MS3: Direct Digital Synthesis

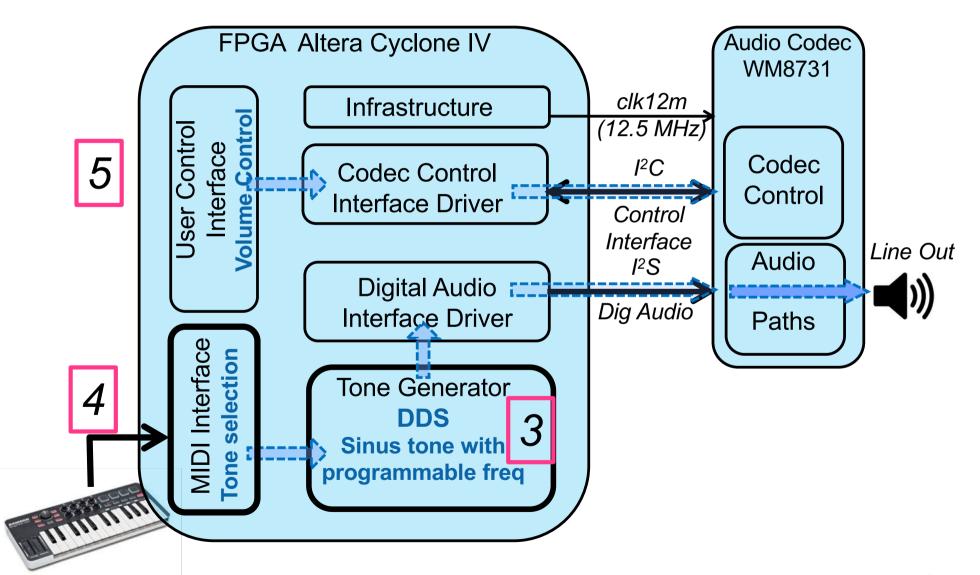


Direct Digital Synthesis (DDS)

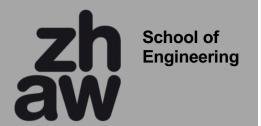
- Block Partitioning
- Basic Principle
- Implementation Example
- Counter-Length and Increment Calculation
- VHDL Implementation
- MIDI Integration

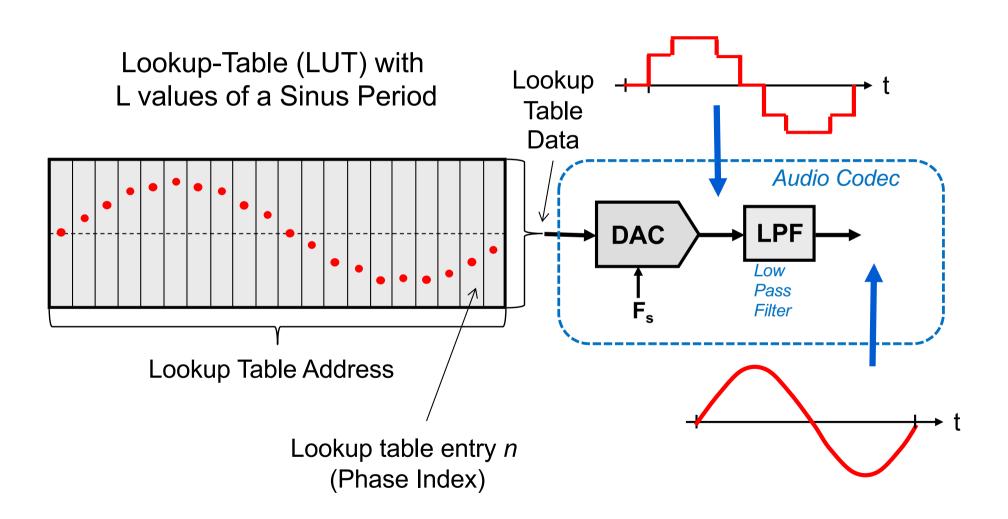
Milestones 3 bis 5 Tone Generator und MIDI-Interface





