DSB Portfolio 1

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1 Draw the double sided spectrum of the amplitudes for the signal $\mathbf{x}(t)$

By using Eulers princible x(t) can be rewritten as:

$$\begin{array}{rcl} x(t) & = & 2 \cdot \frac{e^{i2\pi \cdot 1000t} + e^{i2\pi \cdot 1000t}}{2} + \frac{e^{i2\pi \cdot 3000t} + e^{i2\pi \cdot 3000t} + e^{i2\pi \cdot 3000t} + e^{i2\pi \cdot 19000t} + e^{i2\pi \cdot 19000t}}{2} \\ x(t) & = & e^{i2\pi \cdot 1000t} + e^{i2\pi \cdot 1000t} + \frac{1}{2} \cdot e^{i2\pi \cdot 3000t} + \frac{1}{2} \cdot e^{-i2\pi \cdot 3000t} + \frac{1}{2} \cdot e^{i2\pi \cdot 19000t} + \frac{1}{2} \cdot e^{-i2\pi \cdot 19000t} \\ \end{array}$$

Which gives us the coefficients:

$$c_{1} = 1$$

$$c_{-1} = 1$$

$$c_{3} = 0.5$$

$$c_{-3} = 0.5$$

$$c_{19} = 0.5$$

$$c_{-19} = 0.5$$
(2)

And the double sided spectrum:

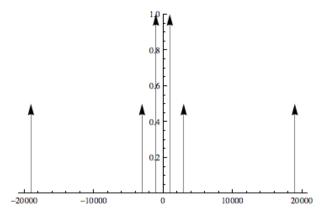


Figure 1: Double Sided Spectrum without aliasing

2 Draw the double sided spectrum xs1(t) for the sampled signal without the anti-aliasing filter

The samplerate of the A/D-converter is 20kHz. The new coefficients will be calculated by by multiplying them with the samplerate. Which gives:

$$c_1 = 20000$$

$$c_{-1} = 20000$$

$$c_3 = 10000$$

$$c_{-3} = 10000$$

$$c_{19} = 10000$$

$$c_{-19} = 10000$$

(3)

And a plot of the double sided spectrum:

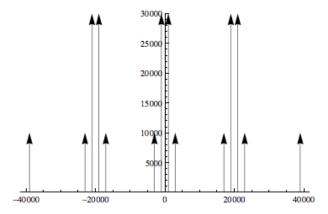


Figure 2: Double Sided Spectrum with aliasing

3 Calculate the A/D converter signal in dB

A/D-converter signal can be calculated by using:

$$SNR_{dB} = 10,79 + 20 \cdot Log_{10}\left(\frac{x_{rms}}{\Delta}\right) \tag{4}$$

Where:

$$\Delta = \frac{x_{max} - x_{min}}{2^m} \tag{5}$$

 x_{max} and x_{min} is the maximum and minimum voltage for the A/D-converter and m is the bitsize. x_{rms} is given by:

$$x_{rms} = \sqrt{\frac{1}{T} \int_0^T x(t)^2 dT} \tag{6}$$

We can then start calcualting:

$$\begin{array}{rcl} x_{rms} & = & \sqrt{3} \\ \Delta = \frac{2-(-2)}{2^8} & = & \frac{1}{64} \\ SNR_{dB} = 10,79 + 20 \cdot Log_{10}(\frac{\sqrt{3}}{\frac{1}{64}}) & = & 51,6848dB \end{array}$$

 SNR_{dB} is the strength of the signal.

4 Determine the quantization voltage for the A/D converter

Quantization is the steps in which the A/D-converter moves. It is the same as Δ which we calculated in the previous section:

$$\Delta = \frac{2 - (-2)}{2^8} = \frac{1}{64}$$

5 Determine the order of the filter for the analog Butterworth

The equation can be converted to dB and solved with regards to n:

$$-48dB = 20 \cdot log_{10} \left(\frac{1}{\sqrt{1 + (\frac{10000}{3000})^{2n}}} \right)$$

$$n = 4,58997 \sim 5$$

The order of the filter is 5.

