

Numerical Methods II

Fourier Transforms and the FFT

Aleksandar Donev
Courant Institute, NYU¹
donev@courant.nyu.edu

¹MATH-GA.2020-001 / CSCI-GA.2421-001, Spring 2023

Jan 24th, 2023

Outline

- 1 Logistics
- 2 Trigonometric Orthogonal Polynomials
- 3 Approximation Theory
- 4 Fast Fourier Transform
- 5 Conclusions

Logistics

Course Essentials

- Course webpage: <https://adonev.github.io/NumMethII>
- Registered students: NYU Brightspace for announcements, submitting homeworks, grades, and sample solutions. **Make sure you have access.**
- Office hours (**TBD**): 3:30-5:30 pm Wednesdays (TBD, or by appointment). Grader's office hours TBD.
- Main textbooks available in PDF format inside the NYU network/proxy: **LeVeque**, Trefethen (see course homepage).
- Other **optional readings** linked on course page; Ph.D. students should consult those sources.
- Computing is an essential part: **MATLAB** forms a common platform but (scientific/numerical) **Python/numpy** or **Julia** are great too.

To-do

- Assignment 0 is a **Questionnaire** with basic statistics about you: submit via email as **plain text** or **PDF**.
- **Download PDFs** of all textbooks for reference.
- There will be regular **homework assignments** (50% of grade), mostly computational. Points from all assignments will be added together.
- Submit the solutions as a PDF (give LaTeX/lyx a try!), via NYU Brightspace. **First assignment posted already and due in two weeks.**
Due time is 5pm the day of class.
- You can choose between a **take-home final** or **final project** (50%), due **TBD**.

Academic Integrity Policy

- If you use any external source, even Wikipedia, make sure you acknowledge it by **referencing all help**.
- It is encouraged to **discuss** with other students the mathematical aspects, algorithmic strategy, code design, techniques for debugging, and compare results.
- Copying of any portion of someone else's solution or allowing others to copy your solution is considered **cheating**.
- **Code sharing is not allowed**: You must write/debug/run your own code.
- Submitting an **individual and independent final** is crucial and **no collaboration** will be allowed for the final.
- Common bad justifications for copying:
 - We are too busy and the homework is very hard, so we cannot do it on our own.
 - We do not copy each other but rather “work together.”
 - I just emailed Joe Doe my solution as a “reference.”

Grading Standards

- **Points** will be **added** over all assignments (50%) and the take-home final (50%).
- **No makeup** points (solutions may be posted on NYU Brightspace).
- The actual grades will be rounded upward (e.g., for those that are close to a boundary), but not downward:
 - 92.5-max = A
 - 87.5-92.5 = A-
 - 80.0-87.5 = B+
 - 72.5-80.0 = B
 - 65.0-72.5 = B-
 - 57.5-65.0 = C+
 - 50.0-57.5 = C
 - 42.5-50.0 = C-
 - min-42.5 = F

Trigonometric Orthogonal Polynomials

Periodic Functions

- Consider now interpolating / approximating **periodic functions** defined on the interval $I = [0, 2\pi]$:

$$\forall x \quad f(x + 2\pi) = f(x),$$

as appear in practice when analyzing signals (e.g., sound/image processing).

- Also consider only the space of complex-valued **square-integrable functions** $L^2_{2\pi}$,

$$\forall f \in L^2_w : \quad (f, f) = \|f\|^2 = \int_0^{2\pi} |f(x)|^2 dx < \infty.$$

- Polynomial functions are not periodic and thus basis sets based on orthogonal polynomials are not appropriate.
- Instead, consider sines and cosines as a basis function, combined together into **complex exponential functions**

$$\phi_k(x) = e^{ikx} = \cos(kx) + i \sin(kx), \quad k = 0, \pm 1, \pm 2, \dots$$

Fourier Basis Functions

$$\phi_k(x) = e^{ikx}, \quad k = 0, \pm 1, \pm 2, \dots$$

- It is easy to see that these are **orthogonal** with respect to the continuous dot product

$$(\phi_j, \phi_k) = \int_{x=0}^{2\pi} \phi_j(x) \phi_k^*(x) dx = \int_0^{2\pi} \exp[i(j-k)x] dx = 2\pi \delta_{ij}$$