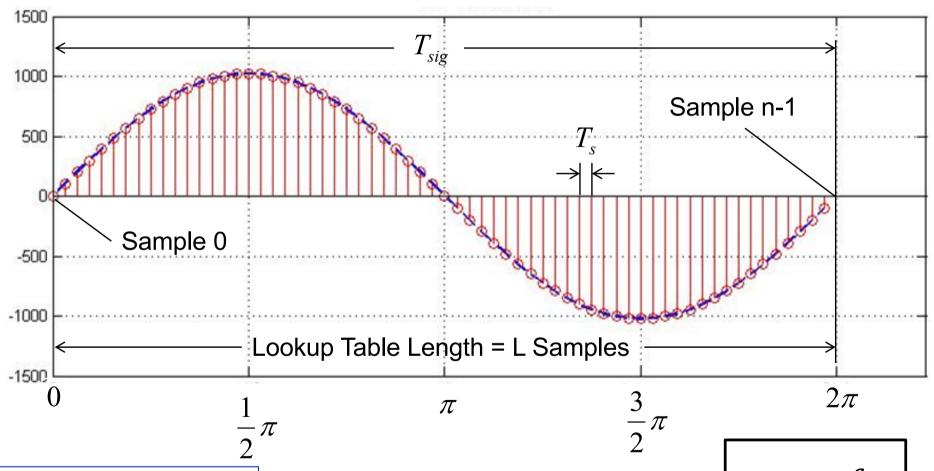
DDS: Basic Principle





Frequency of a synthesized Sinus





f_s= audio sample frequency L = length of LUT

n = index for phase counter f_{sig} = signal frequency

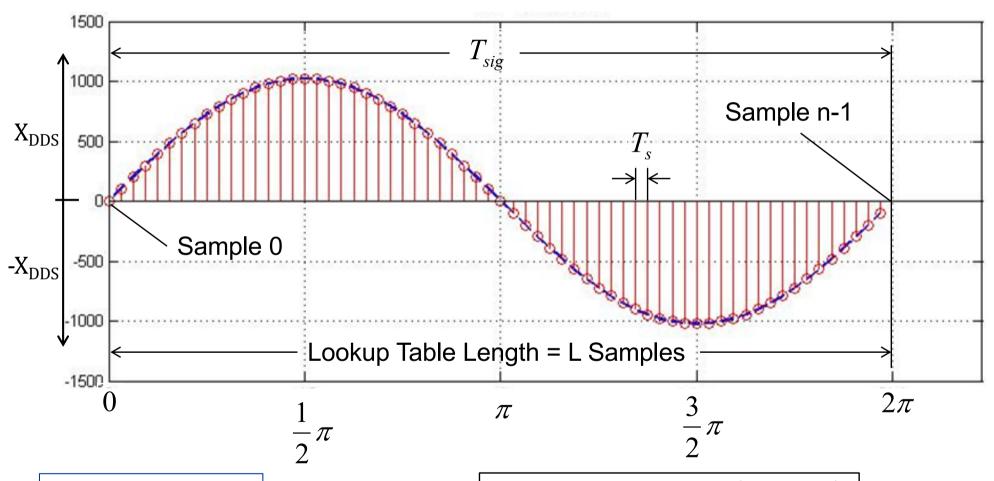
$$f_{sig} = \frac{1}{T_{sig}}$$

$$T_{sig} = L \cdot T_{s}$$

$$f_{sig} = \frac{fs}{L}$$

Amplitude of a Synthesized Sinus





L = length of LUT

n = index of sample

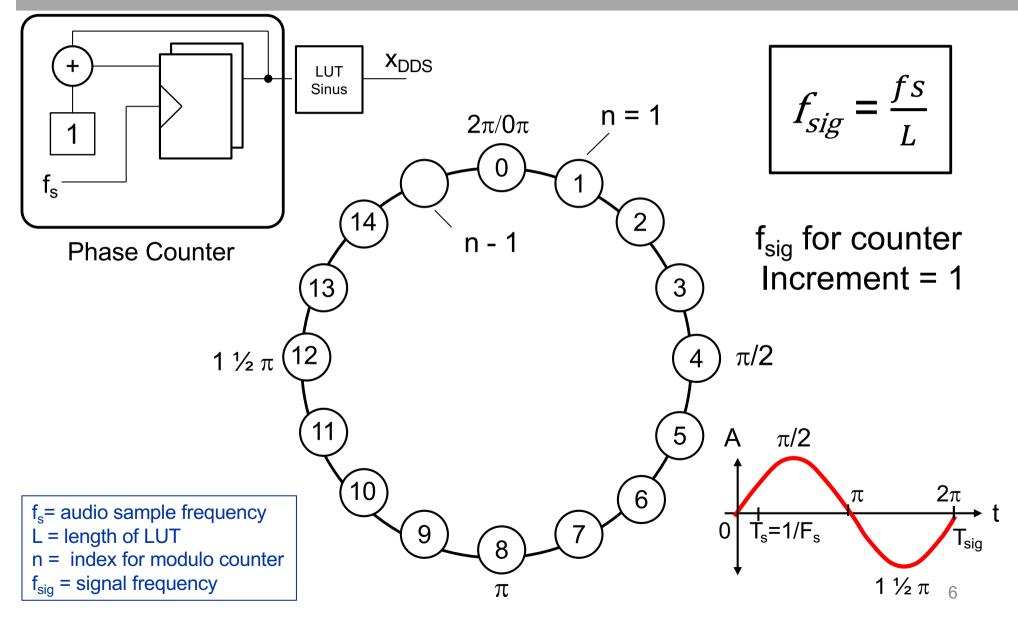
A = max. amplitude

$$x_{DDS}[n] = A \cdot \sin\left(n \cdot \frac{2\pi}{L}\right)$$

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LUT Adress Generation with Modulo Counter



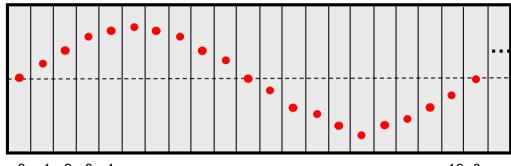


Example when LUT Length L=20



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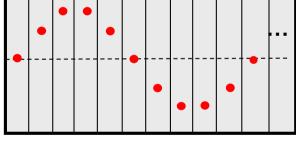




Lookup Table Address $n = 0 \quad 1 \quad 2 \quad 3 \quad 4 \dots$

19 0 ...

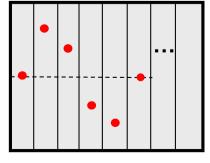
using $M=\Delta n=2$



Lookup Table Address n = 0 2 4 6 4

18 0 ...

