

Implementing digital FIR filters in the lattice form

Lab 5, SDP

Objective

The students should become familiar with *lattice*-type realization structure used for implementing FIR filters.

Theoretical notions

Exercises

1. Find the FIR filter coefficients in direct form, if the reflection coefficients of the lattice FIR structure are: $K_1 = \frac{1}{2}$, $K_2 = 0.6$, $K_3 = -0.7$, $K_4 = \frac{1}{3}$.
2. Find the reflection coefficients of the lattice structure for a FIR filter with system function:

$$H(z) = 1 + \frac{2}{5}z^{-1} + \frac{7}{20}z^{-2} + \frac{1}{2}z^{-3}$$

3. Using the Octave software, use the `fir1()` function to design one of the following FIR filters:
 - a. A low-pass FIR filter of order 4, with cutoff frequency of 5kHz at a sampling frequency of 44.1kHz;
 - b. A high-pass FIR filter of order 4, with cutoff frequency of 2kHz at a sampling frequency of 44.1kHz;
 - c. A band-pass FIR filter of order 4, with passband between 1kHz and 3kHz at a sampling frequency of 44.1kHz.

Read the documentation of the `fir1()` function to find out how to use it.

4. **Not needed; function given.** Create an Octave function to `tf2latc()` to compute the coefficients of the lattice form of a FIR filter, starting from the coefficients of the Transfer Function. Call it like this: `K = tf2lact(coef)`

5. Create an Octave script to filter an input signal `x` with a FIR filter in lattice form, for which the reflection coefficients `K` are known:

```
y = filter_latc(x, K)
```

6. Use the function above to load and low-pass the audio signal `Kalimba.mp3`.

- a) Load the file using `audioread()`
- b) Use `tf2latc()` to convert the filter to lattice form
- c) Filter the signal with `filter_latc()`

Final questions

1. TBD