- The complex exponentials can be shown to form a complete **trigonometric polynomial basis** for the space $L^2_{2\pi}$, i.e.,

$$\forall f \in L^2_{2\pi} : \quad f(x) = \sum_{k=-\infty}^{\infty} \hat{f}_k e^{ikx},$$

where the **Fourier coefficients** can be computed for any **frequency** or **wavenumber** k using:

$$\hat{f}_k = \frac{(f, \phi_k)}{2\pi} = \frac{1}{2\pi} \cdot \int_0^{2\pi} f(x) e^{-ikx} dx.$$

Truncated Fourier Basis

- For a general interval $[0, X]$ the **discrete frequencies** are

$$k = \frac{2\pi}{X} \kappa \quad \kappa = 0, \pm 1, \pm 2, \dots$$

- For non-periodic functions one can take the limit $X \rightarrow \infty$ in which case we get **continuous frequencies**.
- Now consider a **discrete Fourier basis** that only includes the first N basis functions, i.e.,

$$\begin{cases} k = -(N-1)/2, \dots, 0, \dots, (N-1)/2 & \text{if } N \text{ is odd} \\ k = -N/2, \dots, 0, \dots, N/2 - 1 & \text{if } N \text{ is even,} \end{cases}$$

and for simplicity we focus on N odd.

- The least-squares **spectral approximation** for this basis is:

$$f(x) \approx \phi(x) = \sum_{k=-(N-1)/2}^{(N-1)/2} \hat{f}_k e^{ikx}.$$

Discrete Dot Product

- Now also discretize a given function on a set of N **equi-spaced nodes**

$$x_j = jh \text{ where } h = \frac{2\pi}{N}$$

where $j = N$ is the same node as $j = 0$ due to periodicity so we only consider N instead of $N + 1$ nodes.

- We also have the **discrete dot product** between two discrete functions (vectors) $\mathbf{f}_j = f(x_j)$:

$$\mathbf{f} \cdot \mathbf{g} = h \sum_{j=0}^{N-1} f_j g_j^*$$

- The discrete Fourier basis is **discretely orthogonal**

$$\phi_k \cdot \phi_{k'} = 2\pi \delta_{k,k'}$$

Proof of Discrete Orthogonality

The case $k = k'$ is trivial, so focus on

$$\phi_k \cdot \phi_{k'} = 0 \text{ for } k \neq k'$$

$$\sum_j \exp(ikx_j) \exp(-ik'x_j) = \sum_j \exp[i(\Delta k)x_j] = \sum_{j=0}^{N-1} [\exp(ih(\Delta k))]^j$$

where $\Delta k = k - k'$. This is a geometric series sum:

$$\phi_k \cdot \phi_{k'} = \frac{1 - z^N}{1 - z} = 0 \text{ if } k \neq k'$$

since $z = \exp(ih(\Delta k)) \neq 1$ and
 $z^N = \exp(ihN(\Delta k)) = \exp(2\pi i(\Delta k)) = 1$.

Discrete Fourier Transform

- The **Fourier interpolating polynomial** is thus easy to construct

$$\phi_N(x) = \sum_{k=-(N-1)/2}^{(N-1)/2} \hat{f}_k^{(N)} e^{ikx}$$

where the **discrete Fourier coefficients** are given by

$$\hat{f}_k^{(N)} = \frac{\mathbf{f} \cdot \boldsymbol{\phi}_k}{2\pi} = \frac{1}{N} \sum_{j=0}^{N-1} f(x_j) \exp(-ikx_j)$$

- Simplifying the notation and recalling $x_j = jh$, we define the the **Discrete Fourier Transform** (DFT):

$$\hat{f}_k^{(N)} = \frac{1}{N} \sum_{j=0}^{N-1} f_j \exp\left(-\frac{2\pi ijk}{N}\right)$$

Discrete spectrum

- The set of discrete Fourier coefficients $\hat{\mathbf{f}}$ is called the **discrete spectrum**, and in particular,

$$S_k = \left| \hat{f}_k \right|^2 = \hat{f}_k \hat{f}_k^*,$$

is the **power spectrum** which measures the frequency content of a signal.