using $M=\Delta n=4$



Lookup Table Address $n = 0 \ 4 \ 8 \ 12 \ 16 \ 0 \ \dots$

Controlling Frequency of fsig

M = phase increment value

n = index for phase counter

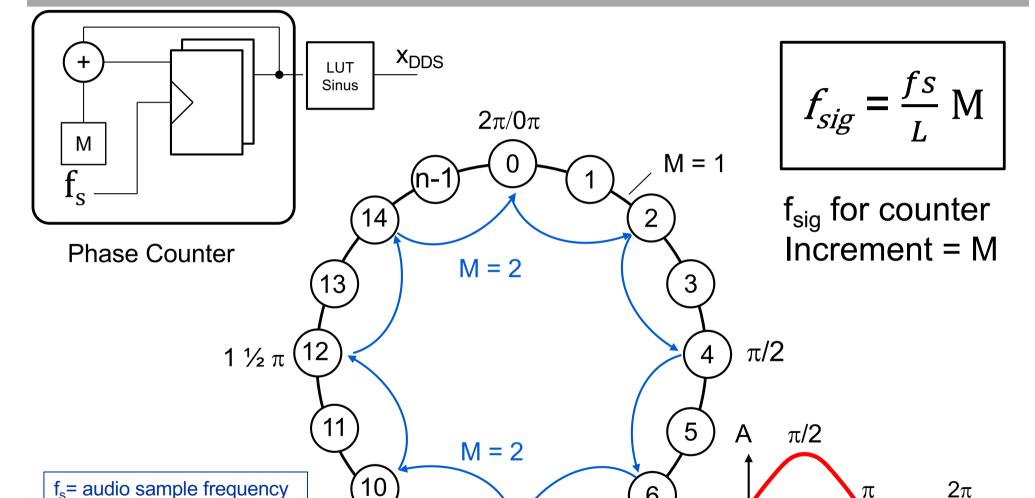
L = length of LUT

 f_{siq} = signal frequency



 $\dot{T}_s = 1/F_s$

1 ½ π 8

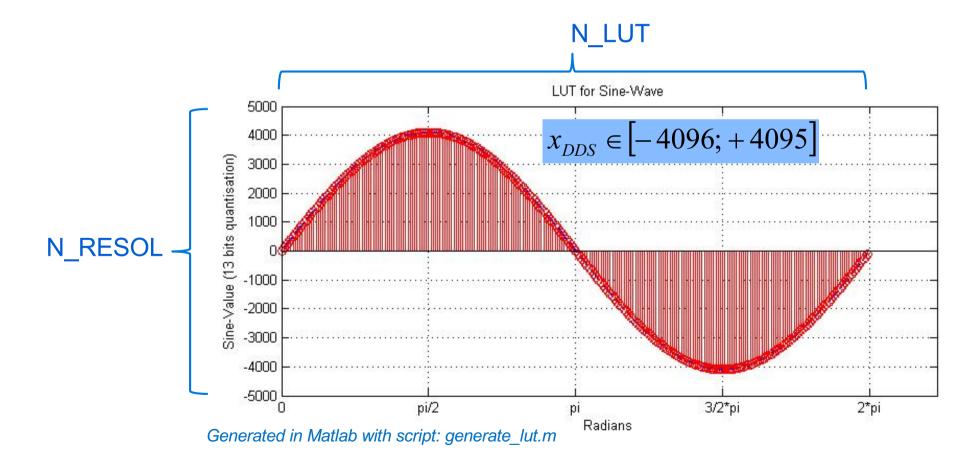


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Proposal for LUT Implementation

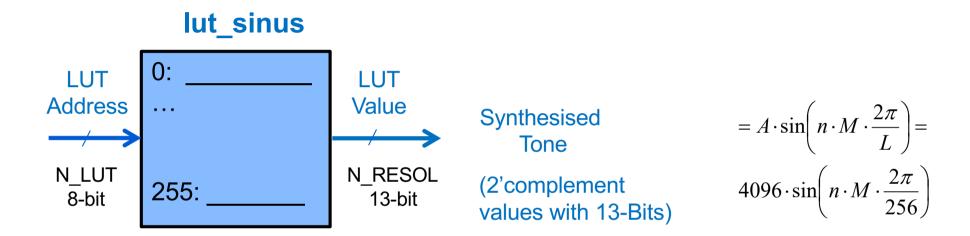


```
N_LUT = 8-bits (lut_sinus length L = 256 entries)
N_RESOL = 13-bits (signed = -4096..+4095)
```



