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Digital Signal Processing Tutorial 1 & 2, Winter Semester 2019/20 (Course #24505)
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Spectral Analysis of Deterministic Signals

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1 DFT

Why just another DFT tutorial? Well, this is a collection of calculus, graphs and exercises that the authors found most useful for teaching purposes in the last decade. The graphs and the didactical approach following [Mös11] might be a way of DFT storytelling that is rarely found in classic textbooks and deserve survival.

1.1 DFT Definitions

The discrete Fourier transform (DFT, often interpreted as signal analysis) and its counterpart the inverse DFT (IDFT, often interpreted as signal synthesis) are defined as [Opp10, eq. (8.11) and (8.12)]

$$X[\mu] = \sum_{k=0}^{N-1} x[k] \cdot W_N^{k\mu} \quad x[k] = \frac{1}{N} \sum_{\mu=0}^{N-1} X[\mu] \cdot W_N^{-k\mu} \quad (1)$$

using the so-called twiddle factor or DFT kernel

$$W_N = e^{-j\frac{2\pi}{N}} \quad (2)$$

relating an N samples discrete-time signal $x[k]$ and its N coefficients (also called bins) discrete-frequency DFT spectrum $X[\mu]$. In general, the DFT/IDFT supports $x[k] \in \mathbb{C}$ and $X[\mu] \in \mathbb{C}$. Later, some useful relations will be obtained when assuming $x[k] \in \mathbb{R}$.

In the literature, the DFT/IDFT pair may be differently defined, such as e.g.

$$X[\mu] = \underbrace{\frac{1}{N}}_{K_{\text{DFT}}} \sum_{n=0}^{N-1} x[k] \cdot W_N^{k\mu} \circ \bullet \quad x[k] = \underbrace{1}_{K_{\text{IDFT}}} \cdot \sum_{\mu=0}^{N-1} X[\mu] \cdot W_N^{-k\mu} \quad (3)$$

$$X[\mu] = \underbrace{\frac{1}{\sqrt{N}}}_{K_{\text{DFT}}} \sum_{n=0}^{N-1} x[k] \cdot W_N^{k\mu} \circ \bullet \quad x[k] = \underbrace{\frac{1}{\sqrt{N}}}_{K_{\text{IDFT}}} \sum_{\mu=0}^{N-1} X[\mu] \cdot W_N^{-k\mu}, \quad (4)$$

where different normalisation schemes are applied. However, a valid DFT/IDFT-pair must always fulfill

$$K = K_{\text{IDFT}} \cdot K_{\text{DFT}} = \frac{1}{N}, \quad (5)$$

so that

$$x[k] = \text{IDFT} \{ \text{DFT} \{ x[k] \} \} \quad (6)$$

$$X[\mu] = \text{DFT} \{ \text{IDFT} \{ X[\mu] \} \} \quad (7)$$

holds. Hence, either K_{IDFT} or K_{DFT} can be chosen according to the specific application and then the other parameter is determined by K . Another convention is related to the sign of $\exp()$ in the twiddle factor. This may also be defined as

$$W_N = e^{+j\frac{2\pi}{N}} \quad (8)$$

if required for specific applications. Note that the DFT/IDFT equations do not care about signal interpretation, they transform a vector to another vector. The general definition of the DFT/IDFT-pair is thus given as

$$X[\mu] = K_{\text{DFT}} \sum_{k=0}^{N-1} x[k] \cdot W_N^{\pm k\mu} \quad x[k] = K_{\text{IDFT}} \sum_{\mu=0}^{N-1} X[\mu] \cdot W_N^{\mp k\mu} \quad (9)$$

using

$$W_N = e^{-j\frac{2\pi}{N}}, \quad K_{\text{IDFT}} \cdot K_{\text{DFT}} = \frac{1}{N}. \quad (10)$$

In the majority of DSP-related books and software the DFT/IDFT pair is defined as

$$X[\mu] = \sum_{k=0}^{N-1} x[k] \cdot e^{-j\frac{2\pi}{N}k\mu} \quad \text{Python numpy } \mathbf{X}=\text{np.fft.fft}(\mathbf{x}), \text{ Matlab: } \mathbf{X}=\text{fft}(\mathbf{x}) \quad (11)$$

$$x[k] = \frac{1}{N} \sum_{\mu=0}^{N-1} X[\mu] \cdot e^{+j\frac{2\pi}{N}k\mu} \quad \text{Python numpy } \mathbf{x}=\text{np.fft.ifft}(\mathbf{X}), \text{ Matlab: } \mathbf{x}=\text{ifft}(\mathbf{X}) \quad (12)$$

which will also be used throughout the lecture and the tutorial. This convention implies that positive constant group delays in the DFT spectrum X are interpreted as causal signal delays for x .

The discrete-time Fourier transform (DTFT) pair

$$X(\Omega) = \sum_{k=-\infty}^{\infty} x[k] \cdot e^{-j\Omega k} \quad (13)$$

$$x[k] = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(\Omega) \cdot e^{+j\Omega k} d\Omega. \quad (14)$$

is defined with the same signs in the $\exp()$ and the prefactor $\frac{1}{2\pi}$ belonging to the synthesis equation in the same way as for the used definition of the DFT pair.

1.2 Matrix Notation and Use of Linear Algebra Fundamentals

The DFT is a classic application for a fundamental linear algebra problem, i.e. solving a set of linear equations, or in another train of thought: transferring one vector to another vector in hope for more convenient data representation, such as here for spectrum analysis. And of course it is not pure hope which helps us, but the appropriate choice of a vector base that (approximately) solves our problem. It is thus worth to grasp these links for an in-depth understanding.

The DFT of length N is a brilliant example for a special base of orthogonal vectors that are set up from the roots of the $z^N = 1$ equation, i.e. equiangularly distributed locations along the unit circle within the complex plane, cf. Fig. 7. To build these vectors, let us recall the twiddle factor

$$W = e^{-j\frac{2\pi}{N}}, \quad (15)$$

here omitting the N subscript to avoid confusion with the following subscripts that indicate matrix and vector dimensions. Now, by inspecting the DFT analysis equation (11) one might

figure what pairs of k and μ are required and bring them into a certain sequence, namely putting them into an appropriate matrix. This so called DFT matrix $\mathbf{W}_{N \times N}$ can be built from the outer product (note the symmetry)

$$\mathbf{A}_{N \times N} = \begin{bmatrix} 0 \\ 1 \\ 2 \\ 3 \\ \vdots \\ N-1 \end{bmatrix} \cdot [0 \quad 1 \quad 2 \quad 3 \quad \dots \quad N-1] = \begin{bmatrix} 0 & 0 & 0 & 0 & \dots & 0 \\ 0 & 1 & 2 & 3 & \dots & (N-1) \\ 0 & 2 & 4 & 6 & \dots & 2(N-1) \\ 0 & 3 & 6 & 9 & \dots & 3(N-1) \\ \vdots & & \dots & & & \vdots \\ 0 & (N-1) & 2(N-1) & 3(N-1) & \dots & (N-1)^2 \end{bmatrix}$$

applying it to the twiddle factor by element-wise operation

$$\mathbf{W}_{N \times N} = (e^{-j\frac{2\pi}{N}})^{\mathbf{A}} = e^{-j\frac{2\pi}{N}\mathbf{A}}. \quad (16)$$

The reader might verify that this matrix is orthogonal (note that the complex dot product is required here). Then, the DFT (11) can be written in matrix notation as

$$\begin{bmatrix} X[\mu=0] \\ X[\mu=1] \\ X[\mu=2] \\ \vdots \\ X[\mu=N-1] \end{bmatrix} = \begin{bmatrix} W^{0 \cdot 0} & W^{0 \cdot 1} & W^{0 \cdot 2} & \dots & W^{0 \cdot (N-1)} \\ W^{1 \cdot 0} & W^{1 \cdot 1} & W^{1 \cdot 2} & \dots & W^{1 \cdot (N-1)} \\ W^{2 \cdot 0} & W^{2 \cdot 1} & W^{2 \cdot 2} & \dots & W^{2 \cdot (N-1)} \\ \vdots & & \dots & & \vdots \\ W^{(N-1) \cdot 0} & W^{(N-1) \cdot 1} & W^{(N-1) \cdot 2} & \dots & W^{(N-1) \cdot (N-1)} \end{bmatrix} \begin{bmatrix} x[k=0] \\ x[k=1] \\ x[k=2] \\ \vdots \\ x[k=N-1] \end{bmatrix}, \quad (17)$$

shortly as

$$\mathbf{X}_{N \times 1} = \mathbf{W}_{N \times N} \mathbf{x}_{N \times 1}. \quad (18)$$

Note that linear algebra typical notates matrices with uppercase letters and vectors with lowercase letters. Common signal processing convention uses lowercase letters for signals and uppercase letters for spectrum. Following the latter \mathbf{X} is not a matrix, but a vector containing the DFT spectrum.

The inverse of the DFT matrix reads

$$\mathbf{W}^{-1} = \frac{1}{N} \mathbf{W}^H = \frac{1}{N} \mathbf{W}^*, \quad (19)$$

with H denoting conjugate-complex and transpose operation. This constitutes a special case for complex valued, symmetric matrix \mathbf{W} . Since \mathbf{W} is also square, no transpose is required, but only the conjugate complex operation $()^*$, which is just the sign reversal in the $\exp()$ of twiddle factor. If the matrices are normalized, such that

$$\left(\frac{1}{\sqrt{N}} \mathbf{W}^H\right) \left(\frac{1}{\sqrt{N}} \mathbf{W}\right) = \mathbf{I} \quad (20)$$

then they are unitary, i.e. the complex conjugate transpose of the complex square matrix is at the same time the inverse. This is a key feature for DFT.

The IDFT is then solving the set of equations $\mathbf{X} = \mathbf{W}\mathbf{x}$ for a given \mathbf{X} and an unknown \mathbf{x} via

$$\mathbf{x} = \mathbf{W}^{-1} \mathbf{X} = \frac{1}{N} \mathbf{W}^* \mathbf{X} \quad (21)$$

which precisely corresponds to (12). Furthermore, analysis (DFT) and subsequent synthesis (IDFT) yields the input vector again, such as

$$\mathbf{x} = \frac{1}{N} \mathbf{W}^* (\mathbf{W} \mathbf{x}) = \left(\frac{1}{N} \mathbf{W}^* \mathbf{W} \right) \mathbf{x} = \mathbf{x}, \quad (22)$$

due to $\frac{1}{N} \mathbf{W}^* \mathbf{W} = \mathbf{I}$. This is identical to the statement (6).

The reader might verify that the following DFT/IDFT pairs with $N = 4$ pairs hold. The input vector $\mathbf{x} = [1, 1, 1, 1]^T$ (pure DC component) yields $\mathbf{X} = [4, 0, 0, 0]^T$ (energy only at 'zero'-th frequency). The spectrum $\mathbf{X} = [1, 1, 1, 1]^T$ (all frequencies exhibit equal energy) yields $\mathbf{x} = [1, 0, 0, 0]^T$ (the discrete Dirac impulse).

1.3 FFT as a Fast Calculation of the DFT

The DFT/IDFT equations include redundant calculation steps due to the twiddle factor characteristics. A straightforward implementation for large N demands very high computing load. For example, most obviously the same $W_N^{k\mu}$ is derived for $\mu = 1, k = 2$ and $\mu = 2, k = 1$, cf. the entries of the matrices \mathbf{A} and \mathbf{W} . For large N , many angles $\frac{2\pi}{N} k\mu$ are equivalent. This is exemplarily shown in Fig. 1, Fig. 2 and Fig. 3 for $N = 8$, $N = 9$ and $N = 48$ DFTs. These plots indicate certain symmetries, which is to be expected since the underlying matrix has symmetry as well.

Algorithms that make use of these symmetries in order to reduce or simplify computation steps are subsumed as Fast Fourier Transform (FFT). Explained in a sloppy way, FFTs calculate certain repeatedly used twiddle factor results only once. The first proposed FFT algorithm is from Cooley and Tukey [Coo65], that relies on $N = 2^m$, $m \in \mathbb{N}$. Nowadays, many other and improved FFT algorithms exist, so that if N is accessible for a prime factorisation, an FFT can be calculated with much less processing load than a DFT. In fact, the invention of fast calculation of the DFT is a (if not the) milestone in DSP making these huge technological steps in the last decades of the digital age.

The FFT concept can be derived from the linear algebra problem finding a suitable factorization for \mathbf{W} that involves sparse matrices and plain vector permutations (which not need to be calculated as matrix operations) and thus reduces multiplications/additions. You will reinvent this basis concept in one of the homework assignments.

1.4 DFT Frequency Resolution

The relation between physical temporal frequency f , sampling frequency f_s and discrete-time angular frequency Ω is known as

$$\Omega = 2\pi \frac{f}{f_s} = \frac{\omega}{f_s}. \quad (23)$$

Furthermore, the twiddle factor and the DFT matrix tell us that the unit circle is equiangularly sampled, i.e. the angular frequency resolution

$$\Delta\Omega = \frac{2\pi}{N} \quad (24)$$

holds. From both equations

$$\Delta\Omega = 2\pi \frac{\Delta f}{f_s} = \frac{\Delta\omega}{f_s} \equiv \Delta\Omega = \frac{2\pi}{N} \quad (25)$$

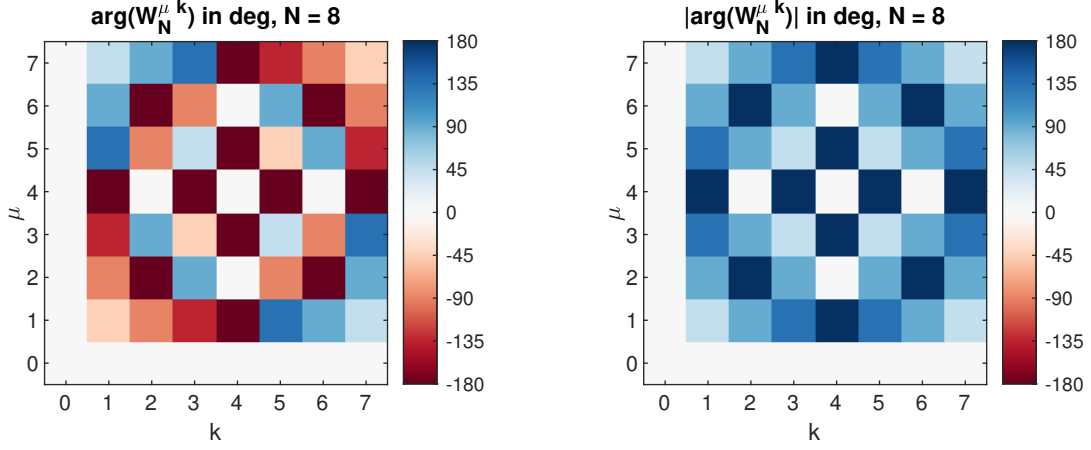


Figure 1: $\arg(W_N^{\mu k})$ (left) and $|\arg(W_N^{\mu k})|$ (right) as matrix over μ and k for $N = 8$.