- If f is real, then \hat{f} satisfies the **conjugacy property**

$$\hat{f}_{-k} = \hat{f}_k^*,$$

so that half of the spectrum is redundant and \hat{f}_0 is real.

- For an even number of points N the largest frequency $k = -N/2$ does not have a conjugate partner. It is special and **must be treated with care**.

Fourier Spectral Approximation

- **Discrete Fourier Transform (DFT):**

$$\text{Forward } \mathbf{f} \rightarrow \hat{\mathbf{f}} : \quad \hat{f}_k = \frac{1}{N} \sum_{j=0}^{N-1} f_j \exp \left(-\frac{2\pi i j k}{N} \right)$$

$$\text{Inverse } \hat{\mathbf{f}} \rightarrow f : \quad f(x_j) \approx \phi(x_j) = \sum_{k=-(N-1)/2}^{(N-1)/2} \hat{f}_k \exp \left(\frac{2\pi i j k}{N} \right)$$

- There is a very fast algorithm for performing the **forward and backward DFTs (FFT)**.
- There is **different conventions** for the DFT depending on the interval on which the function is defined and placement of factors of N and 2π .

Read the documentation to be consistent!

Spectral Convergence (or not)

- The Fourier interpolating polynomial $\phi(x)$ has **spectral accuracy**, i.e., exponential in the number of nodes N

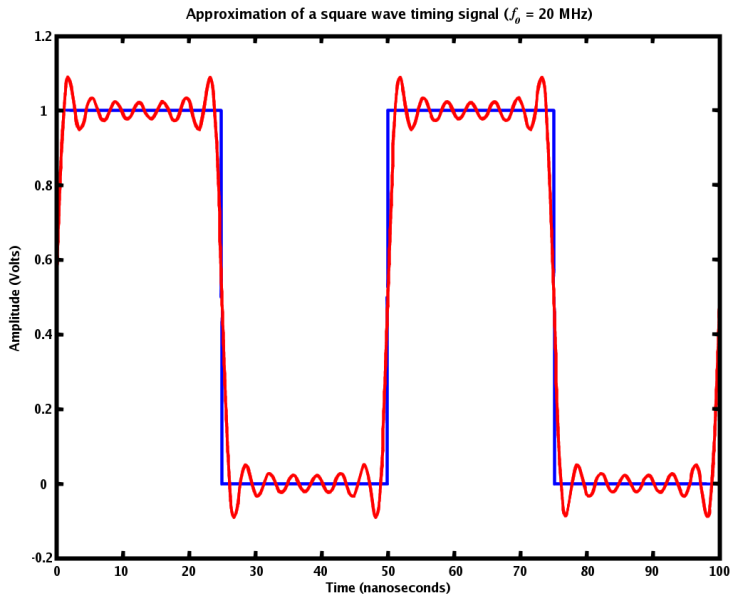
$$\|f(x) - \phi(x)\| \sim e^{-N}$$

for **analytic functions** (more details shortly).

- Specifically, nice functions exhibit **rapid decay of the Fourier coefficients** with k , e.g., exponential decay $|\hat{f}_k| \sim e^{-|k|}$.
- Discontinuities cause slowly-decaying Fourier coefficients, e.g., power law decay $|\hat{f}_k| \sim k^{-1}$ for **jump discontinuities**.
- Jump discontinuities lead to slow convergence of the Fourier series for non-singular points (and no convergence at all near the singularity), so-called **Gibbs phenomenon** (ringing):

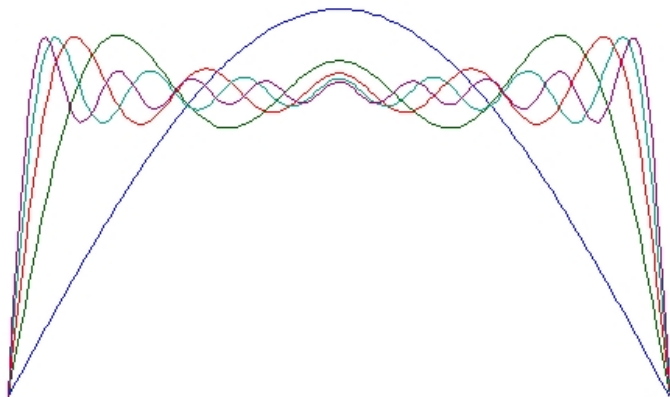
$$\|f(x) - \phi(x)\| \sim \begin{cases} N^{-1} & \text{at points away from jumps} \\ \text{const.} & \text{at the jumps themselves} \end{cases}$$