Required ROM size





$$LUT_size = 2^8 \cdot 13 = 3'328 \ bits$$

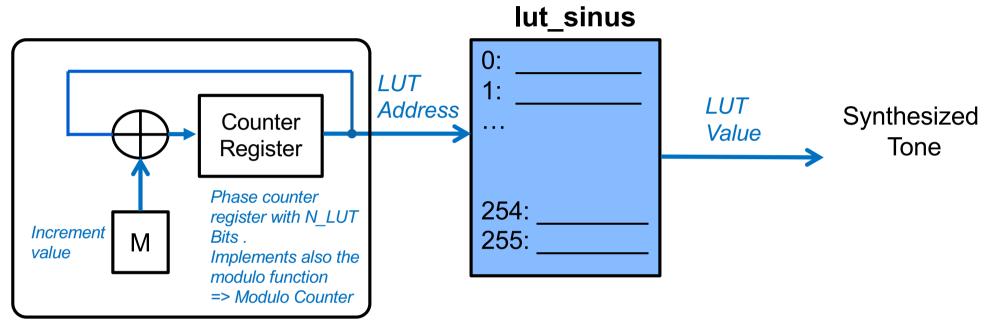
Available Memory Bits in 4CE-115 FPGA = 3'888'000 bits

Resulting Frequency Steps

when N_LUT = 8 bits



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Phase Counter

Which frequency is synthesised with update on phase counter every F_s = 48kHz and:

$$f_{sig} = F_s \cdot \frac{M}{L} = 48k \cdot \frac{M}{256}$$
 \Rightarrow M=1 ... $\Rightarrow f_{sig} = 187,5Hz$ \Rightarrow M=2 ... $\Rightarrow f_{sig} = 375Hz$ \Rightarrow M=3 ... $\Rightarrow f_{sig} = 562,5Hz$

Keyboard Tasten vs. Frequenzen

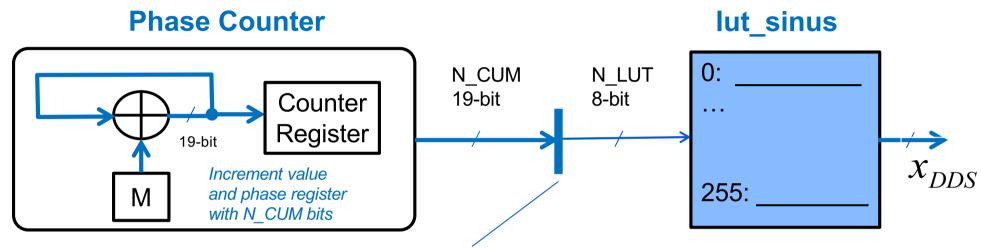


```
-- Piano Mid-Octave (white keys)
                                                                          C4
-- DO-C4 tone ~261.63Hz
-- RE D4 tone ~293.66Hz
-- MI E4 tone ~329.63Hz
-- FA F4 tone -349.23Hz
                                                                      A- 220 Hz A- 440 Hz A- 880 Hz A- 1760 Hz
-- SOL G4 tone ~392.00Hz
-- LA A4 tone ~440.00Hz
                                              FIGURE 22-4
-- SI B4 tone ~493.88Hz
                                              The Piano keyboard. The keyboard of the piano is a logarithwic frequency scale, with the fundamental frequency doubling after every seven white keys. These white keys are the notes: A, B, C, D, E, F and G.
-- DO C5 tone ~523.25Hz
-- Piano Mid-Octave (black keys)
   DOS C4S tone ~277.18Hz
    RES D4S tone ~311.13Hz
   FAS F4S tone ~369.99Hz
   SOLS G4S tone ~415.30Hz
-- LAS A4S tone ~466.16Hz
                                                                                    F
                                                                                E
                                                                                       GA
                                                                                                В
```

Making Counter Bits > N_LUT



Obs.: this means taking a fractional value for the increment!



Truncation:

Take only most significant 8-bits of N_CUM value for LUT Address

$$\Delta f = \frac{F_s}{2^{N-CUM}} = \frac{48k}{2^{19}} = 0,0916Hz$$

0.0916 Hz is precision of tone setting

Frequency steps when N_LUT = 19 bits Zn



Engineering

Which frequency is synthesised with update on phase counter every f_s = 48kHz and N LUT = 19 bits?

