Blueprint of DSynkant & reverse engineering of Roland D-50

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Introduction

1.1 What is this document about?

The goal of this document is to describe in all details every missing information from the official documentation of Roland D-50.

1.2 What this document is not?

This document is not a user manual of DSynkant. For this purpose you should referee to the manual of DSynkant available on the official website of DSynkant

http://dsynkant.sourceforge.net

1.3 How to read this document?

This document contains both answers and questions not answered yet. When you see green somewhere in the document you'll know it is a question. If you have the answer of any question please contact me at

a-lin[try_to_remove_that]@users.sourceforge.net

if you notice english mistakes please contact me too...

The engine

In this chapter it is question of the engine of the synthesizer and also everything concerning the conversion between the parameters manipulated by the user (user parameters) and the internal parameters of the engine (engine parameters).

2.1 Notations

2.1.1 User parameters and engine parameters

To recognize the difference between the user parameters and the engine parameters, the user parameters will have the post-fix _usp added and the engine parameters no particular notation.

2.1.2 Mathematical notations

Int[n.m] the set of integers between n and m included

Real[n, m] the set of reals between n and m included

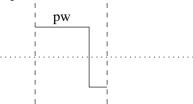
log(x) the natural logarithm, that is the inverse function of exp(x)

2.2 Wave generator

2.2.1 Wave form

Synthesizer sound generator

• Square form:



The relation between $pulseWidth_usp \in Int[0, 100]$ and $pw \in Real[0.5, 1]$ can be well approximated by the following equation :

$$pw = \gamma \times log(\alpha * pulseWidth_usp + \beta)$$

where $\alpha=1.17844$, $\beta=11.7539$ and $\gamma=0.201171$. Like for the sawtooth form, the square form seems to have a resonnance filter when the pitch goes higher. Filter to determine.

• Sawtooth form:

The sawtooth form is described by the following function:

$$st(t) = sq(t) \times cos(w \times t + \epsilon)$$

where sq(t) is the square form defined above, w the angular frequency $(w = 2\pi \times f)$ and ϵ a slight shifting forward phase that occurs when the pitch is lower. Determine ϵ . In addition there is a slight resonance filter when the pitch gets higher. Filter to determine.

PCM sound generator

2.3 Chorus

The chorus of D-50 takes in input a mono signal and outputs a stereo signal.

2.4 Miscellaneous or not yet classified

• It seems that the only stereo in a patch comes from chorus and reverb.

Banks and patches

Most of the information is well detailed in the documentation of VC-1 so for the moment I see no open question about banks and patches format.

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