# Effects of finite word length representation of the filter coefficients

Lab 8, SDP

## **Objective**

Students should observe the effects of having fixed point coefficients in a digital filter, and be able to mitigate the effects.

#### Theoretical notions

#### **Exercises**

1. Convert the following binary number to the decimal value:

11011.0101

2. Convert in binary fixed point format (signed, 6 integer bits, 6 fractionary bits - 1S6I6F the following numbers:

273.21875

- 3. Convert in binary fixed point format (signed, 6 integer bits, 6 fractionary bits 1S6I6F the following negative numbers. Negative numbers shall be represented in sign-value, 1's complement (C1) and 2's complement (C2) formats.
  - a. -273
  - b. -273.21875
- 4. Quantize the samples  $x_1 = 0.42625$  and  $x_2 = -0.4333$  the fixed point format 1S0I4F via:
  - a. Truncation
  - b. Rounding
  - c. Truncation in absolute value

The negative values shall be represented in C2 format.

- 5. Use Matlab's fdatool to design a low-pass IIR filter, Butterworth type, order 4, with cutoff frequency of 4kHz for a sampling frequency of 44.1kHz. Export the coefficients of the direct form II implementation to the Matlab Workspace as b and a.
- 6. In Matlab's fdatool, set the filter arithmetic to "fixed-point arithmetic" and modify the following:
  - a. Set the format to fixed point 1S2I7F. How does the filter's transfer function change?
  - b. Increase the number of bits in the fractionary part. How does the filter's transfer function change? For what number of bits do you consider the errors to be negligible?
  - c. Export the coefficients of the direct form II implementation to Matlab's Workspace as b1 and a1.
- 7. Repeat the preceding exercise with the filter implemented in series form ("Second-Order-Sections"). Which implementation has smallest errors? Export the coefficients to Matlab's Workspace as b2 and a2.
- 8. Load the mtlb audio signal from Matlab. Use filter() to filter the signal with the original filter (b and a) and with the fixed point coefficients (b1 and a1).
  - a. Plot the difference between the two filtered signals.
  - b. Plot the histogram of the difference signal. What is it's shape? What is the average value of the errors?

### **Final questions**

1. TBD