$$f_{sig} = f_s \cdot \frac{M}{L} = 48k \cdot \frac{M}{2^{19}}$$

$$ightharpoonup$$
 M=1 ... $\Rightarrow f_{sig} = 0.092Hz$

$$ightharpoonup$$
 M=2 ... $\Rightarrow f_{sig} = 0.183Hz$

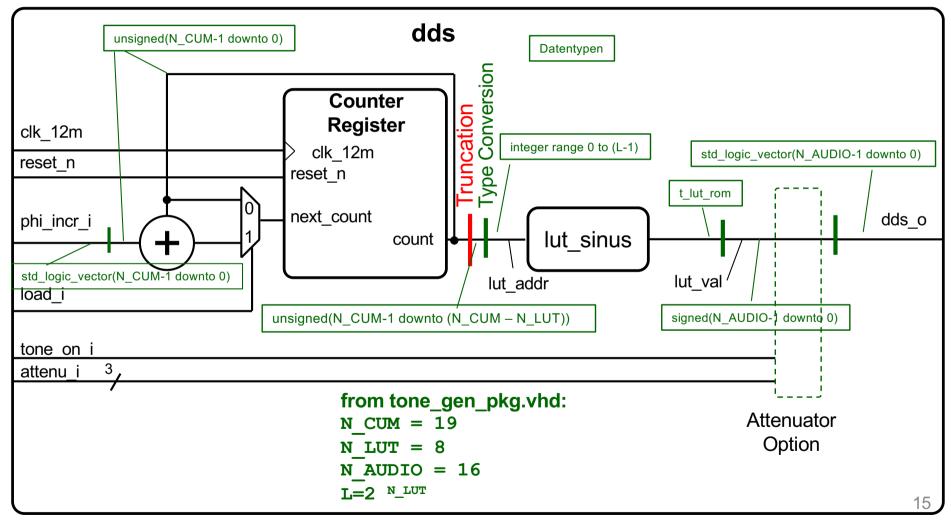
$$M = \frac{f_{sig} * L}{f_s}$$

Example:

$$M = \frac{f_{sig}*L}{f_s} = \frac{261.63Hz*2^{19}}{48kHz} = 2858$$

Aufbau DDS Block





¹⁾ Wenn load=1 wird das Counter Register mit dem Wert von next_count aktualisiert





```
-- Constant Declaration
 constant N LUT: natural :=8;
                                   -- number of bits in LUT address
               natural := 2**N LUT; -- length of LUT
 constant L:
 constant N_RESOL: natural := 13;
                                   -- Attention:1 bit reserved for sign
- Type Declaration
 subtype t audio range is integer range -(2**(N RESOL-1))to(2**(N RESOL-1))-1;
                                   -- range: [-2^12; +(2^12)-1]
 type t lut rom is array (0 to L-1) of t audio range;
 constant lut sinus : t lut rom :=(
  0, 101, 201, 301, 401, 501, 601,7 00, 799, 897, 995, ...
  ...501, 401, 301, 201, 101, 0, -101, -201, -301, -401, -501, -601, -700, ...);
```

Truncation of LUT Address



```
CONSTANT N_CUM:

natural :=19; --width counter register

constant N_LUT:

natural :=8; --width counter register

constant L:

natural :=2**N_LUT; -- length of LUT

...

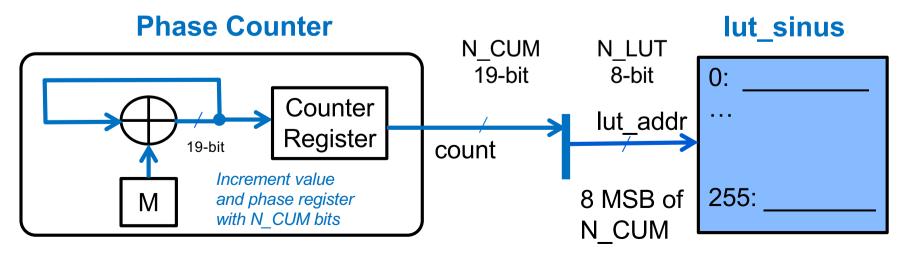
SIGNAL count:

unsigned(N_CUM-1 downto 0);

SIGNAL lut_addr :

integer range 0 to L-1;

lut_addr <= to_integer(count(N_CUM-1 downto N_CUM - N_LUT));
```