Gibbs Phenomenon



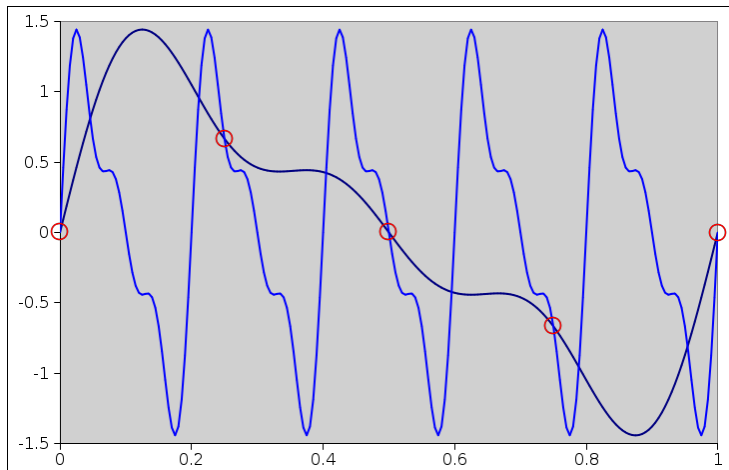
Gibbs Phenomenon

Reconstruction of the periodic square waveform with 1, 3, 5, 7, 9 sinusoids



Aliasing

If we sample a signal at too few points the Fourier interpolant may be wildly wrong: **aliasing** of frequencies k and $2k, 3k, \dots$



Approximation Theory

Trigonometric projection vs. interpolation

- I will temporarily switch to notation in paper on periodic chebfun in paper of Trefethen *et al*, assuming *odd* number of points for simplicity:

$f(t \in [0, 2\pi])$ discretized with $N = 2n + 1$ points $t_m = \frac{2\pi m}{N}$

Trigonometric projection:
$$f_n(t) = \sum_{k=-n}^n c_k e^{ikt}$$

Trigonometric interpolant:
$$p_n(t) = \sum_{k=-n}^n \tilde{c}_k e^{ikt}.$$

- Aliasing** means that one cannot distinguish two different Fourier modes on a given grid:

$$\exp(ikt_m) = \exp(i(k + jN)t_m)$$

Poisson Summation Formula

- Observe that because of aliasing:

$$\begin{aligned} f(t_m) &= \sum_{k=-\infty}^{\infty} c_k e^{ikt_m} = \sum_{k=-n}^n \sum_{j=-\infty}^{\infty} c_{k+jN} e^{i(k+jN)t_m} \\ &= \sum_{k=-n}^n \left(\sum_{j=-\infty}^{\infty} c_{k+jN} \right) e^{ikt_m} \end{aligned}$$

$$\text{Recall: } p_n(t_m) = \sum_{k=-n}^n \tilde{c}_k e^{ikt_m}$$

- Since the trigonometric interpolant is unique, we get Poisson's summation formula

$$\tilde{c}_k = \sum_{j=-\infty}^{\infty} c_{k+jN}$$

The importance of smoothness

- **Total variation** of differentiable function (can be generalized):

$$\text{TV}[f] = \int_0^{2\pi} |f'(x)| dx, \quad \text{denote } V = \text{TV} [f^{(\nu)}].$$

- We have two cases where we have nice error estimates:
 - If f is $\nu \geq 0$ times **differentiable**, then

$$|c_k| \leq \frac{V}{2\pi |k|^{\nu+1}}$$

which can be proved by integrating $c_k = (2\pi)^{-1} \int_0^{2\pi} f(x) e^{-ikx} dx$ by parts $\nu + 1$ times.