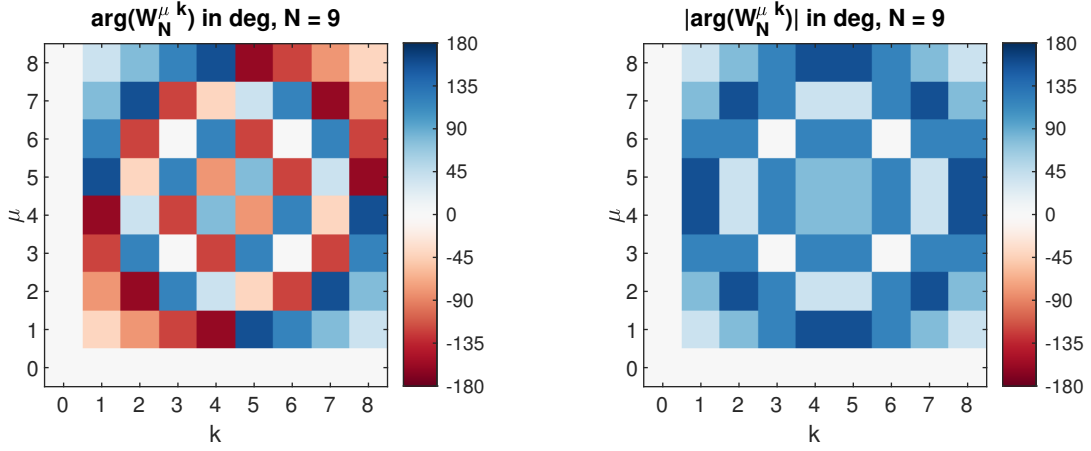


Figure 2: $\arg(W_N^{\mu k})$ (left) and $|\arg(W_N^{\mu k})|$ (right) as matrix over μ and k for $N = 9$.

we can deduce the DFT resolution in terms of the physical frequency f as

$$\Delta f = \frac{f_s}{N}. \quad (26)$$

This corresponds to the frequency distance between two spectral lines, i.e. between two bins. From that, all so called eigenfrequencies of the DFT, i.e. frequencies where the DFT bins are located, can be derived as

$$f_{\text{DFT}} = \mu \Delta f = \mu \frac{f_s}{N} \quad \text{for} \quad 0 \leq \mu \leq N-1, \mu \in \mathbb{N}. \quad (27)$$

These DFT eigenfrequencies can also be given for the discrete-time angular frequency

$$\Omega_{\text{DFT}} = \mu \Delta \Omega = \mu \frac{2\pi}{N} = \frac{2\pi f_{\text{DFT}}}{f_s}. \quad (28)$$

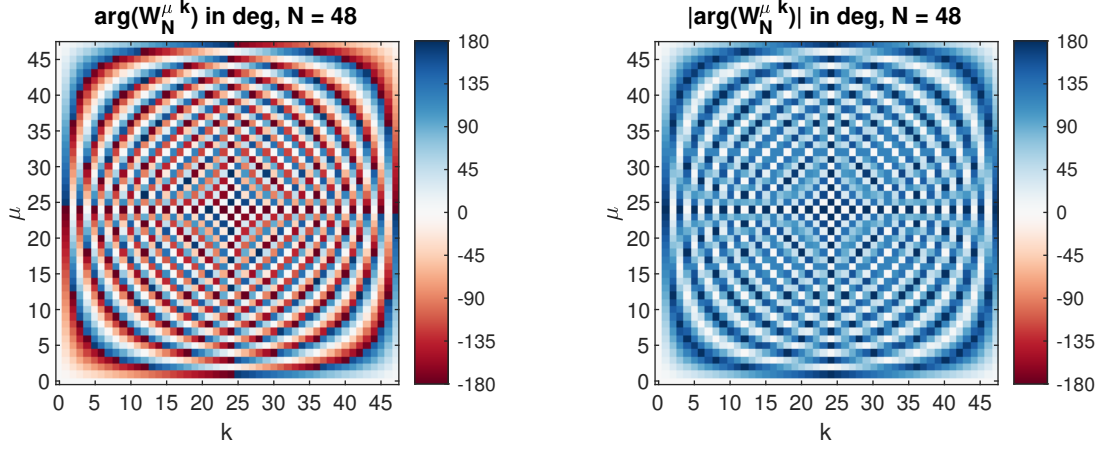


Figure 3: $\arg(W_N^{\mu k})$ (left) and $|\arg(W_N^{\mu k})|$ (right) as matrix over μ and k for $N = 48$.

1.5 Periodicity and Symmetry of the DFT

The signals $x[k]$ and $X[\mu]$ of an DFT/IDFT pair exhibit a periodicity of N . This is shown with $k, \mu, k', m \in \mathbb{Z}$:

$$x[k'] = \frac{1}{N} \sum_{\mu=0}^{N-1} X[\mu] e^{j \frac{2\pi}{N} k' \mu} \quad (29)$$

$$x[k' = k + mN] = \frac{1}{N} \sum_{\mu=0}^{N-1} X[\mu] e^{j \frac{2\pi}{N} (k+mN) \mu} \quad (30)$$

$$= \frac{1}{N} \sum_{\mu=0}^{N-1} X[\mu] e^{j \frac{2\pi}{N} k \mu} \cdot \underbrace{e^{j \frac{2\pi}{N} mN \mu}}_{=1} = x[k]. \quad (31)$$

A similar proof yields the identity $X[\mu] = X[\mu + mN]$, cf. Fig. 4. This is equivalent to a DTFT spectrum that exhibits a 2π periodicity, i.e. $X(\Omega) = X(\Omega + m \cdot 2\pi)$. The baseband of the DFT for $0 \leq \mu \leq N - 1$ corresponds to the spectrum of the signal $x[k]$ for $0 \leq k \leq N - 1$. Thus, both signals $x[k]$ and $X[\mu]$ inherently exhibit periodicity with N .

A further very important characteristic of the DFT spectrum is observed, when $x[k] \in \mathbb{R}$ (such as audio and video signals) is assumed. Then the symmetries

$$\operatorname{Re} \left\{ X \left[\frac{N}{2} + m \right] \right\} = \operatorname{Re} \left\{ X \left[\frac{N}{2} - m \right] \right\}, \quad \operatorname{Im} \left\{ X \left[\frac{N}{2} + m \right] \right\} = -\operatorname{Im} \left\{ X \left[\frac{N}{2} - m \right] \right\}, \quad (32)$$

$$\left| X \left[\frac{N}{2} + m \right] \right| = \left| X \left[\frac{N}{2} - m \right] \right|, \quad \arg \left(X \left[\frac{N}{2} + m \right] \right) = -\arg \left(X \left[\frac{N}{2} - m \right] \right) \quad (33)$$

hold that can be written shortly as

$$X \left[\frac{N}{2} + m \right] = X \left[\frac{N}{2} - m \right]^*, \quad (34)$$

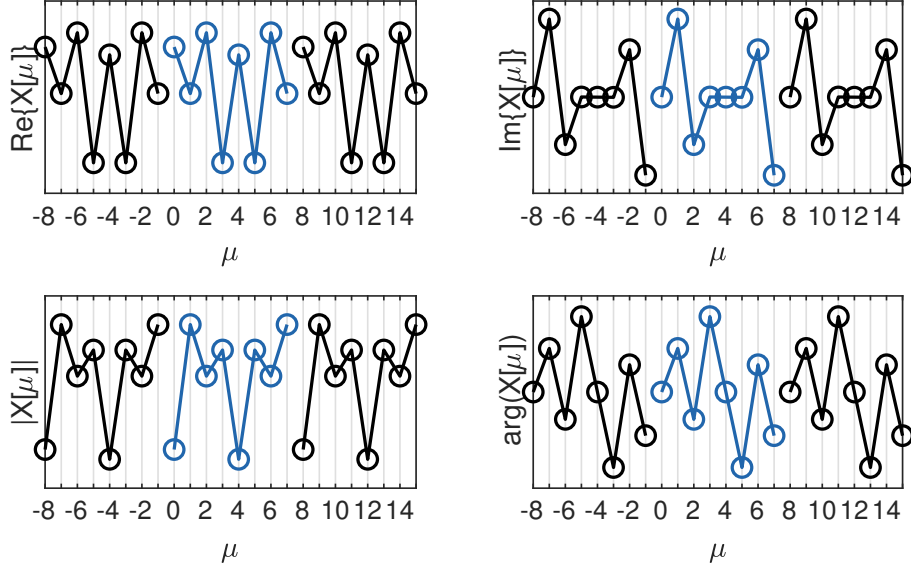


Figure 4: Periodicity of the DFT for a real signal with $N = 8$, blue: baseband, black: periodic repetitions.

with $()^*$ again denoting conjugate-complex operation. Think of N as being an even number here, so $m \in \mathbb{Z}$. The case of odd-numbered N is considered later. The symmetry property of eq. (34) is shown by inserting and simplifying

$$X \left[\mu = \frac{N}{2} + m \right] = \sum_{k=0}^{N-1} x[k] e^{-j \frac{2\pi}{N} k (\frac{N}{2} + m)} = \sum_{k=0}^{N-1} x[k] e^{-j\pi k} \cdot e^{-j \frac{2\pi}{N} km}, \quad (35)$$

$$X \left[\mu = \frac{N}{2} - m \right] = \sum_{k=0}^{N-1} x[k] e^{-j \frac{2\pi}{N} k (\frac{N}{2} - m)} = \sum_{k=0}^{N-1} x[k] e^{-j\pi k} \cdot e^{j \frac{2\pi}{N} km}. \quad (36)$$

The sum of two complex numbers $a, b \in \mathbb{C}$ is just $(a+b)^* = a^* + b^*$. Recall that $e^{-j\pi k} = \pm 1 \in \mathbb{R}$. With the assumption $x[k] \in \mathbb{R}$ it can then be proved

$$X \left[\mu = \frac{N}{2} - m \right]^* = \left(\sum_{k=0}^{N-1} x[k] e^{-j\pi k} \cdot e^{j \frac{2\pi}{N} km} \right)^* \quad (37)$$

$$= \sum_{k=0}^{N-1} x[k] e^{-j\pi k} \cdot \left(e^{j \frac{2\pi}{N} km} \right)^* \quad (38)$$

$$= \sum_{k=0}^{N-1} x[k] e^{-j\pi k} e^{-j \frac{2\pi}{N} km} \quad (39)$$

$$= X \left[\mu = \frac{N}{2} + m \right]. \quad (40)$$

Since the discrete DFT spectrum is only defined at $\mu \in \mathbb{Z}$ m must be defined as follows

$$m = \begin{cases} M & \text{for even } N \\ M + \frac{1}{2} & \text{for odd } N \end{cases} \quad (41)$$

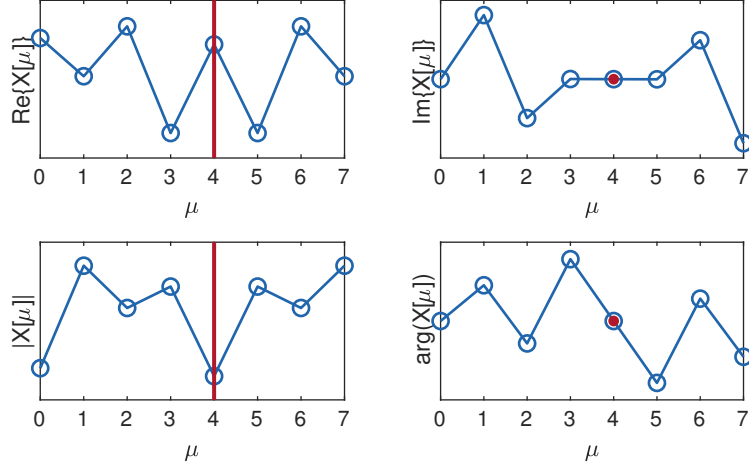


Figure 5: Symmetry of the DFT for $x[k] \in \mathbb{R}$ with $N = 8$. Point and axis for symmetry is at $\frac{N}{2} = 4$. Left: real part and magnitude of $X[\mu]$ are axisymmetric, right: imaginary part and phase of $X[\mu]$ are point-symmetric.

using $M \in \mathbb{Z}$, cf. Fig. 5 and Fig. 6. Due to the periodicity of $X[\mu]$ in general and the special $\frac{N}{2}$ symmetries of $X[\mu]$ when $x[k] \in \mathbb{R}$ (point-symmetric for imaginary part and phase, axisymmetric for real part and magnitude, cf. again Fig. 5 and Fig. 6) some information is redundant in the DFT spectrum. For the interpretation of the DFT spectrum, only

$$\begin{aligned} M &= \frac{N}{2} + 1 & \text{for even } N \\ M &= \frac{N+1}{2} & \text{for odd } N \end{aligned} \quad (42)$$

bins contain non-redundant information. This corresponds to the frequency band from DC ($f = 0$ Hz) to half the sampling frequency ($f_s/2$). It is defined for **even** N as

$$0 \leq \mu \Delta f \leq \frac{f_s}{2} \quad \text{for} \quad 0 \leq \mu \leq \frac{N}{2} \quad (43)$$

and for **odd** N as

$$0 \leq \mu \Delta f < \frac{f_s}{2} \quad \text{for} \quad 0 \leq \mu \leq \frac{N-1}{2}. \quad (44)$$

Thus, for even N the symmetry axis $\mu = \frac{N}{2}$ is exactly at the bin indicating half of the sampling frequency

$$f = \mu \frac{f_s}{N} = \frac{N}{2} \frac{f_s}{N} = \frac{f_s}{2}, \quad (45)$$

whereas for odd N the symmetry axis is located between the two bins around $\frac{f_s}{2}$. Therefore, an odd N DFT does not include the 'half of the sampling frequency' bin, cf. Fig. 7, Fig. 8.

1.6 DFT as Sampling of the DTFT

The following discussion of the discrete Fourier transform (DFT) as a sampled DTFT and explanations concerning windowing were inspired by the brilliant script [Mös11] and became extended. A very similar didactic concept is approached in [Kam02, Ch. 7.3, 7.4]

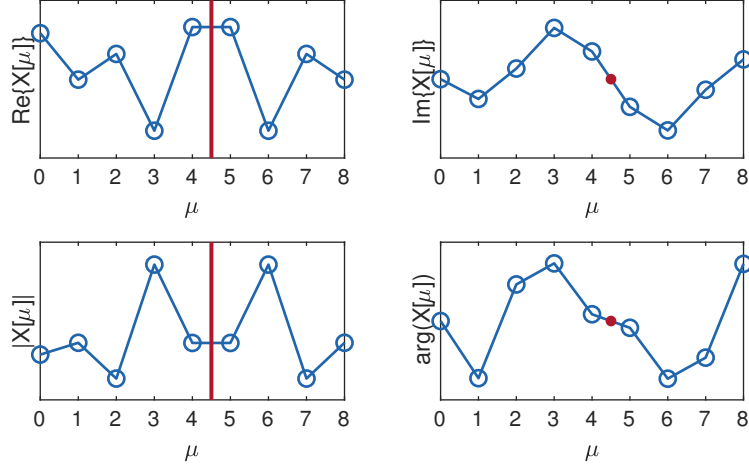


Figure 6: Symmetry of the DFT for $x[k] \in \mathbb{R}$ with $N = 9$. Point and axis for symmetry is at $\frac{N}{2} = 4.5$, i.e. between two DFT bins. Left: real part and magnitude of $X[\mu]$ are axisymmetric, right: imaginary part and phase of $X[\mu]$ are point-symmetric.

The DFT contains all possible spectral information that can be derived from the signal. Since N time samples are available, only maximum N (complex valued) spectral samples are available; for $x[k] \in \mathbb{R}$ less than N samples contain non-redundant information as shown above. The DFT can be derived by sampling the DTFT spectrum equiangularly on the unit circle with $\Delta\Omega = \frac{2\pi}{N}$, cf. Fig. 7, Fig. 8. Thinking of the DFT $X[\mu]$ as being the spectrum of the non-zero part of a sampled one-time signal $x[k]$ for $0 \leq k \leq N-1$, $x[k] = 0$ for all other k (so-called *single signal model* [Mös11]), inversely allows for interpolation towards the DTFT spectrum. To this end, the synthesis equation for the one-time signal eq. (12)

$$x[k] = \frac{1}{N} \cdot \sum_{\mu=0}^{N-1} X[\mu] \cdot e^{j\frac{2\pi}{N}k\mu} \quad (46)$$

is inserted into the analysis equation of the DTFT eq. (13) and gets rearranged:

$$X(\Omega) = \sum_{k=-\infty}^{\infty} x[k] \cdot e^{-j\Omega k} \quad (47)$$

$$= \sum_{k=0}^{N-1} \frac{1}{N} \cdot \sum_{\mu=0}^{N-1} X[\mu] \cdot e^{j\frac{2\pi}{N}k\mu} \cdot e^{-j\Omega k} \quad (48)$$

$$= \sum_{\mu=0}^{N-1} X[\mu] \cdot \frac{1}{N} \cdot \sum_{k=0}^{N-1} e^{-jk(\Omega - \frac{2\pi}{N}\mu)}. \quad (49)$$

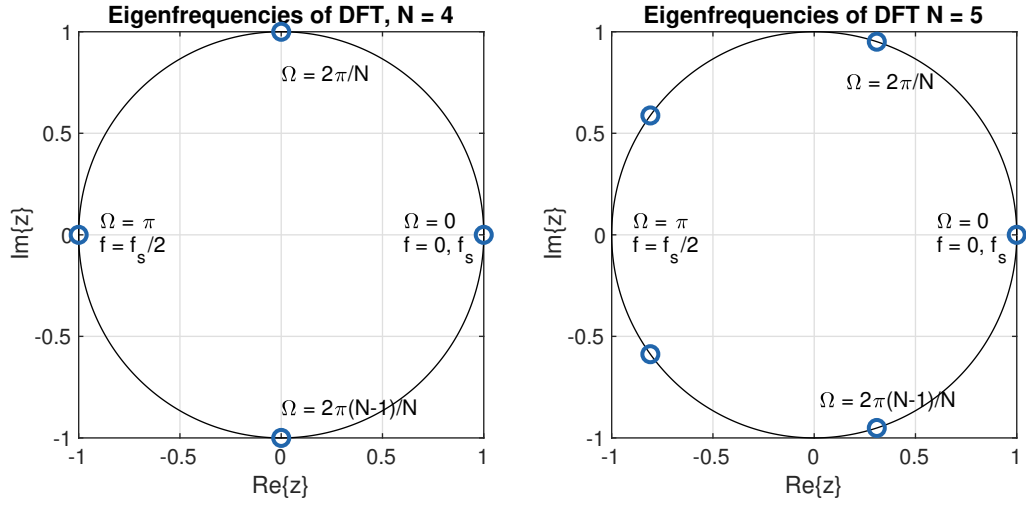


Figure 7: DFT eigenfrequency locations on the unit circle for $N = 4$ (left), $N = 5$ (right).