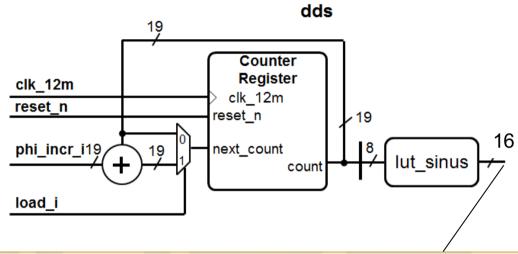


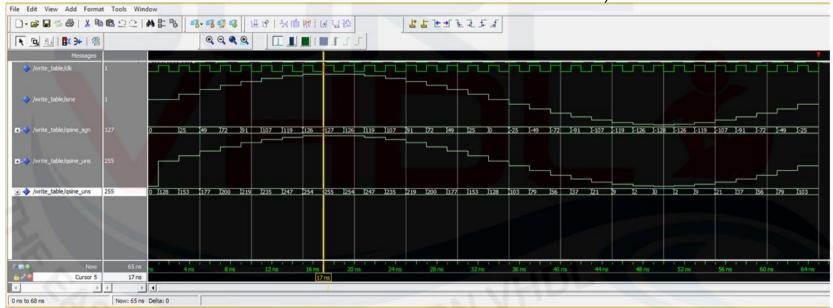
Audio Output (accessing lut_sinus) The School of Engineering

Syntax to grab N_LUT MSBs and use as address to search value in LUT remember to convert address to integer before using it as index!

Simulation of LUT Output







Attenuator

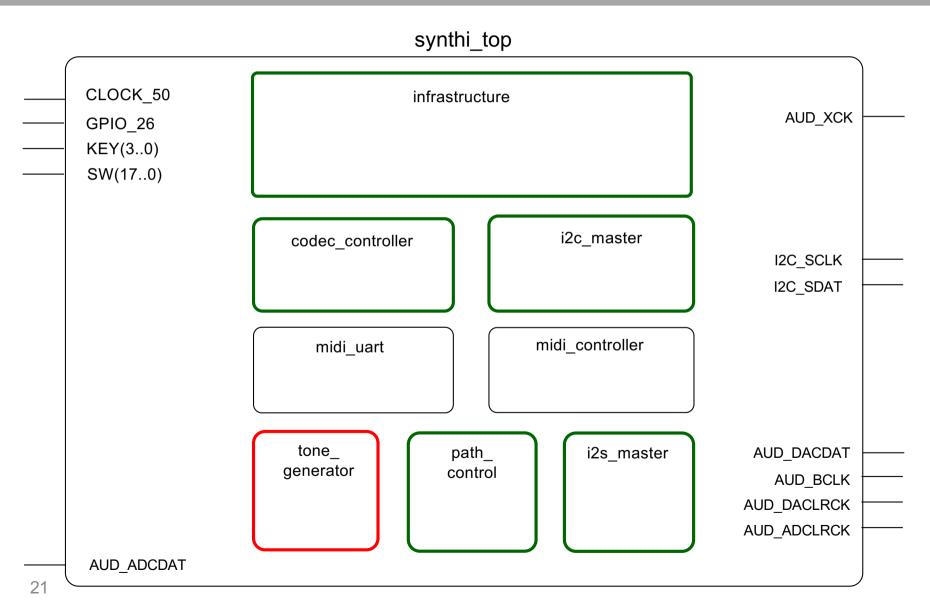


```
atte := to_integer(unsigned(attenu_i));

case atte is
when 0 => dds_o <= to_integer(lut_val);
when 1 => dds_o <= to_integer(shift_right(lut_val,1));
when 2 => dds_o <= to_integer(shift_right(lut_val,2))
...
when others => ...
end case;
```