6 Determine the damping of the amplitude of the analog signal with 1000, 3000 or 10000 Hz

$$20 \cdot Log_{10}\left(\frac{1}{\sqrt{1 + \left(\frac{1000}{3000}\right)^{2 \cdot 5}}}\right) = -0,00074dB$$

$$20 \cdot Log_{10}\left(\frac{1}{\sqrt{1 + \left(\frac{3000}{3000}\right)^{2 \cdot 5}}}\right) = -3,0103dB$$

$$20 \cdot Log_{10}\left(\frac{1}{\sqrt{1 + \left(\frac{10000}{3000}\right)^{2 \cdot 5}}}\right) = -52,2879dB$$

$$(7)$$

6

7 Draw the double-sided spectrum xs2 for the sampled signal with the designed anti-aliasing filter

Calculating magnitudes for frequencies 1000Hz, 3000Hz and 1900Hz:

$$\frac{1}{\sqrt{1 + (\frac{1000}{3000})^{2 \cdot 5}}} = 0,999992$$

$$\frac{1}{\sqrt{1 + (\frac{3000}{3000})^{2 \cdot 5}}} = 0,707107$$

$$\frac{1}{\sqrt{1 + (\frac{19000}{3000})^{2 \cdot 5}}} = 0,000098$$
(8)

These magnitudes are multiplied by the coefficients and the 10kHz sample frequency to get the real magnitudes:

$$0,999992 \cdot 1 \cdot 10kHz = 9999, 92$$

$$0,707107 \cdot 0, 5 \cdot 10kHz = 3535, 54$$

$$0,000098 \cdot 0, 5 \cdot 10kHz = 0,49$$
(9)

Double sided spectrum plot. The third signal is left out as it is so small compared to the two others:

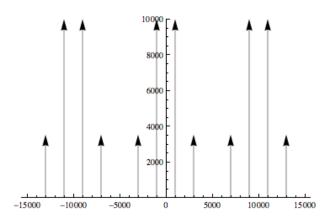


Figure 3: Double Sided Spectrum

8 Calculate the filter order n for an reconstruction filter of Butterworth type

By adapting and making two equations we can solve two variables:

$$-0.5dB \leq 20 \cdot Log_{10}\left(\frac{1}{\sqrt{1 + \left(\frac{3000}{f_c}\right)^{2n}}}\right)$$

$$20 \cdot Log_{10}\left(\frac{1}{\sqrt{1 + \left(\frac{10000}{f_c}\right)^{2n}}}\right) = 20 \cdot Log_{10}\left(\frac{1}{\sqrt{1 + \left(\frac{3000}{f_c}\right)^{2n}}}\right)$$

$$n = 5, 5 \sim 6$$

$$f_c = 3636, 87$$

$$(10)$$

The order of the filter is 6. We then find the limits of f_c :

$$-48dB \leq 20 \cdot Log_{10}\left(\frac{1}{\sqrt{1 + (\frac{10000}{f_c})^{2 \cdot 6}}}\right)$$

$$f_c \leq 3981, 08$$

$$-0, 5dB \geq 20 \cdot Log_{10}\left(\frac{1}{\sqrt{1 + (\frac{3000}{f_c})^{2 \cdot 6}}}\right)$$

$$f_c \geq 3574, 81$$
(11)

 f_c must lie between 3574,81 and 3981,08. f_c is set to 3700. And the new filter:

$$20 \cdot Log_{10}\left(\frac{1}{\sqrt{1 + (\frac{3000}{3700})^{2 \cdot 6}}}\right) \tag{12}$$

9 Determine the damping of the amplitude in dB of the reconstructed signal

Damping of the frequencies 1000Hz and 3000Hz.

$$20 \cdot Log_{10}\left(\frac{1}{\sqrt{1 + (\frac{1000}{3700})^{2 \cdot 6}}}\right) = -6,597^{-7}dB$$

$$20 \cdot Log_{10}\left(\frac{1}{\sqrt{1 + (\frac{3000}{3700})^{2 \cdot 6}}}\right) = -0,337^{-7}dB$$
(13)