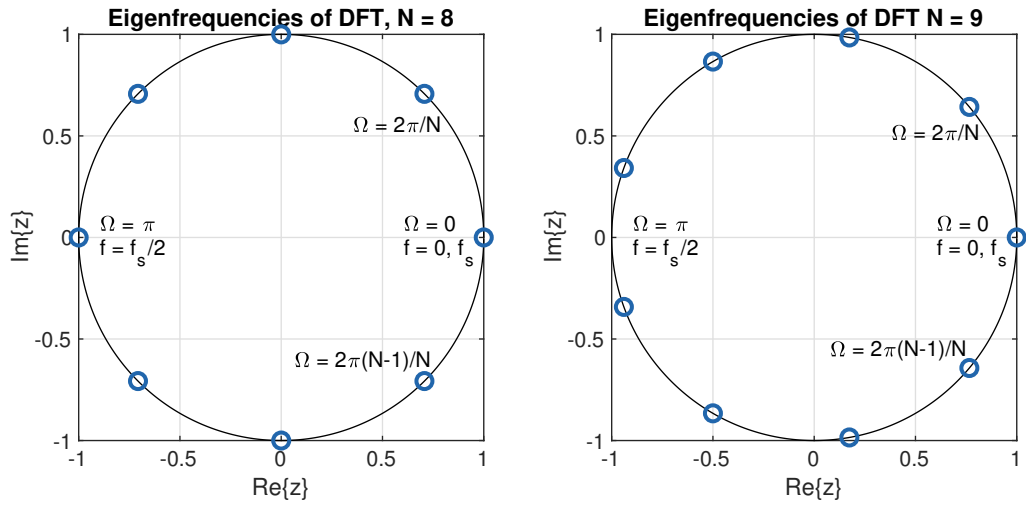


Figure 8: DFT eigenfrequency locations on the unit circle for $N = 8$ (left), $N = 9$ (right).

The sum over k is a geometric series and can be rearranged with [Lyo11, (3-39)] to

$$X(\Omega) = \sum_{\mu=0}^{N-1} X[\mu] \cdot \frac{1}{N} \cdot \frac{1 - e^{-j(\Omega - \frac{2\pi}{N}\mu)N}}{1 - e^{-j(\Omega - \frac{2\pi}{N}\mu)}} \quad (50)$$

$$= \sum_{\mu=0}^{N-1} X[\mu] \cdot \frac{1}{N} \cdot \frac{e^{-j\frac{(\Omega - \frac{2\pi}{N}\mu)N}{2}}}{e^{-j\frac{\Omega - \frac{2\pi}{N}\mu}{2}}} \cdot \frac{e^{j\frac{(\Omega - \frac{2\pi}{N}\mu)N}{2}} - e^{-j\frac{(\Omega - \frac{2\pi}{N}\mu)N}{2}}}{e^{j\frac{\Omega - \frac{2\pi}{N}\mu}{2}} - e^{-j\frac{\Omega - \frac{2\pi}{N}\mu}{2}}} \quad (51)$$

$$= \sum_{\mu=0}^{N-1} X[\mu] \cdot \frac{1}{N} \cdot e^{-j\frac{(\Omega - \frac{2\pi}{N}\mu)(N-1)}{2}} \cdot \frac{e^{j\frac{(\Omega - \frac{2\pi}{N}\mu)N}{2}} - e^{-j\frac{(\Omega - \frac{2\pi}{N}\mu)N}{2}}}{e^{j\frac{\Omega - \frac{2\pi}{N}\mu}{2}} - e^{-j\frac{\Omega - \frac{2\pi}{N}\mu}{2}}}. \quad (52)$$

With the Euler identity $2j \cdot \sin(x) = e^{jx} - e^{-jx}$ this can be simplified to [Mös11, eq. (2.41)]

$$X(\Omega) = \sum_{\mu=0}^{N-1} X[\mu] \cdot e^{-j\frac{(\Omega - \frac{2\pi}{N}\mu)(N-1)}{2}} \cdot \frac{1}{N} \cdot \frac{\sin\left(N\frac{\Omega - \frac{2\pi}{N}\mu}{2}\right)}{\sin\left(\frac{\Omega - \frac{2\pi}{N}\mu}{2}\right)}. \quad (53)$$

The interpolation kernel is the so-called aliased or periodic sinc function

$$\text{psinc}_N(\Omega) = \begin{cases} \frac{1}{N} \cdot \frac{\sin(\frac{N}{2}\Omega)}{\sin(\frac{1}{2}\Omega)} & \text{for } \Omega \neq 2\pi m, \\ (-1)^{m(N-1)} & \text{for } \Omega = 2\pi m \end{cases}, \quad m \in \mathbb{Z}, \quad (54)$$

in Matlab and Python's `scipy.diric(Omega,N)`, and a phase shift. Therefore, eq. (53) can be written as

$$X(\Omega) = \sum_{\mu=0}^{N-1} X[\mu] \cdot e^{-j\frac{(\Omega - \frac{2\pi}{N}\mu)(N-1)}{2}} \cdot \text{psinc}_N\left(\Omega - \frac{2\pi}{N}\mu\right). \quad (55)$$

Thus, a DTFT value at any Ω can be interpolated using eq. (55). It is easy to show, that for $\Omega = \frac{2\pi}{N}\mu'$ this evaluates exactly to the DFT value since this is a DFT eigenfrequency:

$$X\left(\Omega = \frac{2\pi}{N}\mu'\right) = \sum_{\mu=0}^{N-1} X[\mu] \cdot e^{-j\frac{(\frac{2\pi}{N}\mu' - \frac{2\pi}{N}\mu)(N-1)}{2}} \cdot \frac{1}{N} \cdot \frac{\sin\left(N\frac{\frac{2\pi}{N}\mu' - \frac{2\pi}{N}\mu}{2}\right)}{\sin\left(\frac{\frac{2\pi}{N}\mu' - \frac{2\pi}{N}\mu}{2}\right)}, \quad (56)$$

$$= \sum_{\mu=0}^{N-1} X[\mu] \cdot e^{-j\frac{\pi(\mu' - \mu)(N-1)}{N}} \cdot \frac{1}{N} \cdot \frac{\sin(\pi(\mu' - \mu))}{\sin\left(\frac{\pi}{N}(\mu' - \mu)\right)}. \quad (57)$$

For all $\mu \neq \mu'$ the term $\sin(\pi(\mu' - \mu)) = 0$. For $\mu = \mu'$ the periodic sinc evaluates to 1 and $e^{-j\frac{\pi(\mu' - \mu)(N-1)}{N}} = 1$. Thus, the interpolation yields

$$X\left(\Omega = \frac{2\pi}{N}\mu'\right) = X[\mu'], \quad (58)$$

and it can be seen that the DTFT and DFT spectrum are identical at all DFT eigenfrequencies

$$\Omega_{\text{DFT}} = \frac{2\pi}{N}\mu \quad \text{for } 0 \leq \mu \leq N-1, \mu \in \mathbb{N}. \quad (59)$$

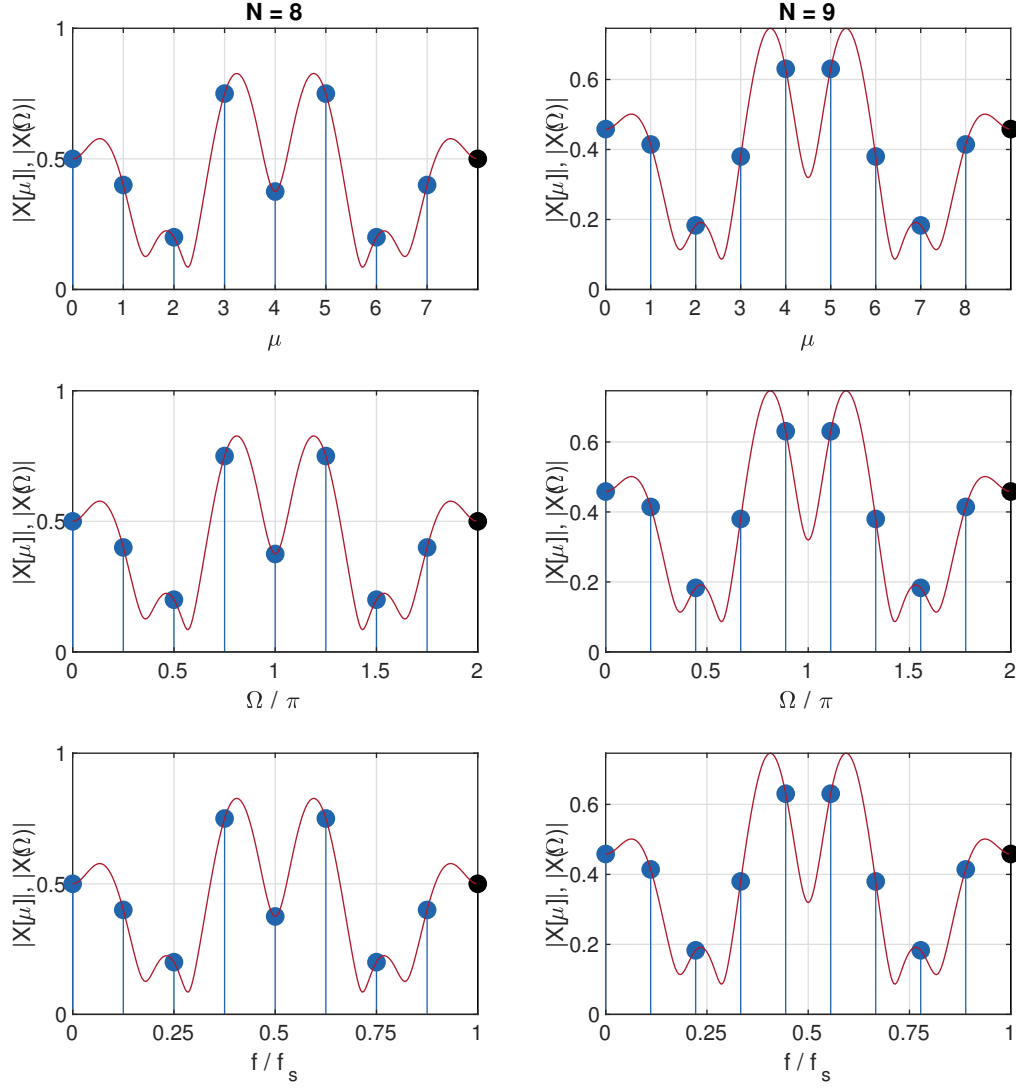


Figure 9: DFT (stem, blue) and interpolated DTFT (line, red) magnitude spectra over μ (top), Ω/π (middle) and $\frac{f}{f_s}$ (bottom) for $N = 8$ (left) and $N = 9$ (right).

For all other frequencies, the DTFT spectrum follows the interpolation kernel, cf. Fig. 9. As already introduced in eq. (28)

$$\Omega = \frac{2\pi f}{f_s} \quad \Omega_{\text{DFT}} = \frac{2\pi f_{\text{DFT}}}{f_s} \quad (60)$$

hold as well. Thus, a DFT and an interpolated DTFT spectrum can be visualised over μ , Ω and, for a given sampling frequency f_s , over f as depicted in Fig. 9.

1.7 Windowing

Many textbooks treat the derivation of the DFT and windowing separately, which is sensible if the concept of the DFT is to be introduced solely as a transform. For the spectral analysis of signals, it might be helpful to discuss the DFT and windowing together as e.g. in [Mös11], [Kam02, Ch. 7.3, 7.4].

Let $x[k]$ for $-\infty \leq k \leq \infty$ be a sequence with the corresponding continuous and periodic DTFT spectrum $X(\Omega)$. For a computer-assisted spectral analysis, it is necessary to cut out a finite number of values out of a representative signal section. Mathematically, this is described by multiplication with a weighting or window sequence $w[k]$, that equals zero outside the desired signal section:

$$w[k] = 0 \quad \text{for } k < 0 \text{ and } k \geq N, \quad w \in \mathbb{R}. \quad (61)$$

For example, the definition of a rectangular window is

$$w[k] = \begin{cases} 1 & \text{for } 0 \leq k \leq N-1 \\ 0 & \text{otherwise} \end{cases}. \quad (62)$$

Of course the chosen signal section does not have to start at $k = 0$, but it makes calculations easier. The sequence $w[k]$ has a corresponding DTFT spectrum $W(\Omega)$ as well:

$$\begin{aligned} W(\Omega) &= \sum_{k=-\infty}^{\infty} w[k] e^{-j\Omega k} \\ w[k] &= \frac{1}{2\pi} \int_{-\pi}^{\pi} W(\Omega) e^{j\Omega k} d\Omega. \end{aligned} \quad (63)$$

Cutting-out the signal section means multiplication of the signal with the window sequence

$$x_N[k] = x[k] \cdot w[k], \quad (64)$$

which can be expressed as a cyclic convolution of the corresponding spectra in the frequency domain:

$$X_N(\Omega) = \frac{1}{2\pi} (X \circledast W)(\Omega) \quad (65)$$

$$= \frac{1}{2\pi} \int_{-\pi}^{\pi} X(\Omega - \nu) \cdot W(\nu) d\nu \quad (66)$$

$$= \frac{1}{2\pi} \int_{-\pi}^{\pi} X(\nu) \cdot W(\Omega - \nu) d\nu. \quad (67)$$

The DTFT spectrum $X_N(\Omega)$ contains the "original" DTFT spectrum $X(\Omega)$ that was to be analysed, but it is smeared due to the convolution with the window spectrum. The only possibility to obtain $X_N(\Omega) = X(\Omega)$ is to convolve with a window that is the neutral element of the convolution

$$W_\delta(\Omega) = 2\pi \sum_{n=-\infty}^{\infty} \delta(\Omega + 2\pi n) \quad (68)$$

which yields

$$X_N(\Omega) = \int_{-\pi}^{\pi} X(\Omega - \nu) \cdot \delta(\nu) d\nu = \int_{-\pi}^{\pi} X(\nu) \cdot \delta(\Omega - \nu) d\nu = X(\Omega). \quad (69)$$

However, $W_\delta(\Omega)$ corresponds to [Opp10, tab. 2.3, p. 90]

$$W_\delta(\Omega) = 2\pi \sum_{n=-\infty}^{\infty} \delta(\Omega + 2\pi n) \quad \bullet \text{---} \circ \quad w_\delta[k] = 1 \quad \text{for } -\infty \leq k \leq \infty, \quad (70)$$

which constitutes an infinitely long window. For the time domain follows

$$x_N[k] = x[k] \cdot w_\delta[k] = x[k], \quad (71)$$

which delivers a consistent theory, but is not helpful for practical computation.

Therefore, $w_\delta[k]$ is not a window that can be employed in practice (and it is not actually not even a real "window" as it is constantly 1), but it should be kept in mind that $W_\delta(\Omega)$ would be the ideal window spectrum for the analysis of $x[k]$. So one has to accept that instead of being able to analyse $x[k]$ and $X(\Omega)$, only $x_N[k]$ and $X_N(\Omega)$ are available. How well $X_N(\Omega) \approx X(\Omega)$ can be approximated is dependent on the length N of the cut out section, on the chosen section of $x[k]$ and on the sequence $w[k]$ and its spectrum $W(\Omega)$, that are itself dependent on N .