Synthi Top MS3





MS3: Tone Generator



tone_generator clk 12m dds0 note vector reset n sw_sync(10..4)2) phi incr i lut_midi2dds dds_o tone on i 16 load i attenu i attenu_i 3) sw sync(16..17)²⁾ load i sw $sync(15)^{2}$ clk_12m dds1 2) Nur für Simulation reset n in synthi top phi_incr_i lut midi2dds mit sw sync dds_o tone on i verbinden. In load_i Testbench 1) Optional falls multiple DDS bei MS attenu i³⁾ sum reg SW mit gpi sim «Extra Features» implementiert verbinden. Signale werden nach clk_12m dds2 Implementierung reset n des midi controller phi_incr_i lut_midi2dds dds o ersetzt tone on i 3) Optional falls load i **Volume Control** attenu i³⁾

gewünscht

Aufbau lut_midi2dds



Increment values are pre-calculated and coded as constants in tone_gen_pkg.vhd

```
-- Piano Mid-Octave (white keys)
-- DO-C4 tone ~261.63Hz
-- RE D4 tone ~293.66Hz
-- MI E4 tone ~329.63Hz
                                                                  A- 220 Hz A- 440 Hz A- 880 Hz A- 1760 Hz
                                           A-27.5 Hz
-- FA F4 tone ~349.23Hz
-- SOL G4 tone ~392.00Hz
                                           FIGURE 22-4
-- LA A4 tone ~440.00Hz
                                           The Piano keyboard. The keyboard of the piano is a logarithwic frequency scale, with the fundamental frequency doubling after every seven white keys. These white keys are the notes: A, B, C, D, E, F and G.
-- SI B4 tone -493.88Hz
-- DO C5 tone ~523.25Hz
-- Piano Mid-Octave (black keys)
-- DOS C4S tone ~277.18Hz
-- RES D4S tone ~311.13Hz
-- FAS F4S tone ~369.99Hz
-- SOLS G4S tone ~415.30Hz
                                                                               F
                                                                           E
                                                                                   GA
                                                                                           В
-- LAS A4S tone ~466.16Hz
Example:
constant M LA A4: unsigned(N CUM-1 downto 0):= to unsigned(4806, N CUM));
-- LA A4 tone \sim 440.00Hz M = 2^{19} * 440/48000
```

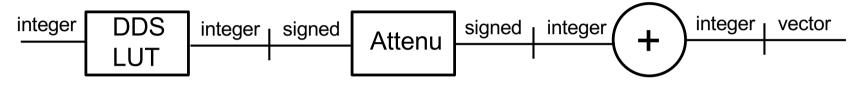
LUT_midi2dds



```
type t lut note number is array (0 to 127) of
                          std logic_vector(N_CUM-1 downto 0);
   constant LUT midi2dds : t lut note number :=(
           CM2 DO , -- Key 0
           CM2S_DOS, -- Key 1
           DM2 RE, -- Key 2
           DM2S_RES, -- Key 3
           EM2 MI, -- Key 4
           FM2_FA, -- Key 5
           FM2S FAS -- Key 6
           E8_MI, -- Key 124
F8_FA, -- Key 125
           F8S FAS, -- Key 126
           G8 SOL -- Key 127
           );
```

Umwandlungen im Signalverlauf





Add-In DDS Outputs for Polyphony



```
comb sum output : process(all)
  variable var sum : integer range -(2**(N AUDIO-1)) to (2**(N AUDIO-1))-1;
begin
  var sum := 0;
  if strobe i = '1' then
   dds sum loop: for i in 0 to 9 loop
     var sum := var sum + dds o(i);
   end loop dds sum loop;
   next sum req <= var sum;</pre>
  else
   next sum reg <= sum reg;</pre>
  end if:
end process comb sum output;
reg sum output : process(clk, rst n)
begin
  if rst n = '0' then
    sum req <= 0;
  elsif rising edge(clk) then
    sum reg <= next sum reg;</pre>
  end if:
end process reg sum output;
```