

## How to decode a Vortex Tracker II "PT3" File

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by Vince "Deater" Weaver, <vince@deater.net>  
[http://www.deater.net/weave/vmwprod/pt3\\_player/](http://www.deater.net/weave/vmwprod/pt3_player/)  
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### Background:

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Vortex Tracker II ( [https://bulba.undergrund.net/vortex\\_e.htm](https://bulba.undergrund.net/vortex_e.htm) )  
 is a music tracker (tool for writing music) that targets systems  
 with the AY-3-8910 sound chip. The music made is most popular  
 on various ZX Spectrum (z80) and Atari ST systems.

I wanted to make a decoder for a 6502-based Apple II with a  
 AY-3-8910 based Mockingboard sound card. The challenge was  
 all the low-level Vortex Tracker documentation is in Russian,  
 and the source code available is either uncommented z80 assembly  
 language or Russian-commented Pascal source code.

This document is based at least partly on the writeup here:  
<http://karoshi.auic.es/index.php?topic=397.msg4641#msg4641>  
 as well as the AY\_emul source code, and lastly just from reverse  
 engineering things while trying to get my code to match the output  
 of AY\_emul exactly.

### The PT3 Format

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#### \* File Header

Note: 16-bit values are little-endian

| Offset      | : Size     | : Description                         | : Contents                            |
|-------------|------------|---------------------------------------|---------------------------------------|
| -----       |            |                                       |                                       |
| \$00 - \$0C | : 13 bytes | : Magic                               | : "ProTracker 3."                     |
| \$0D        | : 1 byte   | : Version                             | : '5' for Vortex Tracker II           |
| \$0E - \$1D | : 16 bytes | : String                              | : " compilation of "                  |
| \$1E - \$3E | : 32 bytes | : Name                                | : Name of the module                  |
| \$3E - \$41 | : 4 bytes  | : String                              | : " by "                              |
| \$42 - \$62 | : 32 bytes | : Author                              | : Author of the module.               |
| \$63        | : 1 byte   | : Frequency table (from 0 to 3)       |                                       |
| \$64        | : 1 byte   | : Speed/Delay                         |                                       |
| \$65        | : 1 byte   | : Number of patterns+1 (Max Patterns) |                                       |
| \$66        | : 1 byte   | : LPosPtr                             | : Pattern Loop Pointer                |
| \$67 - \$68 | : 2 bytes  | : PatsPtrs                            | : Pointers to the patterns            |
| \$69 - \$A8 | : 64 bytes | : SamPtrs[32]                         | : Pointers to the samples             |
| \$A9 - \$C8 | : 32 bytes | : OrnPtrs[16]                         | : Pointers to the ornaments           |
| \$C9 - ???  | :          | : Patterns[]                          | : \$FF terminated, More on this below |

Version: is at offset \$0D, which is just the ASCII value of  
 the version number from the magic. Usually you  
 subtract off '0' to get the proper value.  
 If the value is not a valid number, '6' is assumed.

#### \* Pattern list

The pattern list starts at \$C9 and is terminated by \$FF

The pattern is multiplied by 3.

A sample pattern list might look like

\$03, \$06, \$09, \$FF

which corresponds to playing in order patterns

1, 2, 3

## \* Samples

The pt3 file allows for 32 samples. These samples contain values that are applied to the music notes as they are playing.

The 16-bit address of a sample X can be found by getting the 16-bit little-endian address at offsets \$6A+(X\*2) and \$6B+(X\*2) in the header.

Byte 0: LOOP VALUE -- sample offset to return to once hit the end  
 Byte 1: LENGTH -- number of 32-bit sample values  
 Byte 2+: VALUES -- list of 4-byte (32-bit samples)

## \* Sample format

```
+ Byte 0 --- 7 6 5 4321 0
                Bit 7 - Amplitude sliding
                Bit 6 - 1/0 amplitude slide up/down
                Bit 5 - 1/0 envelope slide up/down
                Bits 4321 - sign extend, envelope slide value
                Bit 0 - Sample has envelope

+ Byte 1 --- 7 6 5 4 3210
                Bit 7 - Envelope sliding
                Bit 6 - Accumulate Tone
                Bit 5 - parameter to envelope sliding?
                Bit 4 - Bit 6+4 used to set Noise/Channel
                        mixer value
                Bits 3210 - Amplitude

+ Byte 2 --- Freq Low -\
+ Byte 3 --- Freq High----- Used as base tone
```

## \* Ornaments

The PT3 file format has 16 ornaments, which are patterns applied to the note. This can be used for ?? effects.