To illustrate the importance of the window spectrum, the signal spectrum of a single harmonic oscillation with the angular frequency $0 \leq \Omega_0 < \pi$ is considered [Opp10, tab. 2.3, p. 90]

$$X_\delta(\Omega) = 2\pi \sum_{n=-\infty}^{\infty} \delta(\Omega - \Omega_0 + 2\pi n). \quad (72)$$

For $X_N(\Omega)$ follows from the convolution $X_{\delta,N}(\Omega) = \frac{1}{2\pi} (X_\delta \otimes W)(\Omega)$ the spectrum of the window sequence shifted by Ω_0

$$X_{\delta,N}(\Omega) = W(\Omega - \Omega_0). \quad (73)$$

Therefore, one must be able to interpret the window spectrum to conclude on $X_N(\Omega)$ or even $X(\Omega)$. The next section starts discussion on the rectangular window which is the 'natural' window of the DFT definition as it just cuts out a section of the signal where all values are weighted equally.

1.7.1 Rectangular Window

A window $w[k]$ of length N is defined with

$$w[k] = 0 \quad \text{for } k < 0 \text{ and } k \geq N, \quad w \in \mathbb{R}. \quad (74)$$

For $0 \leq k \leq N-1$, $w[k]$ can consist of arbitrary real numbers. If $w[k] = 1$ for this range, a so called rectangular window is obtained. To determine the spectrum of a window, one can calculate the DTFT of the infinite sequence $w[k]$, where only values of the sequence that are $\neq 0$ have to be summed up:

$$W(\Omega) = \sum_{k=0}^{N-1} w[k] \cdot e^{-j\Omega k}. \quad (75)$$

For the simple rectangular window this yields

$$W_{\text{Rect}}(\Omega) = \sum_{k=0}^{N-1} 1 \cdot e^{-j\Omega k}. \quad (76)$$

This is a geometric series for which the closed form solution is known [Lyo11, (3-39)] as

$$W_{\text{Rect}}(\Omega) = \frac{1 - e^{-j\Omega N}}{1 - e^{-j\Omega}}. \quad (77)$$

This expression can be rewritten as

$$W_{\text{Rect}}(\Omega) = \frac{e^{-j\frac{\Omega N}{2}}}{e^{-j\frac{\Omega}{2}}} \cdot \frac{e^{j\frac{\Omega N}{2}} - e^{-j\frac{\Omega N}{2}}}{e^{j\frac{\Omega}{2}} - e^{-j\frac{\Omega}{2}}} = e^{-j\frac{\Omega(N-1)}{2}} \cdot \frac{e^{j\frac{\Omega N}{2}} - e^{-j\frac{\Omega N}{2}}}{e^{j\frac{\Omega}{2}} - e^{-j\frac{\Omega}{2}}} \quad (78)$$

and with the Euler identity $2j \cdot \sin(x) = e^{jx} - e^{-jx}$ be simplified to (cf. [Lyo11, (3-42)])

$$W_{\text{Rect}}(\Omega) = e^{-j\frac{\Omega(N-1)}{2}} \cdot \frac{\sin\left(N\frac{\Omega}{2}\right)}{\sin\left(\frac{\Omega}{2}\right)}. \quad (79)$$

The spectrum $W_{\text{Rect}}(\Omega)$ is like all Fourier transformed sequences 2π -periodic. For the following discussion, only the section of the base band $-\pi \leq \Omega < \pi$ is considered.

For the magnitude spectrum only the fraction containing the sines has to be evaluated because $\left|e^{-j\frac{\Omega(N-1)}{2}}\right| = 1$. With the rule of L'Hospital [Olv10, (1.4.15)] the rectangular window at $\Omega = 0$ is

$$W_{\text{Rect}}(\Omega = 0) = N. \quad (80)$$

This is the amplitude that is weighting the current mid frequency in the convolution the strongest, the so-called maximum main lobe amplitude. The zeros $W_{\text{Rect}}(\Omega) = 0$ in the base band result with $m \in \mathbb{Z}$, $m \neq 0$ after considering the argument of the sine in the numerator

$$N\frac{\Omega}{2} = m\pi, \quad (81)$$

in

$$\Omega = \frac{2\pi}{N}m = \Delta\Omega \cdot m. \quad (82)$$

The zeros are equidistantly spaced with $\Delta\Omega = \frac{2\pi}{N}$. For odd N , $N-1$ zeros result for

$$m = \pm 1, \pm 2, \pm 3, \dots \pm \frac{N-1}{2}. \quad (83)$$

For even N , $N-1$ zeros result for

$$m = \pm 1, \pm 2, \pm 3, \dots \pm \left(\frac{N}{2} - 1\right), -\frac{N}{2}. \quad (84)$$

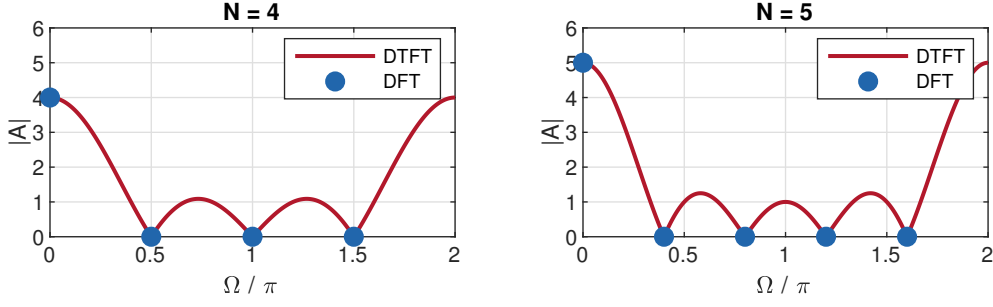


Figure 10: Magnitude of DTFT & DFT spectra for rectangular windows of length N , $0 \leq \Omega < 2\pi$.

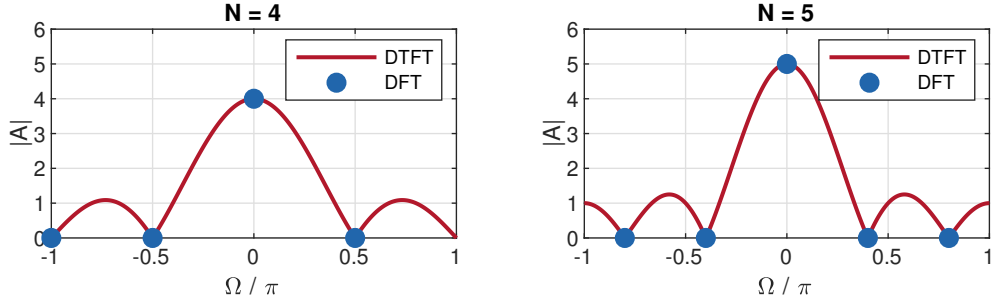


Figure 11: Magnitude of DTFT & DFT spectra for rectangular windows of length N , $-\pi \leq \Omega < \pi$.

In the last case, the last zero $m = -\frac{N}{2}$ is exactly at $\Omega = -\pi$. As it is a 2π -periodic spectrum this zero could be defined with $m = \frac{N}{2}$ as $\Omega = \pi$, but this would be the same zero and it is not counted twice. For $m = 0$, no zero can be found, but instead the main lobe, as has been described above.

Fig. 10 and 11 illustrate the cases $N = 4$ and $N = 5$ for the DTFT of the rectangular window. The $N - 1$ zeros of the spectrum are clearly visible. Fig. 12 and 13 show the spectra with a logarithmic amplitude. Differing from the depiction here, textbooks mostly use normalised representations of window spectra of the form $20 \cdot \log_{10} W[\Omega = 0] = 0$ dB, that are symmetric in $\Omega = 0$, i.e. plotted over $-\pi \leq \Omega < \pi$.

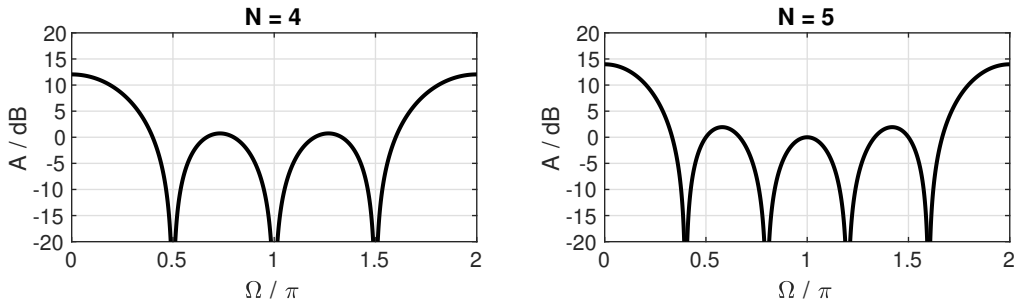


Figure 12: Level of DTFT spectra for rectangular windows of length N , $0 \leq \Omega < 2\pi$.

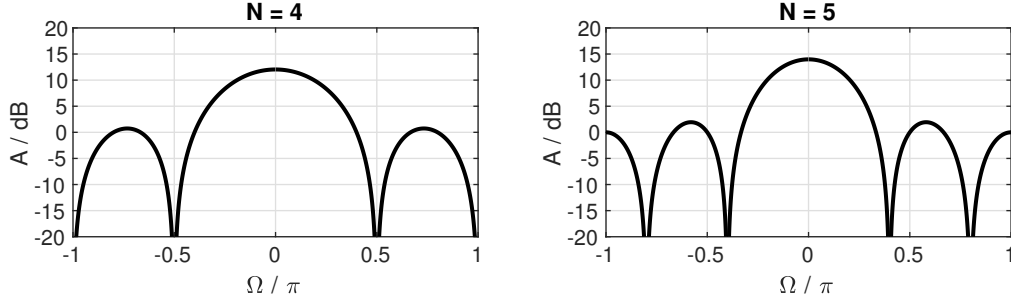


Figure 13: Level of DTFT spectra for rectangular windows of length N , $-\pi \leq \Omega < \pi$.

1.7.2 Discrete Spectrum of the Rectangular Window

As has been explained before, the spectrum from eq. (72)

$$X_\delta(\Omega) = 2\pi \sum_{n=-\infty}^{\infty} \delta(\Omega - \Omega_0 + 2\pi n) \quad (85)$$

evolves with windowing, i.e. the convolution $X_{\delta,N}(\Omega) = \frac{1}{2\pi} (X_\delta \circledast W)(\Omega)$, into

$$X_{\delta,N}(\Omega) = W(\Omega - \Omega_0). \quad (86)$$

Now, the spectrum of a finite signal is considered, i.e. a transition from the DTFT to the DFT is performed. For the discrete DFT frequencies that contain the complete spectral information, $X_\delta[\mu]$ can be written as

$$X_\delta[\mu] = W \left[\frac{2\pi}{N} \mu - \Omega_0 \right]. \quad (87)$$

With eq. (79) follows

$$X_\delta[\mu] = W_{\text{Rect}} \left[\frac{2\pi}{N} \mu - \Omega_0 \right] = e^{-j \frac{(2\pi \mu - \Omega_0)(N-1)}{2}} \cdot \frac{\sin \left(N \frac{2\pi \mu - \Omega_0}{2} \right)}{\sin \left(\frac{2\pi \mu - \Omega_0}{2} \right)} \quad (88)$$

$$= e^{-j \left(\frac{N-1}{2} \left(\frac{2\pi}{N} \mu - \Omega_0 \right) \right)} \cdot \frac{\sin \left(\frac{N}{2} \left(\frac{2\pi}{N} \mu - \Omega_0 \right) \right)}{\sin \left(\frac{1}{2} \left(\frac{2\pi}{N} \mu - \Omega_0 \right) \right)}. \quad (89)$$

As can be seen, the DFT coefficients $X_\delta[\mu]$ strongly depend on the choice of Ω_0 , which will be illustrated in the following. The definition

$$\Omega_0 = \frac{2\pi}{N}(\mu_0 + \alpha) \quad \text{with} \quad \pm \alpha \leq \frac{1}{2} \quad (90)$$

is inserted into eq. (89) to yield

$$X_\delta[\mu] = e^{-j \left(\frac{N-1}{2} \left(\frac{2\pi}{N} \mu - \frac{2\pi}{N}(\mu_0 + \alpha) \right) \right)} \cdot \frac{\sin \left(\frac{N}{2} \left(\frac{2\pi}{N} \mu - \frac{2\pi}{N}(\mu_0 + \alpha) \right) \right)}{\sin \left(\frac{1}{2} \left(\frac{2\pi}{N} \mu - \frac{2\pi}{N}(\mu_0 + \alpha) \right) \right)} \quad (91)$$

$$= e^{-j \left(\frac{N-1}{2} \left(\frac{2\pi}{N}(\mu - \mu_0) - \frac{2\pi}{N} \alpha \right) \right)} \cdot \frac{\sin \left(\pi((\mu - \mu_0) - \alpha) \right)}{\sin \left(\frac{\pi}{N}((\mu - \mu_0) - \alpha) \right)}. \quad (92)$$

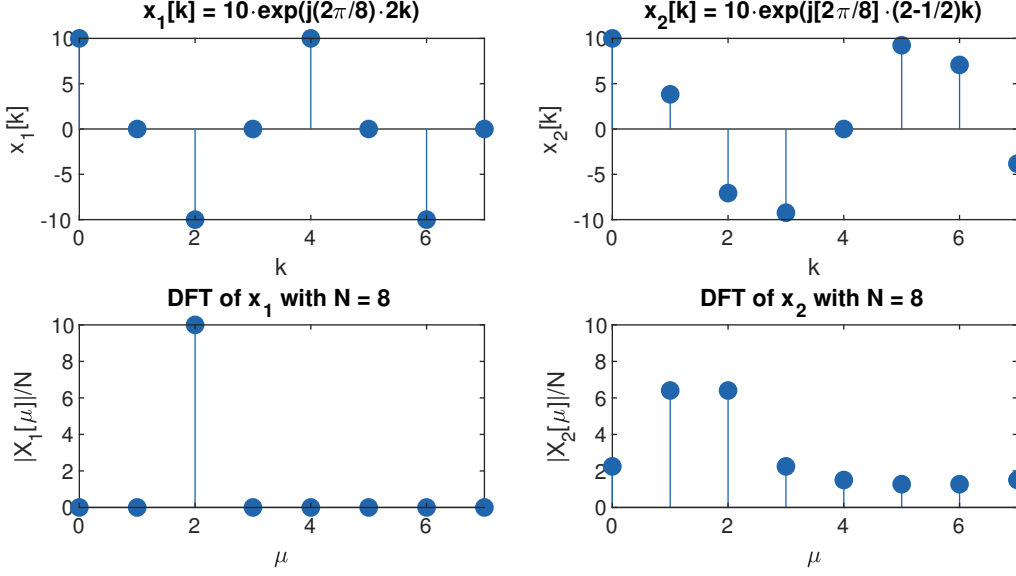


Figure 14: Rectangular windowing for $N = 8$, best case ($\mu = 2$), worst case ($\mu = 2 - \frac{1}{2}$).

For the case $\alpha = 0$, we have $\Omega_0 = \frac{2\pi}{N}\mu_0$ and the fraction containing the sines evaluates to N (cf. periodic sinc in eq. (54)), so the DFT spectrum becomes

$$X_\delta[\mu] = \begin{cases} 0 & \text{for } \mu \neq \mu_0 \\ N & \text{for } \mu = \mu_0 \end{cases}. \quad (93)$$

Therefore, if the chosen frequency from eq. (72) happens to be a DFT bin, the frequency spectrum contains the exact amplitude of this frequency (after normalising with $\frac{1}{N}$). The other values of the spectrum coincide with the zeros of the window spectrum. This is the best case because the amplitude of the frequency Ω_0 can be analysed exactly, cf. Fig. 14 left. The worst case occurs when $\alpha = \pm\frac{1}{2}$. For the rectangular window, two bins share the main energy and the other bins differ from zero as well, cf. Fig. 14 right. It is not as easy anymore how to interpret the DFT spectrum. From the example it is known that the signal consists of only one spectral component, but in the spectrum this is not so obvious. So what is to do when the frequencies and corresponding amplitudes of the signal are not known (as is of course usually the case in real applications)?