- If $f(t)$ is **analytic** in a half-strip around the real axis of half-width α and bounded by $|f(t)| < M$, then

$$|c_k| \leq M e^{-\alpha|k|}$$

which can be proved by shifting the contour of integration above or below the real line by α .

Approximation error: Differentiable

- If f is $\nu \geq 1$ times **differentiable** then

$$\begin{aligned}\|f - f_n\|_\infty &= \left\| \sum_{|k|>n} c_k e^{ikt} \right\|_\infty \leq \sum_{|k|>n} |c_k| \\ &\leq 2 \sum_{k=n+1}^{\infty} \frac{V}{2\pi k^{\nu+1}} \approx 2 \int_n^{\infty} \frac{V}{2\pi k^{\nu+1}} dk\end{aligned}$$

- Performing the integral we get that if f is $\nu \geq 1$ times **differentiable**, then

$$\|f - f_n\|_\infty \leq \frac{V}{\pi \nu n^\nu}$$

- You can replace f_n with p_n if you multiply the r.h.s. by 2 to account for the additional aliasing error.

Approximation error: Analytic

- If $f(t)$ is **analytic** in the half strip then

$$\|f - f_n\|_{\infty} \leq 2 \sum_{k=n+1}^{\infty} M e^{-\alpha k} = \frac{2M e^{-\alpha n}}{e^{\alpha} - 1} \text{ (geometric series sum)}$$

- You can replace f_n with p_n if you multiply the r.h.s. by 2 to account for the additional aliasing error.
- The Fourier interpolating trigonometric polynomial is spectrally accurate and a really great approximation for (very) smooth functions.

Trapezoidal Rule

- Consider using the trapezoidal rule to approximate a **periodic integral**:

$$I = \int_0^{2\pi} f(x) dx = c_0$$

$$I_N = \frac{2\pi}{N} \sum_{m=1}^N f(t_m) = \tilde{c}_0.$$

- If f is $\nu \geq 1$ times **differentiable** then

$$|I_N - I| \leq \frac{4V}{N^{\nu+1}}.$$

- If $f(t)$ is **analytic** in the half strip then **trapezoidal rule is spectrally accurate**:

$$|I_N - I| \leq \frac{4\pi M}{e^{\alpha N} - 1}.$$

Fast Fourier Transform

DFT

- Recall the transformation from real space to frequency space and back:

$$\mathbf{f} \rightarrow \hat{\mathbf{f}} : \quad \hat{f}_k = \frac{1}{N} \sum_{j=0}^{N-1} f_j \exp\left(-\frac{2\pi i j k}{N}\right), \quad k = -\frac{(N-1)}{2}, \dots, \frac{(N-1)}{2}$$

$$\hat{\mathbf{f}} \rightarrow \mathbf{f} : \quad f_j = \sum_{k=-(N-1)/2}^{(N-1)/2} \hat{f}_k \exp\left(\frac{2\pi i j k}{N}\right), \quad j = 0, \dots, N-1$$

- We can make the forward-reverse **Discrete Fourier Transform** (DFT) more symmetric if we shift the frequencies to $k = 0, \dots, N$:

$$\text{Forward } \mathbf{f} \rightarrow \hat{\mathbf{f}} : \quad \hat{f}_k = \frac{1}{\sqrt{N}} \sum_{j=0}^{N-1} f_j \exp\left(-\frac{2\pi i j k}{N}\right), \quad k = 0, \dots, N-1$$

$$\text{Inverse } \hat{\mathbf{f}} \rightarrow \mathbf{f} : \quad f_j = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \hat{f}_k \exp\left(\frac{2\pi i j k}{N}\right), \quad j = 0, \dots, N-1$$