The 16-bit address of an ornament X can be found by getting the 16-bit little-endian address at offsets \$A9+(X\*2) and \$AA+(X\*2) in the header.

Byte 0: LOOP VALUE -- ornament offset to return to once hit the end  
 Byte 1: LENGTH -- number of ornament values  
 Byte 2+: VALUES -- list of single-byte values applied to the notes

## \* Pattern data

For each of three channels (A,B,C) there is a nul-terminated stream of bytes that gives the pattern data.

To find the pattern data for a pattern, look it up in

A 6-byte chunk with the 16-bit addresses for A,B,C for pattern X can be found by

```
a_addr_l = [address in ($68/$67)] + (X*6)+0
a_addr_h = [address in ($68/$67)] + (X*6)+1
b_addr_l = [address in ($68/$67)] + (X*6)+2
b_addr_h = [address in ($68/$67)] + (X*6)+3
c_addr_l = [address in ($68/$67)] + (X*6)+4
c_addr_h = [address in ($68/$67)] + (X*6)+5
```

Follow that address to find the nul-terminated pattern data.

The data can be interpreted this way:

```
+ $00 - NUL termination, end of pattern
+ $01-$0f - effects, see later.
```

Note that parameters to the effect appear in the bytestream \*after\* the note to play

- + \$10-\$1f - set envelope type
  - + \$10 means disable envelope
    - sample number is next byte
  - + \$11-\$1F
    - envelope type is bottom 4 bits
    - envelope period is next 2 bytes (big endian)
    - envelope delay is next 1 byte
    - sample number is next byte
- + \$20-\$3f - set noise
  - Noise set to the (\$20...\$3F) value - \$20
- + \$40-\$4f - set ornament
  - ornament sent to bottom 4 bits
  - possibly setting 0 counts as disabling
- + \$50-\$af - play note

|     | 0   | 1   | 2   | 3   | 4   | 5   | 6   | 7   | 8   | 9   | a   | b   | c   | d   | e   | f   |
|-----|-----|-----|-----|-----|-----|-----|-----|-----|-----|-----|-----|-----|-----|-----|-----|-----|
| 50: | C-1 | C#1 | D-1 | D#1 | E-1 | F-1 | F#1 | G-1 | G#1 | A-1 | A#1 | B-1 | C-2 | C#2 | D-2 | D#2 |
| 60: | E-2 | F-2 | F#2 | G-2 | G#2 | A-2 | A#2 | B-2 | C-3 | C#3 | D-3 | D#3 | E-3 | F-3 | F#3 | G-3 |
| 70: | G#3 | A-3 | A#3 | B-3 | C-4 | C#4 | D-4 | D#4 | E-4 | F-2 | F#4 | G-4 | G#4 | A-4 | A#4 | B-4 |
| 80: | C-5 | C#5 | D-5 | D#5 | E-5 | F-5 | F#5 | G-5 | G#5 | A-5 | A#5 | B-5 | C-6 | C#6 | D-6 | D#6 |
| 90: | E-6 | F-6 | F#6 | G-6 | G#6 | A-6 | A#6 | B-6 | C-7 | C#7 | D-7 | D#7 | E-7 | F-7 | F#7 | G-7 |
| a0: | G#7 | A-7 | A#7 | B-7 | C-8 | C#8 | D-8 | D#8 | E-8 | F-8 | F#8 | G-8 | G#8 | A-8 | A#8 | B-8 |