For the interpretation of such spectra, knowledge about windowing and different window types as well as experience in analysing spectra is necessary. In a first step one can determine for each window which amplitude error arises between best and worst case. This gives an indication for the "true" spectrum. For the rectangular window, this calculation is quite easy and is shown here: With eq. (92), the amplitude for a certain μ_0 is generally

$$|X_\delta[\mu_0]| = \left| \frac{\sin(-\pi\alpha)}{\sin(-\frac{\pi}{N}\alpha)} \right| = \left| \frac{\sin(\pi\alpha)}{\sin(\frac{\pi}{N}\alpha)} \right| \quad (94)$$

and for the worst case $\alpha = \pm\frac{1}{2}$

$$|X_\delta[\mu_0]| = \left| \frac{\sin(\pm\frac{\pi}{2})}{\sin(\pm\frac{\pi}{2N})} \right| = \left| \frac{\pm 1}{\pm \sin(\frac{\pi}{2N})} \right| = \left| \frac{1}{\sin(\frac{\pi}{2N})} \right|. \quad (95)$$

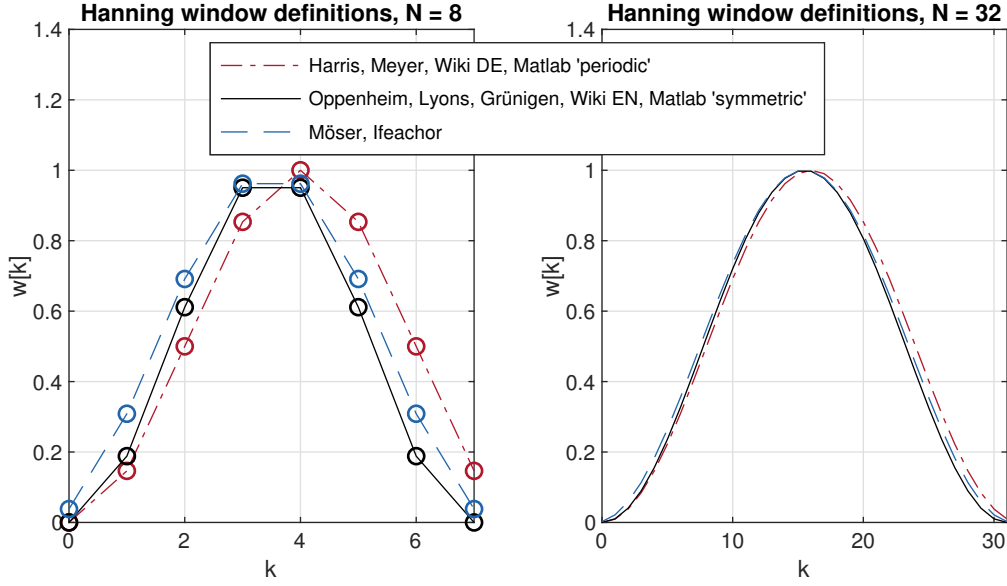


Figure 15: Different definitions of the Hanning window.

For very large N , this yields due to $\sin(x) \approx x$ for small x the amplitude

$$|X_\delta[\mu_0]| = \frac{2N}{\pi}. \quad (96)$$

The amplitude error between the best case for $\alpha = 0$ and the worst case for $\alpha = \pm \frac{1}{2}$ is then for large N

$$\frac{|X_\delta[\mu_0]_{\alpha=\pm \frac{1}{2}}|}{|X_\delta[\mu_0]_{\alpha=0}|} \approx \frac{\frac{2N}{\pi}}{\frac{2N}{\pi}} \approx \frac{2}{\pi} \approx 0.6366 \triangleq -3.9224 \text{ dB}. \quad (97)$$

The amplitude loss from the value 10 to approx. 6.3 can be found in Fig. 14. It is still not known, though, which frequencies the signal consists of ($\mu = 1$ or $\mu = 2$ or both?). This could be accomplished by an increase of N . Additionally, the quite high amplitudes in the other bins are confusing. This point can be counteracted with a window with a higher side lobe attenuation that is presented in the next subsection discussing the Hanning window.

1.7.3 Hanning Window

For the Hanning window (and for other windows as well), different definitions exist in the literature. Fig. 15 shows differently defined Hanning windows for two lengths N . The red line starts at $w[k=0] = 0$, but ends with $w[k=N-1] \neq 0$. This is based upon the reason that (at least for spectral analysis) a symmetric window is not necessary [Har78, p. 52]. Instead, the window is constructed in a such a way that for periodic extension $w[N] = w[0]$. The black line defines a window that is symmetric around $\frac{N-1}{2}$ and at both terminal points $w[k=0] = 0$ and $w[k=N-1] = 0$ holds, because of the common reasoning that the windowed signal must be zero at the terminal points to ensure continuous transitions for periodic repetitions [Opp10, Lyo11]. The blue curve defines a window that is symmetric around $\frac{N-1}{2}$ as well, but both terminal points $w[k=0] \neq 0$ and $w[k=N-1] \neq 0$. The reason for this is that N ‘real’ DFT bins should be retrieved [Mös11, Ife02]. This is why the sequence $w[k]$ for the blue line

consists of N coefficients that differ from 0 while it is only $N - 2$ for the black line. A DFT of length N for a sequence with only $N - 2$ coefficients differing from 0 yields only $N - 2$ ‘real’ bins (compare to zero-padding). For scientific purposes, it should be stated which definition of a standard window has been used.

The application can determine which definition of a window is appropriate. For spectral analysis a window should be used that can be extended periodically without losing information in bins. Those windows do not have to be symmetric around the middle of the window. In FIR filter design one aims at reducing an infinite impulse response to a finite length. This is mostly done by symmetric windows to window symmetrically around the main peak of the impulse response. Therefore, Matlab discriminates between the variants **periodic** and **symmetric** in window design.

Here, the window definition of [Mös11] is used, although the Matlab window as **periodic** version would be better suited, but the calculation is easier. For the window $w[k]$ of length N ,

$$w[k] = 0 \quad \text{for } k < 0 \text{ and } k \geq N, \quad w \in \mathbb{R} \quad (98)$$

is still valid. For $0 \leq k \leq N - 1$, $w[k]$ shall consist of arbitrary real numbers. In this case here, those numbers are the ones of a Hanning window. Additionally, the window should be symmetric

$$w[N - 1 - k] = w[k] \quad (99)$$

and without zeros in the definition range. The Hanning window version that fulfills this reads

$$w[k] = \begin{cases} 1 - \cos\left(\frac{2\pi}{N}\left(k + \frac{1}{2}\right)\right) & \text{for } 0 \leq k \leq N - 1 \\ 0 & \text{else} \end{cases}. \quad (100)$$

The spectrum can be calculated with the help of

$$\cos\left(\frac{2\pi}{N}\left(k + \frac{1}{2}\right)\right) = \frac{1}{2} \left(e^{j\frac{\pi}{N}} e^{j\frac{2\pi}{N}k} + e^{-j\frac{\pi}{N}} e^{-j\frac{2\pi}{N}k} \right) \quad (101)$$

and

$$W(\Omega) = \sum_{k=0}^{N-1} w[k] e^{-j\Omega k}. \quad (102)$$

Therefore, the spectrum of the window is

$$W_{\text{Hann}}(\Omega) = \sum_{k=0}^{N-1} \left(1 - \cos\left(\frac{2\pi}{N}\left(k + \frac{1}{2}\right)\right) \right) e^{-j\Omega k} \quad (103)$$

$$= \sum_{k=0}^{N-1} \left(1 - \frac{1}{2} \left(e^{j\frac{\pi}{N}} e^{j\frac{2\pi}{N}k} + e^{-j\frac{\pi}{N}} e^{-j\frac{2\pi}{N}k} \right) \right) e^{-j\Omega k} \quad (104)$$

$$= \sum_{k=0}^{N-1} e^{-j\Omega k} - \frac{1}{2} e^{j\frac{\pi}{N}} \sum_{k=0}^{N-1} e^{-j(\Omega - \frac{2\pi}{N})k} - \frac{1}{2} e^{-j\frac{\pi}{N}} \sum_{k=0}^{N-1} e^{-j(\Omega + \frac{2\pi}{N})k}. \quad (105)$$

Using the results for the rectangular windows (cf. eq. (76)), this can be abbreviated to

$$W_{\text{Hann}}(\Omega) = W_{\text{Rect}}(\Omega) - \frac{1}{2} e^{j\frac{\pi}{N}} W_{\text{Rect}}(\Omega - \frac{2\pi}{N}) - \frac{1}{2} e^{-j\frac{\pi}{N}} W_{\text{Rect}}(\Omega + \frac{2\pi}{N}). \quad (106)$$

As can be seen, the spectrum of the Hanning window is composed of the spectrum of a rectangular window and two weighted spectra of rectangular windows that are shifted by one bin. With eq. (79), this can be rewritten as

$$\begin{aligned}
W_{\text{Hann}}(\Omega) &= e^{-j\frac{\Omega(N-1)}{2}} \cdot \frac{\sin\left(N\frac{\Omega}{2}\right)}{\sin\left(\frac{\Omega}{2}\right)} \\
&\quad - \frac{1}{2} e^{j\frac{\pi}{N}} e^{-j\frac{(\Omega-\frac{2\pi}{N})(N-1)}{2}} \cdot \frac{\sin\left(N\frac{\Omega-\frac{2\pi}{N}}{2}\right)}{\sin\left(\frac{\Omega-\frac{2\pi}{N}}{2}\right)} \\
&\quad - \frac{1}{2} e^{-j\frac{\pi}{N}} e^{-j\frac{(\Omega+\frac{2\pi}{N})(N-1)}{2}} \cdot \frac{\sin\left(N\frac{\Omega+\frac{2\pi}{N}}{2}\right)}{\sin\left(\frac{\Omega+\frac{2\pi}{N}}{2}\right)} \tag{107}
\end{aligned}$$

$$\begin{aligned}
&= e^{-j\frac{\Omega(N-1)}{2}} \cdot \frac{\sin\left(\frac{N}{2}\Omega\right)}{\sin\left(\frac{1}{2}\Omega\right)} \\
&\quad - \frac{1}{2} e^{j\frac{\pi}{N}} e^{-j\frac{(\Omega-\frac{2\pi}{N})(N-1)}{2}} \cdot \frac{\sin\left(\frac{N}{2}\left(\Omega-\frac{2\pi}{N}\right)\right)}{\sin\left(\frac{1}{2}\left(\Omega-\frac{2\pi}{N}\right)\right)} \\
&\quad - \frac{1}{2} e^{-j\frac{\pi}{N}} e^{-j\frac{(\Omega+\frac{2\pi}{N})(N-1)}{2}} \cdot \frac{\sin\left(\frac{N}{2}\left(\Omega+\frac{2\pi}{N}\right)\right)}{\sin\left(\frac{1}{2}\left(\Omega+\frac{2\pi}{N}\right)\right)}. \tag{108}
\end{aligned}$$

Because

$$e^{j\frac{\pi}{N}} e^{-j\frac{(\Omega-\frac{2\pi}{N})(N-1)}{2}} = e^{-j\frac{\Omega(N-1)}{2}} e^{j\frac{\pi}{N}} e^{j\frac{(N-1)}{2}\frac{2\pi}{N}} = e^{-j\frac{\Omega(N-1)}{2}} \underbrace{e^{j\pi\left(\frac{1}{N}+\frac{N-1}{N}\right)}}_{=-1} \tag{109}$$

and

$$e^{-j\frac{\pi}{N}} e^{-j\frac{(\Omega+\frac{2\pi}{N})(N-1)}{2}} = e^{-j\frac{\Omega(N-1)}{2}} e^{-j\frac{\pi}{N}} e^{-j\frac{(N-1)}{2}\frac{2\pi}{N}} = e^{-j\frac{\Omega(N-1)}{2}} \underbrace{e^{-j\pi\left(\frac{1}{N}+\frac{N-1}{N}\right)}}_{=-1}, \tag{110}$$

the spectrum can be simplified to

$$W_{\text{Hann}}(\Omega) = e^{-j\frac{\Omega(N-1)}{2}} \cdot \frac{\sin\left(\frac{N}{2}\Omega\right)}{\sin\left(\frac{1}{2}\Omega\right)} + \frac{1}{2} e^{-j\frac{\Omega(N-1)}{2}} \cdot \frac{\sin\left(\frac{N}{2}\left(\Omega-\frac{2\pi}{N}\right)\right)}{\sin\left(\frac{1}{2}\left(\Omega-\frac{2\pi}{N}\right)\right)} + \frac{1}{2} e^{-j\frac{\Omega(N-1)}{2}} \cdot \frac{\sin\left(\frac{N}{2}\left(\Omega+\frac{2\pi}{N}\right)\right)}{\sin\left(\frac{1}{2}\left(\Omega+\frac{2\pi}{N}\right)\right)} \tag{111}$$

$$= e^{-j\frac{\Omega(N-1)}{2}} \cdot \left(\frac{\sin\left(\frac{N}{2}\Omega\right)}{\sin\left(\frac{1}{2}\Omega\right)} + \frac{1}{2} \cdot \frac{\sin\left(\frac{N}{2}\left(\Omega-\frac{2\pi}{N}\right)\right)}{\sin\left(\frac{1}{2}\left(\Omega-\frac{2\pi}{N}\right)\right)} + \frac{1}{2} \cdot \frac{\sin\left(\frac{N}{2}\left(\Omega+\frac{2\pi}{N}\right)\right)}{\sin\left(\frac{1}{2}\left(\Omega+\frac{2\pi}{N}\right)\right)} \right). \tag{112}$$

The Hanning window exhibits only $N - 3$ zeros compared to the rectangular window with $N - 1$ zeros, so two zeros are lost for usage in spectral shaping, cf. fig 16 and 17. In the region close to $\Omega = \pm\pi$, the weighted and shifted spectra of the rectangular windows (red and dark blue line) cancel out, which can be observed in the logarithmic depiction in Fig. 18 as well, i.e. the Hanning window achieves a better side lobe attenuation than the rectangular window. Around $\Omega \approx 0$, all three spectral components add up in a constructive way, which leads to a broader main lobe compared to the rectangular window.

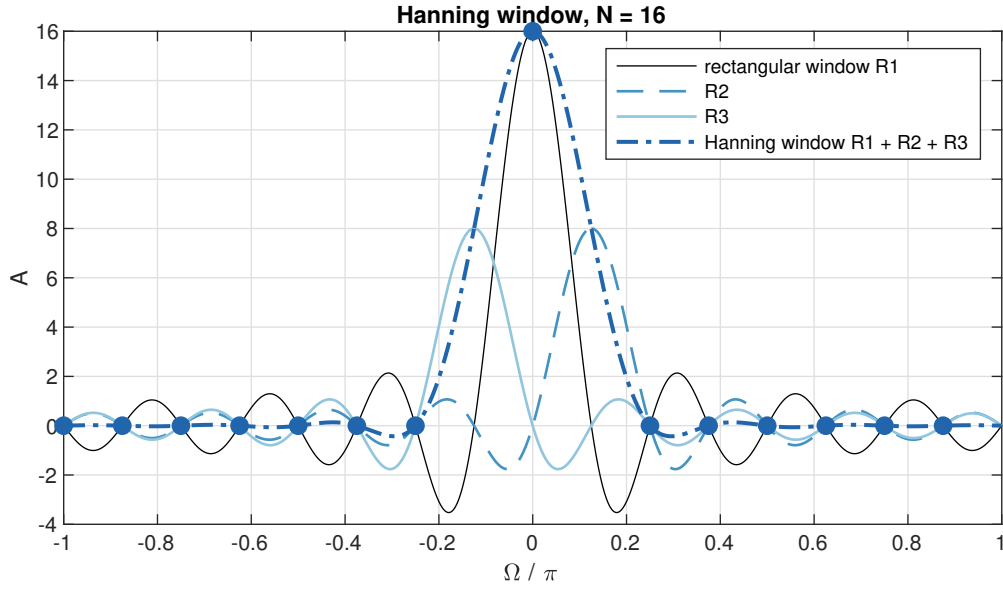


Figure 16: Spectrum of Hanning window from superposition of weighted rectangular windows.