FFT

- We can write the transforms in matrix notation:

$$\hat{\mathbf{f}} = \frac{1}{\sqrt{N}} \mathbf{U}_N \mathbf{f}$$

$$\mathbf{f} = \frac{1}{\sqrt{N}} \mathbf{U}_N^* \hat{\mathbf{f}},$$

where the **unitary Fourier matrix** (`fft(eye(N))` in MATLAB) is an $N \times N$ matrix with entries

$$u_{jk}^{(N)} = \omega_N^{jk}, \quad \omega_N = e^{-2\pi i/N}.$$

- A **direct** matrix-vector multiplication algorithm therefore takes $O(N^2)$ multiplications and additions.
- Is there a faster way to compute the **non-normalized**

$$\hat{f}_k = \sum_{j=0}^{N-1} f_j \omega_N^{jk} \quad ?$$

FFT

- For now assume that N is even and in fact a power of two, $N = 2^n$.
- The idea is to split the transform into two pieces, **even and odd** points:

$$\sum_{j=2j'} f_j \omega_N^{jk} + \sum_{j=2j'+1} f_j \omega_N^{jk} = \sum_{j'=0}^{N/2-1} f_{2j'} (\omega_N^2)^{j'k} + \omega_N^k \sum_{j'=0}^{N/2-1} f_{2j'+1} (\omega_N^2)^{j'k}$$

- Now notice that

$$\omega_N^2 = e^{-4\pi i/N} = e^{-2\pi i/(N/2)} = \omega_{N/2}$$

- This leads to a **divide-and-conquer algorithm**:

$$\hat{f}_k = \sum_{j'=0}^{N/2-1} f_{2j'} \omega_{N/2}^{j'k} + \omega_N^k \sum_{j'=0}^{N/2-1} f_{2j'+1} \omega_{N/2}^{j'k}$$

$$\hat{f}_k = \mathbf{U}_N \mathbf{f} = (\mathbf{U}_{N/2} \mathbf{f}_{\text{even}} + \omega_N^k \mathbf{U}_{N/2} \mathbf{f}_{\text{odd}})$$

FFT Complexity

- The **Fast Fourier Transform** algorithm is **recursive**:

$$FFT_N(\mathbf{f}) = FFT_{\frac{N}{2}}(\mathbf{f}_{\text{even}}) + \mathbf{w} \oslash FFT_{\frac{N}{2}}(\mathbf{f}_{\text{odd}}),$$

where $w_k = \omega_N^k$ and \oslash denotes element-wise product. When $N = 1$ the FFT is trivial (identity).

- To compute the whole transform we need $\log_2(N)$ steps, and at each step we only need N multiplications and $N/2$ additions at each step.
- The total **cost of FFT** is thus much better than the direct method's $O(N^2)$: **Log-linear**

$$O(N \log N).$$

- Even when N is not a power of two there are ways to do a similar **splitting** transformation of the large FFT into many smaller FFTs.
- Note that there are different **normalization conventions** used in different software.

In MATLAB

- The forward transform is performed by the function $\hat{f} = \text{fft}(f)$ and the inverse by $f = \text{ifft}(\hat{f})$. Note that $\text{ifft}(\text{fft}(f)) = f$ and f and \hat{f} may be complex.
- In MATLAB, and other software, the frequencies are not ordered in the “normal” way $-(N-1)/2$ to $+(N-1)/2$, but rather, the nonnegative frequencies come first, then the positive ones, so the “funny” ordering is

$$0, 1, \dots, (N-1)/2, \quad -\frac{N-1}{2}, -\frac{N-1}{2} + 1, \dots, -1.$$

This is because such ordering (shift) makes the forward and inverse transforms symmetric.

- The function *fftshift* can be used to order the frequencies in the “normal” way, and *ifftshift* does the reverse:

$$\hat{f} = \text{fftshift}(\text{fft}(f)) \text{ (normal ordering).}$$

FFT-based noise filtering (1)

```
Fs = 1000; % Sampling frequency
dt = 1/Fs; % Sampling interval
L = 1000; % Length of signal
t = (0:L-1)*dt; % Time vector
T=L*dt; % Total time interval

% Sum of a 50 Hz sinusoid and a 120 Hz sinusoid
x = 0.7*sin(2*pi*50*t) + sin(2*pi*120*t);
y = x + 2*randn(size(t)); % Sinusoids plus noise

figure(1); clf;
plot(t(1:100),y(1:100),'b--'); hold on
title('Signal Corrupted with Zero-Mean Random Noise')
xlabel('time')
```

