# Neuroprothetik – Exercise 1: Introduction

## 1 Generate a Signal

The following Matlab code generates a signal following eq. 1 and takes into account following input arguments: array of frequencies in Hz, array of amplitudes, signal duration in s, sampling rate in Hz.

$$f(t) = A_0 + \sum_{i=1}^{n} A_i \cdot \sin(2\pi F_i \cdot t)$$
(1)

#### 1.1 Plot the signal

a)

See code.

b)

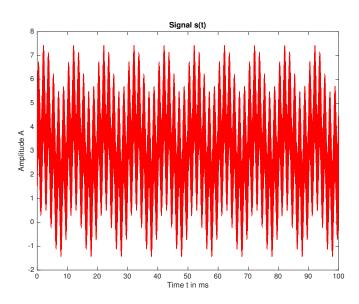


Figure 1: Sinusoidal signal s(t) consisting of superposed frequencies 100 Hz, 600 Hz and 9 kHz with amplitudes 1, 1.5 and 2 as well as an offset of 3 at a sampling rate of 100 kHz.

# 2 Calculate the Spectrum

**a**)

See code.

### 2.1 Plot the Spectrum

**a**)

In the following plots you see the amplitude spectrum for the different sampling rates (100 kHz, 20 kHz, 10 kHz).

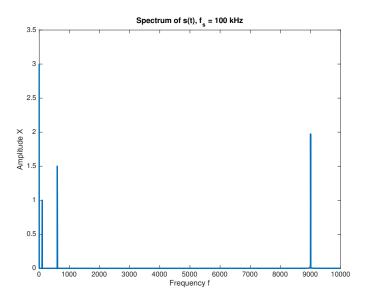


Figure 2: Amplitude spectrum for signal from section 1, with sampling rate of 100 kHz.

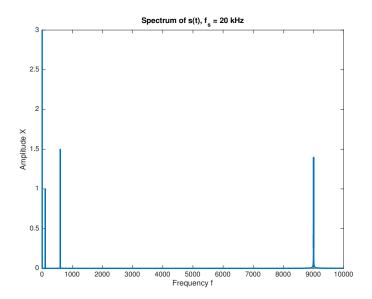


Figure 3: Amplitude spectrum for signal from section 1, with sampling rate of 20 kHz.

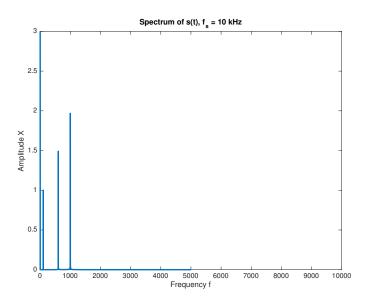


Figure 4: Amplitude spectrum for signal from section 1, with sampling rate of 10 kHz.

b)

For figures 1 and 2: the peak at f = 0 Hz is the offset with the correct amplitude A = 3. The next peak from the left is at f = 100 Hz (1<sup>st</sup> frequency component of the signal) with its amplitude A = 1. At f = 600 Hz is the 2<sup>nd</sup> frequency component of the signal with A = 1.5. The last peak at f = 9 kHz with an amplitude of A = 2 is the 3<sup>rd</sup> frequency component of the signal.

For figure 3, there is no peak at f = 9 kHz, but an additional peak at f = 1 kHz. That peak was shifted from the *left-side spectrum*, as the FFT spectrum initially was symmetric, but only by  $f_s = 10$  kHz, thus resulting in f = 1 kHz ((10-9) kHz = 1 kHz), see also following paragraph).

The last spectrum (the one that was calculated based on the signal sampled at a rate of 10 kHz) violates the Nyquist-Shannon sampling theorem. The Nyquist-Shannon sampling theorem states that for a given bandlimit B of a signal, the sampling frequency  $f_s$  needs to be at least twice the bandlimit:

$$2 \cdot B < f_s \tag{2}$$

This condition is **not fulfilled**, as the maximum frequency in signal s(t) is 9 kHz, i.e.:

$$2 \cdot 9 \text{ kHz} = 18 \text{ kHz} < 10 \text{ kHz}$$
 (3)

Therefore, aliasing artifacts occur in the amplitude spectrum of signal  $s_3(t)$  between the two main spikes at 100 Hz and 600 Hz. The maximum admissible frequency in any signal s(t) must not exceed  $f_s/2$  for proper sampling and perfect reconstruction of the signal.

Edit (11/05/2016): The necessary