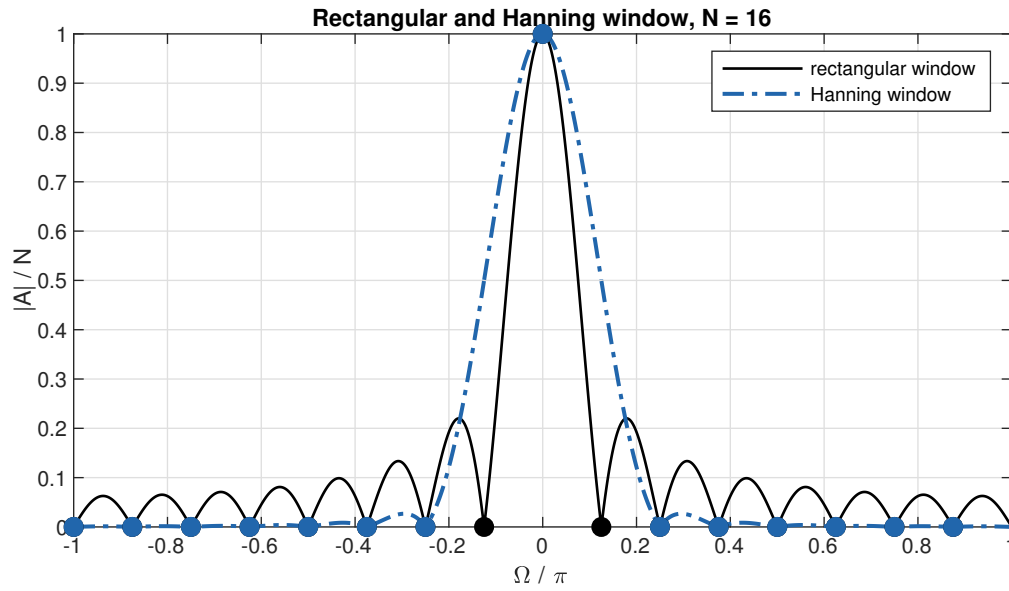


Figure 17: Magnitude of DTFT spectra for rectangular and Hanning window, $-\pi \leq \Omega < \pi$.

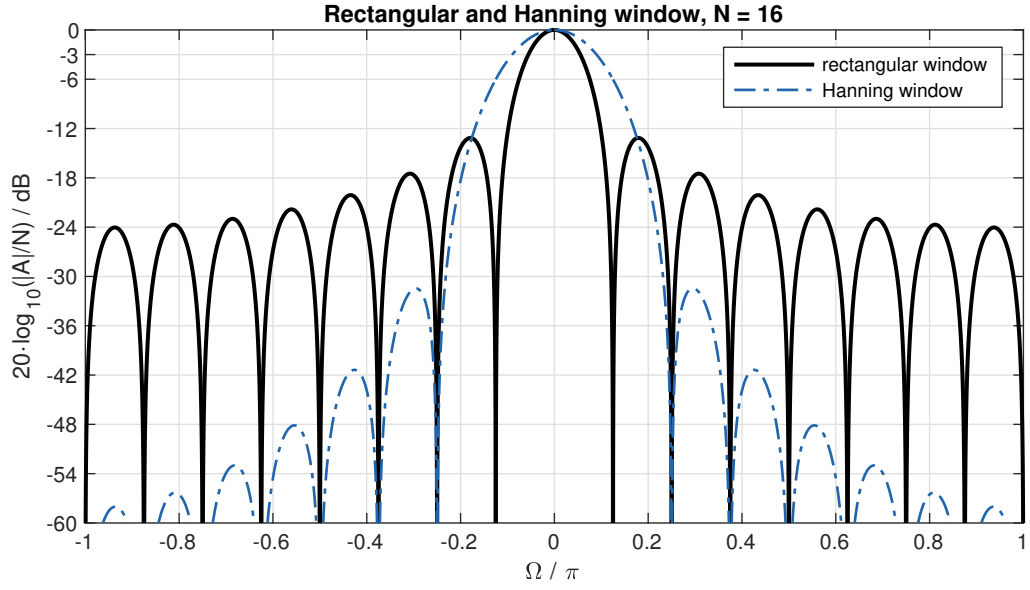


Figure 18: Level of DTFT spectra for rectangular and Hanning window, $-\pi \leq \Omega < \pi$.

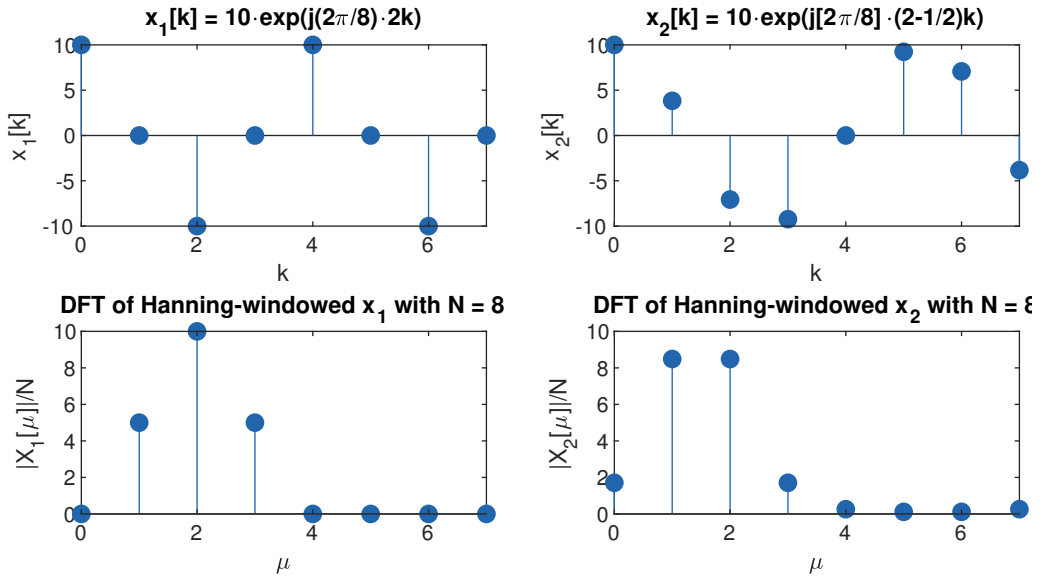


Figure 19: Hanning windowing for $N = 8$, best case ($\mu = 2$), worst case ($\mu = 2 - \frac{1}{2}$).

Fig. 19 illustrates the best and worst case scenarios for the Hanning window when a discrete frequency exactly coincides with a DFT bin (left, best) or lies in the middle between two bins (right, worst). In contrast to the rectangular window, the exact case ($\mu = 2$, left) leads to three bins that are differing from zero with the middle bin representing the exact amplitude of the discrete frequency. The worst case ($\mu = 2 - \frac{1}{2}$, right) leads to two main bins and further bins differing from zero. The amplitude error between best and worst case for large N is

$$\frac{|X[\mu_0]_{\alpha=\pm\frac{1}{2}}|}{|X[\mu_0]_{\alpha=0}|} = \frac{3\pi}{8} \approx 1.1781 \hat{=} -1.4236 \text{ dB}, \quad (113)$$

which is smaller than the error for the rectangular window. This discloses the following rule: The broader the main lobe of a window, the smaller the amplitude error in spectral analysis for a harmonic signal with the frequency

$$\Omega_0 = \frac{2\pi}{N}(\mu_0 + \alpha) \quad \text{with} \quad \pm \alpha \leq \frac{1}{2}. \quad (114)$$

However, a broad main lobe leads to a poor frequency resolution, i.e. frequencies that are close together get smeared in the convolution process $X_N(\Omega) = \frac{1}{2\pi}(X_N \circledast W)(\Omega)$ and cannot be separated clearly anymore. So, there is always a compromise involved in frequency resolution and amplitude resolution.

One last remark on the amplitudes of the windows: In the literature, windows are often normalised to a maximum value = 1, which reads for the Hanning window according to [Mös11]

$$w[k] = \begin{cases} 0.5 - 0.5 \cos\left(\frac{2\pi}{N}\left(k + \frac{1}{2}\right)\right) & \text{for } 0 \leq k \leq N-1 \\ 0 & \text{else} \end{cases}. \quad (115)$$

The definition that has been used above (cf. eq. (100)),

$$w[k] = \begin{cases} 1 - \cos\left(\frac{2\pi}{N}\left(k + \frac{1}{2}\right)\right) & \text{for } 0 \leq k \leq N-1 \\ 0 & \text{else} \end{cases}. \quad (116)$$

was chosen to result in equal amplitudes N for the rectangular and the Hanning window. In the famous and classic paper [Har78, tab. 1] this normalisation factor is treated as the so-called ‘coherent gain’.

1.7.4 Hamming Window

The Hanning window with only $N - 3$ zeros in its spectrum lets two potential zeros unused. Note that using all possible N zeros would not allow to create a distinct main lobe at $\Omega = 0$, therefore only $N - 1$ zeros are used for optimal window design. The Hamming window adds these zeros again to the Hanning window spectrum in order to improve the attenuation of the first side lobes (left and right of the main lobe). To that end, the Hanning window spectrum is modified with a factor β (cf. [Rab75, Ch. 3.10]):

$$W_{\text{Hamm}}(\Omega) = e^{-j\frac{\Omega(N-1)}{2}} \cdot \left(\frac{\sin\left(\frac{N}{2}\Omega\right)}{\sin\left(\frac{1}{2}\Omega\right)} + \frac{\beta}{2} \cdot \frac{\sin\left(\frac{N}{2}\left(\Omega - \frac{2\pi}{N}\right)\right)}{\sin\left(\frac{1}{2}\left(\Omega - \frac{2\pi}{N}\right)\right)} + \frac{\beta}{2} \cdot \frac{\sin\left(\frac{N}{2}\left(\Omega + \frac{2\pi}{N}\right)\right)}{\sin\left(\frac{1}{2}\left(\Omega + \frac{2\pi}{N}\right)\right)} \right). \quad (117)$$

At $\Omega = \frac{2\pi}{N} \cdot 2.5$, i.e. at $\mu = 2.5$ in the DFT spectrum, an additional zero shall be inserted at the maximum of the first side lobe. As $w[k]$ shall be real, this zero must have a symmetric

counterpart (so this yields the desired two additional zeros), but only one has to be specified. For small arguments, the approximation $\sin(x) \approx x$ holds, which is fulfilled in the denominators for large N when $\Omega = \frac{2\pi}{N} \cdot 2.5$, and for $\Omega = 5\pi/N$ the result is:

$$\left| W_{\text{Hamm}}(\Omega = \frac{5\pi}{N}) \right| \approx \left| \frac{\sin(\frac{N}{2} \frac{5\pi}{N})}{\frac{1}{2} \frac{5\pi}{N}} + \frac{\beta}{2} \cdot \frac{\sin(\frac{N}{2} (\frac{5\pi}{N} - \frac{2\pi}{N}))}{\frac{1}{2} (\frac{5\pi}{N} - \frac{2\pi}{N})} + \frac{\beta}{2} \cdot \frac{\sin(\frac{N}{2} (\frac{5\pi}{N} + \frac{2\pi}{N}))}{\frac{1}{2} (\frac{5\pi}{N} + \frac{2\pi}{N})} \right|. \quad (118)$$

Searching for the zero

$$\left| W_{\text{Hamm}}(\Omega = \frac{5\pi}{N}) \right| = \left| \frac{1}{\frac{5\pi}{2N}} + \frac{\beta}{2} \cdot \frac{-1}{\frac{3\pi}{2N}} + \frac{\beta}{2} \cdot \frac{-1}{\frac{7\pi}{2N}} \right| \stackrel{!}{=} 0 \quad (119)$$

and solving for the corresponding β yields

$$\frac{\beta}{2} \cdot \left(\frac{2N}{3\pi} + \frac{2N}{7\pi} \right) = \frac{2N}{5\pi} \quad (120)$$

$$\Leftrightarrow \beta = \frac{21}{25} = 0.84. \quad (121)$$

The Hamming window can be rearranged to

$$W_{\text{Hamm}}(\Omega) = (1 - \beta) W_{\text{Rect}}(\Omega) + \beta W_{\text{Hann}}(\Omega), \quad (122)$$

which means for the sequence

$$w_{\text{Hamm}}[k] = \begin{cases} (1 - \beta)w_{\text{Rect}}[k] + \beta w_{\text{Hann}}[k] & \text{for } 0 \leq k \leq N - 1 \\ 0 & \text{else} \end{cases} \quad (123)$$

with

$$w_{\text{Rect}}[k] = \begin{cases} 1 & \text{for } 0 \leq k \leq N - 1 \\ 0 & \text{else} \end{cases} \quad (124)$$

and

$$w_{\text{Hann}}[k] = \begin{cases} 1 - \cos\left(\frac{2\pi}{N}\left(k + \frac{1}{2}\right)\right) & \text{for } 0 \leq k \leq N - 1 \\ 0 & \text{else} \end{cases}. \quad (125)$$

Finally, the sequence of the Hamming window can be rewritten to

$$w_{\text{Hamm}}[k] = \begin{cases} 1 - 0.84 \cos\left(\frac{2\pi}{N}\left(k + \frac{1}{2}\right)\right) & \text{for } 0 \leq k \leq N - 1 \\ 0 & \text{else} \end{cases}. \quad (126)$$

The derivation from [Har78, p. 62] concludes that the Hamming window as it is defined in the standard literature must have been developed by rounding, which means that the zero is at $\Omega \approx \frac{2\pi}{N} \cdot 2.6$. If this zero is used in the calculation here according to [Mös11], $\beta \approx 0.852$ results. If now the windows are scaled to have a maximum of 1 for odd N , the common representation of the Hamming window in the literature results (here written in the form of [Mös11]):

$$w_{\text{Hamm}}[k] \approx 0.54 - 0.54 \beta \cdot \cos\left(\frac{2\pi}{N}\left(k + \frac{1}{2}\right)\right) \quad (127)$$

$$= 0.54 - 0.46 \cdot \cos\left(\frac{2\pi}{N}\left(k + \frac{1}{2}\right)\right). \quad (128)$$

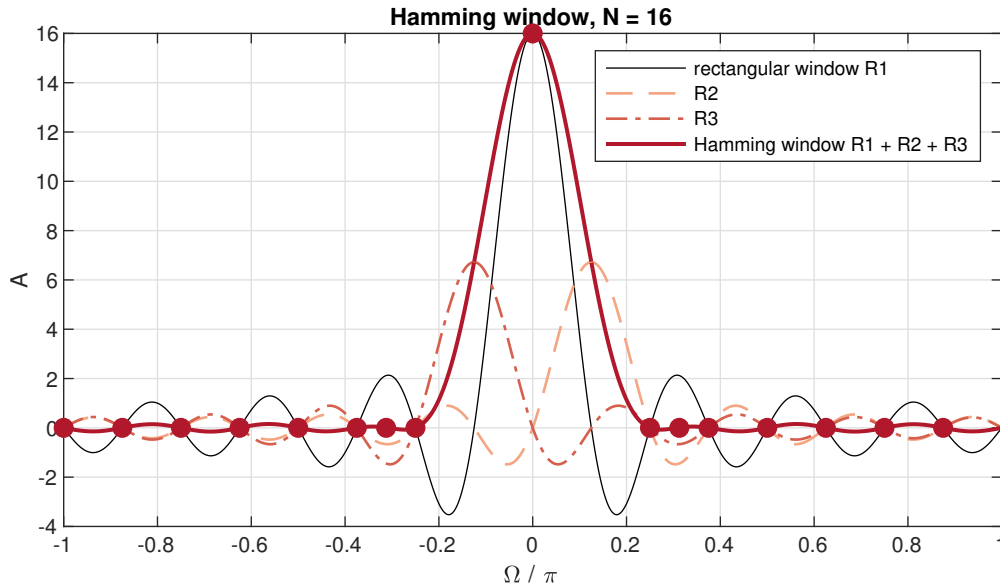


Figure 20: Spectrum of Hamming window from superposition of weighted rectangular windows.

Fig. 20 shows the additionally added zeros which leads to strongly attenuated first side lobes visible in the logarithmic spectrum in Fig. 21. Fig. 22 displays a comparison of the rectangular, Hanning and Hamming window. It shows that the strongly attenuated first side lobes of the Hamming window are traded for a slower decrease of the side lobes, but the main lobe is slightly narrower than for the Hanning window. Fig. 23 shows the best and worst case scenario for a Hamming window with $\beta = 0.84$. Compared to the Hanning window the secondary maxima are lower, but the amplitude error in the worst case is larger. It is approx. -1.78 dB and is termed "scalloping loss" in [Har78].

