be a bit more difficult to determine than I expected.

Here is what I've found so far:

s bits	A B C D ( X indicates wave present)
%000	 X
%001	 X
%010	 X
%011	 X
%100	 X X
%101	 X X X
%110	 X X X
%111	X X X X

*Handwritten notes:*  
 2 = sine wave  
 3 = sawtooth  
 4 = noise

these bits are a little strange in that there doesn't seem to be a one-to-one relation between bit and wave ( except wave C which seems to be turned on by bit 2 ). Wave D is always on, and waves A and B are on where there are Xs in the above table. I assume there are subtleties that I have missed so far.

the w bits seem to control the shape of wave form A. When all the w bits are 0 wave A is sine shape. When all the bits are 1 wave A is 'noise', and between those values, the wave progresses from sine to sawtooth to noise in a 'linear' fashion.

*Handwritten mark:* 27



Patch register B8 - waveform control 2

$r \ / \ x_5 \ x_4 \ x_3 \ x_2 \ x_1 \ x_0$

$r$  - if set output to right channel

$l$  - if set output to left channel

It follows that if neither  $r$  nor  $l$  are set there is no output (not strictly true - some leaks thru) and that if both  $r$  and  $l$  are set then the output is centered between right and left channels.

~~$x$  bits - qualitatively - they at least control some phase modulation in waveform D and possibly all the wave forms.~~

~~A bitwise functional description is forthcoming.~~

resonance

Note - qualitative functional descriptions of envelope timing is pending better measurement instruments to get exact timing.

### Qualitative Envelope Diagram