FFT-based noise filtering (2)

```

if (0)
    N=(L/2)*2; % Even N
    y_hat = fft(y(1:N));
    % Frequencies ordered in a funny way:
    f_funny = 2*pi/T* [0:N/2-1, -N/2:-1];
    % Normal ordering:
    f_normal = 2*pi/T* [-N/2 : N/2-1];
else
    N=(L/2)*2-1; % Odd N
    y_hat = fft(y(1:N));
    % Frequencies ordered in a funny way:
    f_funny = 2*pi/T* [0:(N-1)/2, -(N-1)/2:-1];
    % Normal ordering:
    f_normal = 2*pi/T* [-(N-1)/2 : (N-1)/2];
end

```

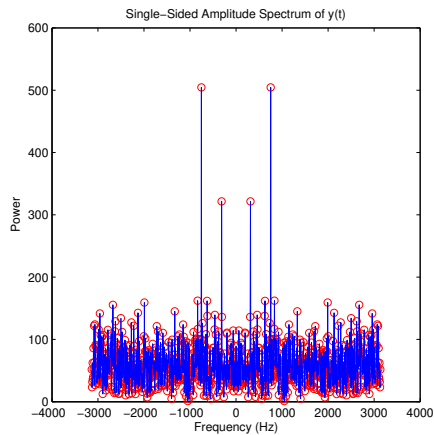
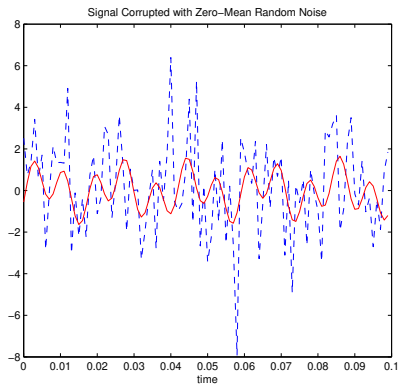
FFT-based noise filtering (3)

```
figure(2); clf; plot(f_funny, abs(y_hat), 'ro'); hold
y_hat=fftshift(y_hat);
figure(2); plot(f_normal, abs(y_hat), 'b-');

title('Single-Sided Amplitude Spectrum of y(t)')
xlabel('Frequency (Hz)')
ylabel('Power')

y_hat(abs(y_hat)<250)=0; % Filter out noise
y_filtered = ifft(ifftshift(y_hat));
figure(1); plot(t(1:100), y_filtered(1:100), 'r-')
```

FFT results



Multidimensional FFT

- DFTs and FFTs generalize straightforwardly to higher dimensions due to separability: **Transform each dimension independently**

$$\hat{f} = \frac{1}{N_x N_y} \sum_{j_y=0}^{N_y-1} \sum_{j_x=0}^{N_x-1} f_{j_x, j_y} \exp \left[-\frac{2\pi i (j_x k_x + j_y k_y)}{N} \right]$$

$$\hat{\mathbf{f}}_{k_x, k_y} = \frac{1}{N_x} \sum_{j_y=0}^{N_y-1} \exp \left(-\frac{2\pi i j_y k_x}{N} \right) \left[\frac{1}{N_y} \sum_{j_x=0}^{N_x-1} f_{j_x, j_y} \exp \left(-\frac{2\pi i j_x k_y}{N} \right) \right]$$

- For example, in two dimensions, **do FFTs of each column, then FFTs of each row of the result:**

$$\hat{\mathbf{f}} = \mathcal{F}_{\text{row}} (\mathcal{F}_{\text{col}} (\mathbf{f}))$$

- The cost is N_y one-dimensional FFTs of length N_x and then N_x one-dimensional FFTs of length N_y :

$$N_x N_y \log N_x + N_x N_y \log N_y = N_x N_y \log (N_x N_y) = N \log N$$

Conclusions

Conclusions/Summary

- **Periodic functions** can be approximated using basis of **orthogonal trigonometric polynomials**.
- The Fourier basis is **discretely orthogonal** and gives **spectral accuracy** for smooth functions.
- Functions with discontinuities are not approximated well: **Gibbs phenomenon**.
- The **Discrete Fourier Transform** can be computed very efficiently using the **Fast Fourier Transform** algorithm: $O(N \log N)$.