2 Exercises

Exercise 1: dft/idft Implementation

Write Matlab/Python functions $X = \text{my_dft}(x)$ and $x = \text{my_idft}(X)$ that calculate the DFT/IDFT-pair in eq. (11) and (12) without using the pre-built functions `fft()` and `ifft()` functions. Check validity and performance against the built-in functions (try large N). You might consider the matrix operation approach rather than a for-loop implementation.

Solution: Python TBD

→ Matlab: `UE1_Exercise1_Test_dft_idft_functions.m` together with `my_dft.m`, `my_idft.m` and `get_scaledSpectrum.m`

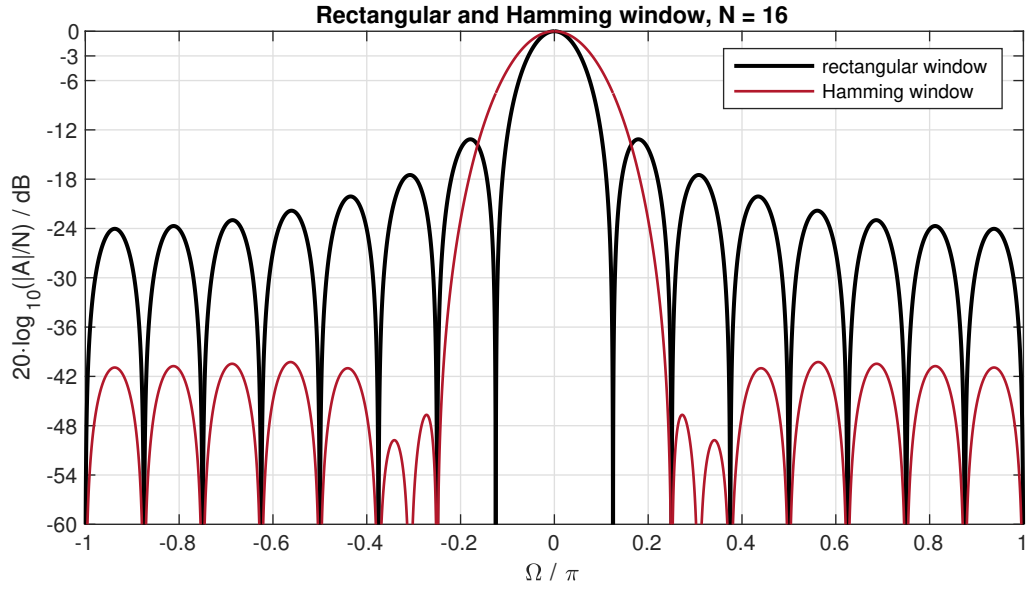


Figure 21: Level of DTFT spectra for rectangular and Hamming window.

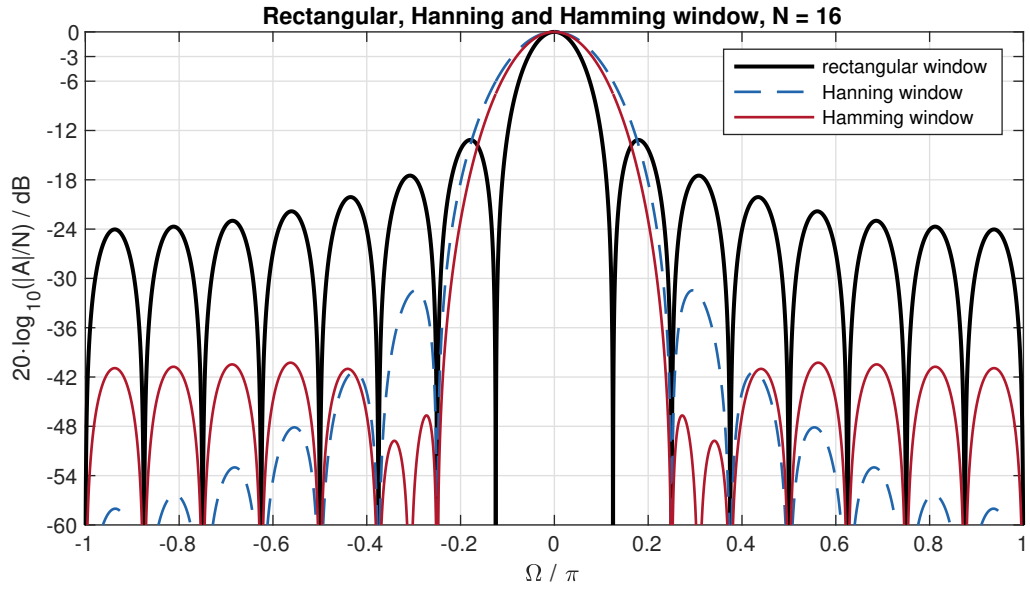


Figure 22: Level of DTFT spectra for rectangular, Hanning and Hamming window.

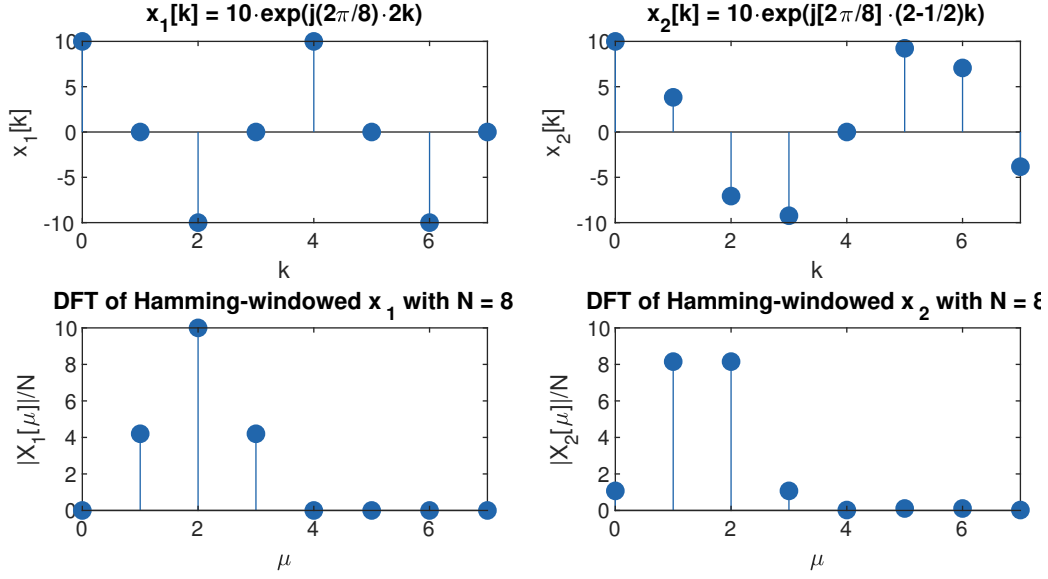


Figure 23: Hamming windowing for $N = 8$, best case ($\mu = 2$), worst case ($\mu = 2 - \frac{1}{2}$).

Exercise 2: IDFT

The discrete-time signal $x[k]$

$$x[k] = -2 \cdot \sin\left(\frac{2\pi}{4}k\right) + 3 \cdot \cos\left(\frac{2\pi}{4} \cdot 2k\right) + 1 \quad \text{for } 0 \leq k \leq 3 \quad (129)$$

with $k \in \mathbb{Z}$ is given.

- Calculate the resulting values of $x[k]$ for $0 \leq k \leq 3$.
- Show analytically that the given values of $X[\mu]$, $\mu \in \mathbb{Z}$:

$$X[\mu = 0] = 1, \quad X[\mu = 1] = j, \quad X[\mu = 2] = 3, \quad (130)$$

are the DFT coefficients of $x[k]$ stemming from

$$X[\mu] = \frac{1}{N} \sum_{k=0}^{N-1} x[k] \cdot e^{-j\frac{2\pi}{N}k\mu} \quad (131)$$

with $N = 4$. The following procedure is suggested: Setup the spectral coefficients $X[\mu]$ in the form

$$X[\mu] = A[\mu] \cdot e^{j\phi[\mu]} \quad (132)$$

and specify the missing value $X[\mu = 3]$ so that the IDFT results in $x[k] \in \mathbb{R}$. Then calculate the IDFT as

$$x[k] = \sum_{\mu=0}^{N-1} X[\mu] \cdot e^{j\frac{2\pi}{N}k\mu} \quad (133)$$

showing that this corresponds to the given signal $x[k]$.

- c) Plot the real and imaginary part as well as the magnitude and the phase of $X[\mu]$ over μ .
- d) A DFT-based audio analyser shall exhibit a frequency resolution of $\Delta f = 0.5$ Hz for a sampling frequency $f_s = 44100$ Hz using a rectangular window. Determine the minimum required DFT length N when only lengths $N = 2^M$ ($M \in \mathbb{N}$) are allowed. What is the resulting frequency resolution then?
- e) Check this example with the $N = 4$ DFT Matrix $\mathbf{W} = [\mathbf{w}_0 \ \mathbf{w}_1 \ \mathbf{w}_2 \ \mathbf{w}_3]$ for the set of linear equations

$$\mathbf{X}_{4 \times 1} = \mathbf{W}_{4 \times 4} \cdot \mathbf{x}_{4 \times 1} \quad (134)$$

by interpreting the matrix multiplication as linear combination of orthogonal \mathbf{w} vectors

$$\mathbf{X}_{4 \times 1} = x[0]\mathbf{w}_0 + x[1]\mathbf{w}_1 + x[2]\mathbf{w}_2 + x[3]\mathbf{w}_3. \quad (135)$$

Solution:

a)

$$x[0] = -2 \cdot \sin\left(2\pi \frac{1}{4} \cdot 0\right) + 3 \cdot \cos\left(2\pi \frac{2}{4} \cdot 0\right) + 1 = 3 + 1 = 4$$

$$x[1] = -2 \cdot \sin\left(2\pi \frac{1}{4} \cdot 1\right) + 3 \cdot \cos\left(2\pi \frac{2}{4} \cdot 1\right) + 1 = -2 - 3 + 1 = -4$$

$$x[2] = -2 \cdot \sin\left(2\pi \frac{1}{4} \cdot 2\right) + 3 \cdot \cos\left(2\pi \frac{2}{4} \cdot 2\right) + 1 = 3 + 1 = 4$$

$$x[3] = -2 \cdot \sin\left(2\pi \frac{1}{4} \cdot 3\right) + 3 \cdot \cos\left(2\pi \frac{2}{4} \cdot 3\right) + 1 = 2 - 3 + 1 = 0$$

b)

$$X[\mu = 0] = 1 = 1 \cdot e^{j \cdot 0}$$

$$X[\mu = 1] = j = 1 \cdot e^{j \frac{\pi}{2}}$$

$$X[\mu = 2] = 3 = 3 \cdot e^{j \cdot 0}$$

For $x[k] \in \mathbb{R}$ the symmetry $X[1]^* = X[3]$ holds, thus

$$X[\mu = 3] = 1 \cdot e^{-j \cdot \frac{\pi}{2}}.$$

The IDFT is given as

$$\begin{aligned} x[k] &= \sum_{\mu=0}^{N-1} X[\mu] \cdot e^{j \frac{2\pi}{N} k \mu} \\ &= \sum_{\mu=0}^3 X[\mu] \cdot e^{j \frac{\pi}{2} k \mu}. \end{aligned}$$

With the spectral coefficients in the form $X[\mu] = A[\mu] \cdot e^{j\phi[\mu]}$ this simplifies to

$$\begin{aligned}
 x[k] &= 1 \cdot e^{j \cdot 0} \cdot e^{j \frac{\pi}{2} k \cdot 0} + 1 \cdot e^{j \cdot \frac{\pi}{2}} \cdot e^{j \frac{\pi}{2} k \cdot 1} + 3 \cdot e^{j \cdot 0} \cdot e^{j \frac{\pi}{2} k \cdot 2} + 1 \cdot e^{-j \cdot \frac{\pi}{2}} \cdot e^{j \frac{\pi}{2} k \cdot 3} \\
 &= 1 + e^{j \frac{\pi}{2}} \cdot e^{j \frac{\pi}{2} k} + 3 \cdot e^{j \pi k} + e^{-j \frac{\pi}{2}} \cdot e^{j \frac{3\pi}{2} k} \\
 &= 1 + e^{j \frac{\pi}{2}} \cdot e^{j \frac{\pi}{2} k} + 3 \cdot \underbrace{\left(\cos(\pi k) + j \cdot \sin(\pi k) \right)}_{=0 \text{ if } k \in \mathbb{Z}} - e^{j \frac{\pi}{2}} \cdot e^{-j \frac{\pi}{2} k} \\
 &= 1 + 3 \cdot \cos(\pi k) + \underbrace{e^{j \frac{\pi}{2}}}_{=j} \cdot \left(e^{j \frac{\pi}{2} k} - e^{-j \frac{\pi}{2} k} \right).
 \end{aligned}$$

With Euler's identity

$$2j \cdot \sin(x) = e^{jx} - e^{-jx}$$

one gets

$$\begin{aligned}
 x[k] &= 1 + 3 \cdot \cos(\pi k) + j \cdot 2j \cdot \sin\left(\frac{\pi}{2} k\right) \\
 &= 1 + 3 \cdot \cos(\pi k) - 2 \cdot \sin\left(\frac{\pi}{2} k\right)
 \end{aligned}$$

which finally results as expected

$$x[k] = 1 + 3 \cdot \cos\left(\frac{2\pi}{4} \cdot 2k\right) - 2 \cdot \sin\left(\frac{2\pi}{4} k\right).$$

c)

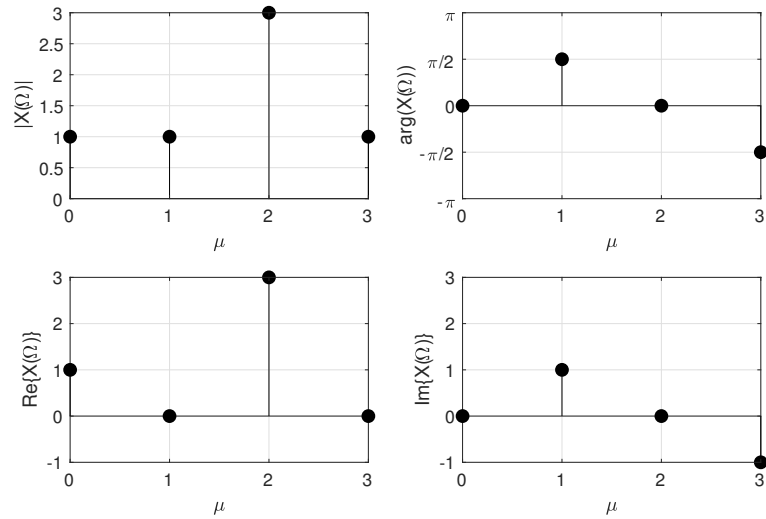


Figure 24: DFT of $x[k]$ for exercise 2 c).

d)

$$\begin{aligned}
\Delta f &= \frac{f_s}{N} = \frac{f_s}{2^M} \stackrel{!}{=} 0.5 \text{ Hz} \\
M &= \left\lceil \log_2 \left(\frac{f_s}{\Delta f} \right) \right\rceil_{\in \mathbb{N}} \\
&= \left\lceil \log_2 \left(\frac{44100 \text{ Hz}}{0.5 \text{ Hz}} \right) \right\rceil_{\in \mathbb{N}} \\
&= \lceil 16.4285... \rceil_{\in \mathbb{N}} \\
&= 17 \\
\Rightarrow N &= 2^M = 2^{17} = 131072
\end{aligned}$$

The resulting frequency resolution is thus

$$\Delta f = \frac{f_s}{N} = \frac{44100 \text{ Hz}}{131072} \approx 0.3365 \text{ Hz}.$$

e) Pay attention that this DFT is $1/N$ normalized. Let us build the (4×4) matrix (element-wise)

$$\mathbf{W} = e^{-j\frac{2\pi}{4}\mathbf{A}} \quad (136)$$

from the twiddle factor

$$W = e^{-j\frac{2\pi}{2}} \quad (137)$$

and from matrix (this is an outer product)

$$\mathbf{A} = \begin{bmatrix} 0 \\ 1 \\ 2 \\ 3 \end{bmatrix} \cdot [0 \ 1 \ 2 \ 3] = \mathbf{A} = \begin{bmatrix} 0 & 0 & 0 & 0 \\ 0 & 1 & 2 & 3 \\ 0 & 2 & 4 & 6 \\ 0 & 3 & 6 & 9 \end{bmatrix} \quad (138)$$

containing all possible products $k\mu$ in a suitable arrangement. We get

$$\mathbf{W} = \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & -j & -1 & j \\ 1 & -1 & 1 & -1 \\ 1 & j & -1 & -j \end{bmatrix}, \mathbf{w}_1 = \begin{bmatrix} 1 \\ 1 \\ 1 \\ 1 \end{bmatrix}, \mathbf{w}_2 = \begin{bmatrix} 1 \\ -j \\ -1 \\ j \end{bmatrix}, \mathbf{w}_3 = \begin{bmatrix} 1 \\ -1 \\ 1 \\ -1 \end{bmatrix}, \mathbf{w}_4 = \begin{bmatrix} 1 \\ j \\ -1 \\ -j \end{bmatrix}. \quad (139)$$

Thus,

$$4 \cdot \mathbf{X}_{4 \times 1} = x[0] \mathbf{w}_0 + x[1] \mathbf{w}_1 + x[2] \mathbf{w}_2 + x[3] \mathbf{w}_3. \quad (140)$$

becomes

$$4 \cdot \mathbf{X}_{4 \times 1} = 4 \cdot \begin{bmatrix} 1 \\ j \\ 3 \\ ? \end{bmatrix} = 4 \begin{bmatrix} 1 \\ 1 \\ 1 \\ 1 \end{bmatrix} - 4 \begin{bmatrix} 1 \\ -j \\ -1 \\ j \end{bmatrix} + 4 \begin{bmatrix} 1 \\ -1 \\ 1 \\ -1 \end{bmatrix} + 0 \begin{bmatrix} 1 \\ j \\ -1 \\ -j \end{bmatrix} \quad (141)$$

where the ? entry is easily deduced to $-j$. Think about, what this linear combination is telling you. Think of the DFT eigensignals (rows) and index of DFT eigenfrequencies (columns) and the signal samples acting as weights.