- + \$b0 - disable envelope, reset ornament position
- + \$b1 - set skip value
  - next byte is how many lines to skip for this note
- + \$b2-\$bf - set envelope
  - envelope type to (bottom 4 bits - 1)
  - envelope period next 2 bytes (big endian)
- + \$c0 - turn off note (volume=0 and disable everything)
- + \$c1-\$cf - set volume
  - volume set to bottom 4 bits
- + \$d0 - done processing this note (end note)
- + \$d1-\$ef - set sample
  - sample set to value-\$d0.
- + f0-\$ff - initialize ornament/sample. Envelope is disabled
  - ornament is bottom 4 bits
  - next byte is sample\*2

Note when parsing if you reach a note, a \$D0 or a \$C0 then you are done parsing the note for this line and should move on to parsing the effects.

#### Effects

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AY\_emul supports having up to one of each effect for each line, and tracks the order in which to apply them. However in the wild I have not seen more than one effect applied per line/channel (I don't think Vortex Tracker lets you add multiple?)

The bytes for the effects follow after the byte that ended the note.

The effect value on disk is not the same as that displayed when viewed in the tracker.

On	In
Disk	Tracker
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\$01      \$01:      Glissando (slide through discrete notes) / Tone Down

First byte: delay  
Next 2 bytes: frequency to add (is negative)

\$01    \$02:    Glissando (slide) / Tone Up

This appears differently in tracker, but is just the same as above but with a positive frequency.

\$02    \$03:    Tone portamento (continuous slide)

First byte: delay  
Next 2 bytes: ignored?  
Next 2 bytes: slide step (little-endian)

\$03    \$04:    Sample Offset  
First byte: value to set the sample position offset

\$04    \$05:    Ornament offset  
First byte: value to set the ornament offset

\$05    \$06:    Vibrato  
Periodic sound off/on in that channel  
First byte: OffOn delay (frames to stay off)  
Second byte: OnOff delay (frames to stay on)

\$08    \$09:    Envelope Glissando -- Frequency decreasing  
First byte: delay  
Next two bytes: Slide add (little-endian)

\$08    \$0A:    Envelope Glissando -- Frequency increasing.  
Like previous but with sign switched?

\$09    \$0B:    Set playing speed (new Delay).  
First byte: new delay (global number of frames per line)

#### \* Frequency Tables

Various versions of the tracker use different frequency tables.  
Players lookup notes in these tables to get the proper frequency.

The value in the frequency table field specifies up to 4 (0...3) tables, but there is an alternate set of tables if the tracker version PT3\_VERSION is 3 or less.

Presumably for space reasons, some players only support tables one and two from the newer set.

These tables in theory can be calculated at runtime, but usually they aren't unless you are extremely space constrained.  
The z80 player has code to do this in z80 assembly.

The tables are a set of 96 16-bit values (8 octaves of 12 notes)

The frequency tables supported by Ay\_Emul:

PT3Version#	Table#	Frequency Table
-----	-----	=====
<=3.3	0	PT3NoteTable_PT_33_34r
all	1	PT3NoteTable_ST
<=3.3	2	PT3NoteTable_ASM_34r
<=3.3	3	PT3NoteTable_REAL_34r
3.4+	0	PT3NoteTable_PT_34_35
all	1	PT3NoteTable_ST
3.4+	2	PT3NoteTable_ASM_34_35
3.4+	3	PT3NoteTable_REAL_34_35

#### \* Volume Tables

There are multiple 256-byte amplitude/volume lookup tables  
These seem to have changed with different versions of Vortex Tracker  
so to be complete you need to contain them all and use the proper  
one at run time

\* Corner cases seen in the wild:

+ What to do if the delay on an effect is set to zero?

\* 6-channel PT3 Files

In the Protracker 3.7 format you can look for a turbosound header  
at the very end of the file.

Type1 (4 bytes), Size (16-bits, little-endian)

Type2 (4 bytes), Size (16-bits, little endian)

TSID: (4 bytes) 02TS

If the type of these files is PT3! then these are describing  
PT3 subfiles, and can be read as two 3-channel files which can  
be interpreted together as one 6-channel file.