

Data and Signal Analysis

6th Exercise Sheet
Winter Semester 2024/2025

Submission unit 31/01/2025 1 p.m. CET

1. Exercise: Signal Analysis (8 Points)

The file `signal_b5.txt` contains a signal composed of various wave packets. Find all the contained frequencies. Proceed as follows:

- Plot the signal and the corresponding spectral power density.
- Calculate the dynamic spectrum. What time resolution provides the best representation of the spectrum?
- Perform a wavelet analysis. Use the Mexican-Hat wavelet and the Morlet wavelet.
- Interpret the results and explain the advantage of the different methods. What should be considered when selecting the mother wavelet?

Use the `cwt` function of the `PyWavelets` library (external package!) to perform the continuous wavelet transform. See: <https://pywavelets.readthedocs.io/en/latest/index.html>

2. Exercise: Magnetopause (4 Points)

In the file `mag_data_themis.txt`, you will find the time information, as well as the x , y , and z components of the magnetic field (in the GSE coordinate system), measured by a THEMIS satellite during its passage through the magnetopause. The direction \vec{n} of the minimum variance

$$\sigma^2 = \frac{1}{N} \sum_{i=1}^N \left[\left(\vec{B}_i - \langle \vec{B} \rangle \right) \cdot \vec{n} \right]^2 \quad (1)$$

subject to the constraint $|\vec{n}|^2 = 1$ corresponds to the normal direction of the local magnetopause surface.

- Physically justify the above statement.
- Apply the minimum variance analysis to the THEMIS time series and thus determine the direction of the magnetopause normal in the GSE coordinate system.

3. Exercise: Filter (8 Points)

In the file `sinus13.txt`, you will find a signal composed of 3 superimposed sinusoidal

oscillations; the sampling period is 1 s. The objective is to retrieve the 3 individual signals in the time domain. To do this, the signal must be filtered using a low-pass, a high-pass, and a band-pass filter.

- a) Graphically represent the entire signal.
- b) Perform a Fourier transform on the signal. Also calculate the amplitude spectrum and plot the result.
- c) Multiply the Fourier-transformed signal by the system response function $H(f)$ for an ideal low-pass filter. This is constructed as follows: For frequencies $|f| < f_{co}$ (cut-off frequency), it is 1, otherwise 0. Note that you need a system response function of the same length as the signal, and that the Fourier transform of a real-valued function is mirrored in the middle and complex conjugated; the second half of the spectrum corresponds to the negative frequencies (not necessary with a real FFT). The cut-off frequency should be chosen so that only the sinusoidal oscillation of the lowest frequency is not filtered out.
- d) Inversely transform the modified Fourier-transformed signal back to the time domain and plot the result. What has changed?
- e) Now also inversely transform the system response function $H(f)$ and plot the result ($h(n)$). How can you tell that the ideal low-pass filter is not causal?
- f) Repeat steps (c) and (d) using an ideal high-pass and band-pass filter to extract the other two signal components from the entire signal.