Exercise 3: DFT analysis using a rectangular window

A sine signal $x[k] = \cos(\Omega k)$ with $\Omega = 2 \cdot \frac{2\pi}{N}$, $N = 8$, $0 \leq k \leq N - 1$ is to be analysed with the DFT eq. (11) assuming a sampling frequency of $f_s = 48$ kHz.

- Calculate the spectrum $X[\mu]$ of $x[k]$ and visualise the real and imaginary part as well as the magnitude and the phase of $X[\mu]$ over $0 \leq \mu \leq N - 1$.
- Check the expected symmetries.
- Implement the interpolation of eq. (55) and visualise this over μ , Ω as well as f as a magnitude spectrum together with $|X[\mu]|$.
- Repeat the steps a) to c) for $N = 9$. What is different?
- Repeat the steps a) to d) for $\Omega = 2.5 \cdot \frac{2\pi}{N}$. What is different now?

Solution: Python TBD

→ Matlab: `UE1_Exercise3_DFT_analysis.m` together with `interpolate_DFT.m`

Exercise 4: DFT of an Impulse Response

The finite length impulse response (FIR filter) $h[k]$ of an LTI system is given as

$$h[k] = \frac{1}{8} \cdot (11 \cdot \delta[k] - 5 \cdot \delta[k - 1] + 7 \cdot \delta[k - 2] - 9 \cdot \delta[k - 3]) \quad (142)$$

- Calculate the DFT analytically

$$H[\mu] = \sum_{k=0}^{N-1} h[k] \cdot e^{-j\frac{2\pi}{N}k\mu} \quad (143)$$

for $N = 4$ and $0 \leq k \leq N - 1$. The DFT spectrum corresponds to the transfer function of the system.

- Calculate the magnitude $|H[\mu]|$ and the phase response $\arg(H[\mu])$ for $0 \leq \mu \leq 3$. State the magnitude as level in dB and the phase in degrees.
- The frequency resolution is assumed to be $\Delta f = 500$ Hz. Sketch the DFT line spectrum and the interpolated DTFT spectrum of the magnitude $|H[\mu]|$ in dB and the phase $\arg(H[\mu])$ in degrees over the frequency axis from $0 \text{ Hz} \leq f \leq 4000 \text{ Hz}$. Estimate the sampling frequency f_s from the known parameters and information. What filter characteristic is realised with this impulse response?

Solution:

a) For $N = 4$, the DFT is

$$H[\mu] = \sum_{k=0}^3 h[k] e^{-j\frac{2\pi}{4}\mu k} = \sum_{k=0}^3 h[k] e^{-j\frac{\pi}{2}\mu k}$$

with the twiddle factors

$$W_4^{\mu k} = e^{-j\frac{\pi}{2}\mu k}.$$

Again, the DFT matrix is built:

$W_4^{\mu k}$	$k = 0$	$k = 1$	$k = 2$	$k = 3$
$\mu = 0$	1	1	1	1
$\mu = 1$	1	-j	-1	j
$\mu = 2$	1	-1	1	-1
$\mu = 3$	1	j	-1	-j

Hence, for $H[\mu]$ follows (this time using the row times column matrix multiplication approach written in detail)

$$H[0] = \sum_{k=0}^3 W_4^{0 \cdot k} \cdot h[k] = \left(\frac{11}{8} \cdot 1\right) + \left(\frac{-5}{8} \cdot 1\right) + \left(\frac{7}{8} \cdot 1\right) + \left(\frac{-9}{8} \cdot 1\right) = \frac{4}{8} = \frac{1}{2}$$

$$H[1] = \sum_{k=0}^3 W_4^{1 \cdot k} \cdot h[k] = \left(\frac{11}{8} \cdot 1\right) + \left(\frac{-5}{8} \cdot (-j)\right) + \left(\frac{7}{8} \cdot (-1)\right) + \left(\frac{-9}{8} \cdot j\right) = \frac{4}{8} - j\frac{4}{8} = \frac{1}{2} - j\frac{1}{2}$$

$$H[2] = \sum_{k=0}^3 W_4^{2 \cdot k} \cdot h[k] = \left(\frac{11}{8} \cdot 1\right) + \left(\frac{-5}{8} \cdot (-1)\right) + \left(\frac{7}{8} \cdot 1\right) + \left(\frac{-9}{8} \cdot (-1)\right) = \frac{32}{8} = 4$$

$$H[3] = \sum_{k=0}^3 W_4^{3 \cdot k} \cdot h[k] = \left(\frac{11}{8} \cdot 1\right) + \left(\frac{-5}{8} \cdot j\right) + \left(\frac{7}{8} \cdot (-1)\right) + \left(\frac{-9}{8} \cdot (-j)\right) = \frac{4}{8} + j\frac{4}{8} = \frac{1}{2} + j\frac{1}{2}$$

b)

$$|H[0]| = \frac{1}{2} \longrightarrow 20 \log_{10} \left(\frac{1}{2} \right) = -6.02 \text{ dB}$$

$$|H[1]| = \sqrt{\frac{1}{2^2} + \frac{1}{2^2}} = \frac{1}{\sqrt{2}} \longrightarrow 20 \log_{10} \left(\frac{1}{\sqrt{2}} \right) = -3.01 \text{ dB}$$

$$|H[2]| = 4 \longrightarrow 20 \log_{10}(4) = 12.04 \text{ dB}$$

$$|H[3]| = \sqrt{\frac{1}{2^2} + \frac{1}{2^2}} = \frac{1}{\sqrt{2}} \longrightarrow 20 \log_{10} \left(\frac{1}{\sqrt{2}} \right) = -3.01 \text{ dB}$$

$$\arg(H[0]) = 0$$

$$\arg(H[1]) = \arctan \left(\frac{-\frac{1}{2}}{\frac{1}{2}} \right) = -0.7854 \hat{=} -45^\circ$$

$$\arg(H[2]) = 0$$

$$\arg(H[3]) = \arctan \left(\frac{\frac{1}{2}}{\frac{1}{2}} \right) = 0.7854 \hat{=} 45^\circ$$

c)

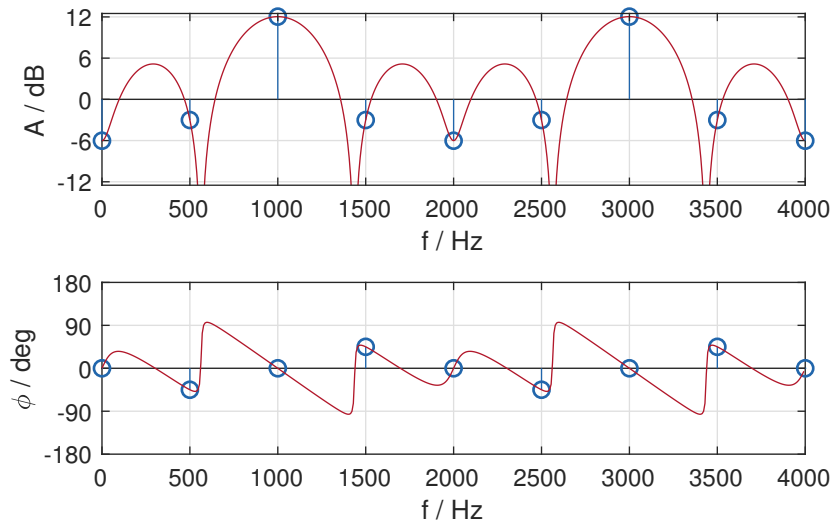


Figure 25: DFT and interpolated DTFT spectrum for $h[k]$

sampling frequency: $\Delta f = \frac{f_s}{N} \Leftrightarrow f_s = \Delta f \cdot N = 500 \text{ Hz} \cdot 4 = 2000 \text{ Hz}$

filter characteristic: highpass filter, -3 dB corner frequency at about 790 Hz, relative sidelobe level about -7 dB, notch at about 560 Hz, 12 dB gain at $\frac{f_s}{2}$

Exercise 5: DFT Parameterisation

A composite signal $x[k] = A_1 \cdot e^{j(\Omega_1 k + \phi_1)} + A_2 \cdot e^{j(\Omega_2 k + \phi_2)}$ with the known frequencies $\Omega_1 = \frac{1}{30}\pi$ und $\Omega_2 = \frac{1}{4}\pi$ but unknown amplitudes and phase relation is given.

- What DFT length N must be set up so that the exact amplitude and phase values for both frequencies Ω_1 and Ω_2 can be determined for a rectangularly windowed signal (i.e. no leakage occurs)?
- In which bins are the frequencies Ω_1 and Ω_2 then found?
- What amplitude deviation occurs when analysing the signal $x[k] = e^{j\Omega_3 k}$ with $\Omega_3 = \frac{30.5}{60}\pi$?

Solution:

- DFT eigenfrequencies: $\Omega_{\text{DFT}} = \frac{2\pi}{N}\mu$

Ω_1 and Ω_2 must be DFT eigenfrequencies:

$$\begin{aligned} \Omega_1 = \frac{\pi}{30} = \frac{2\pi}{N}\mu & \Leftrightarrow N = 60 \cdot \mu \\ \Omega_2 = \frac{\pi}{4} = \frac{2\pi}{N}\mu & \Leftrightarrow N = 8 \cdot \mu \end{aligned}$$

Both conditions together can be fulfilled from $N = 120$ (or multiples thereof).

b)

$$\begin{aligned}\mu_1 &= \Omega_1 \cdot \frac{N}{2\pi} = \frac{\pi}{30} \cdot \frac{120}{2\pi} = 2, & \text{i.e. the 3rd bin} \\ \mu_2 &= \Omega_2 \cdot \frac{N}{2\pi} = \frac{\pi}{4} \cdot \frac{120}{2\pi} = 15, & \text{i.e. the 16th bin}\end{aligned}$$

c) The discrete angular frequency $\Omega_3 = \frac{30.5}{60}\pi$ can be expressed in terms of multiples of the first DFT eigenfrequency $\frac{2\pi}{N}$ as

$$\Omega_3 = n \cdot \frac{2\pi}{N} \quad \leftrightarrow \quad n = \Omega_3 \cdot \frac{N}{2\pi} = \frac{30.5}{60}\pi \cdot \frac{120}{2\pi} = 30.5$$

so that

$$\Omega_3 = 30.5 \cdot \frac{2\pi}{N}.$$

The frequency Ω_3 is not a DFT eigenfrequency as it is not an integer multiple of the first DFT eigenfrequency, but is located in the middle between bins 31 and 32 (for $\mu = 30$ and $\mu = 31$ counting from $\mu = 0$). The neighbouring bins are the bins that can give the best information about the amplitude of the true signal. The spectral values at the neighbouring bins can be calculated by the DFT of a complex exponential (see formularies)

$$e^{j\Omega_3 k} \quad \circ \text{---} \bullet \quad e^{j \frac{(\Omega_3 - \frac{2\pi}{N}\mu)(N-1)}{2}} \cdot \frac{\sin\left(N \frac{\Omega_3 - \frac{2\pi}{N}\mu}{2}\right)}{\sin\left(\frac{\Omega_3 - \frac{2\pi}{N}\mu}{2}\right)}.$$

At $\mu = 30$, the amplitude is

$$\begin{aligned}|X[\mu = 30]| &= \left| e^{j \frac{(\Omega_3 - \frac{2\pi}{N}\mu)(N-1)}{2}} \cdot \frac{\sin\left(N \frac{\Omega_3 - \frac{2\pi}{N}\mu}{2}\right)}{\sin\left(\frac{\Omega_3 - \frac{2\pi}{N}\mu}{2}\right)} \right| \\ &= \frac{\sin\left(N \frac{\Omega_3 - \frac{2\pi}{N}\mu}{2}\right)}{\sin\left(\frac{\Omega_3 - \frac{2\pi}{N}\mu}{2}\right)} \\ &= \frac{\sin\left(N \frac{30.5 \cdot \frac{2\pi}{N} - 30 \cdot \frac{2\pi}{N}}{2}\right)}{\sin\left(\frac{30.5 \cdot \frac{2\pi}{N} - 30 \cdot \frac{2\pi}{N}}{2}\right)} \\ &= \frac{\sin\left(N \frac{\frac{1}{2} \cdot \frac{2\pi}{N}}{2}\right)}{\sin\left(\frac{\frac{1}{2} \cdot \frac{2\pi}{N}}{2}\right)} \\ &= \frac{\sin\left(\frac{\pi}{2}\right)}{\sin\left(\frac{\pi}{2N}\right)} \\ &= \frac{1}{\sin\left(\frac{\pi}{240}\right)} \\ &\approx 76.3966.\end{aligned}$$

At $\mu = 31$, the amplitude is

$$\begin{aligned} |X[\mu = 31]| &= \frac{\sin\left(N \frac{30.5 \cdot \frac{2\pi}{N} - 31 \cdot \frac{2\pi}{N}}{2}\right)}{\sin\left(\frac{30.5 \cdot \frac{2\pi}{N} - 31 \cdot \frac{2\pi}{N}}{2}\right)} \\ &= \frac{\sin\left(-\frac{\pi}{2}\right)}{\sin\left(-\frac{\pi}{2N}\right)} \\ &\approx 76.3966. \end{aligned}$$

If the frequency of the complex exponential had been exactly at a DFT eigenfrequency, e.g. at $\mu = 31$, the DFT spectrum would have contained the true amplitude of the signal. The rule of L'Hospital is needed to calculate this amplitude value:

$$\begin{aligned} &\lim_{\Omega_3 \rightarrow 31 \cdot \frac{2\pi}{N}} \frac{\sin\left(N \frac{\Omega_3 - \frac{2\pi}{N} 31}{2}\right)}{\sin\left(\frac{\Omega_3 - \frac{2\pi}{N} 31}{2}\right)} \\ &= \lim_{\Omega_3 \rightarrow 31 \cdot \frac{2\pi}{N}} \frac{\cos\left(N \frac{\Omega_3 - \frac{2\pi}{N} 31}{2}\right) \cdot \frac{N}{2}}{\cos\left(\frac{\Omega_3 - \frac{2\pi}{N} 31}{2}\right) \cdot \frac{1}{2}} \\ &= \frac{\cos 0}{\cos 0} \cdot N \\ &= N \\ &= 120. \end{aligned}$$

The error is thus

$$20 \cdot \log_{10}\left(\frac{76.3966}{120}\right) = -3.9221 \text{ dB},$$

i.e. the magnitude is underestimated. The case when the signal frequency lies exactly in middle between two DFT eigenfrequencies is the worst case scenario, in [Har78] termed "worst case process loss".

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