

Scientific Computing: An Introductory Survey

Chapter 12 – Fast Fourier Transform

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Outline

- 1 Discrete Fourier Transform
- 2 Fast Fourier Transform
- 3 Applications



Trigonometric Interpolation

- In modeling periodic or cyclic phenomena, sines and cosines are more appropriate functions than polynomials or piecewise polynomials
- Representation as linear combination of sines and cosines decomposes continuous function or discrete data into components of various frequencies
- Representation in *frequency space* may enable more efficient manipulations than in original time or space domain



Complex Exponential Notation

- We will use complex exponential notation based on *Euler's identity*

$$e^{i\theta} = \cos \theta + i \sin \theta$$

where $i = \sqrt{-1}$

- Since $e^{-i\theta} = \cos(-\theta) + i \sin(-\theta) = \cos \theta - i \sin \theta$, we have

$$\cos(2\pi kt) = \frac{e^{2\pi ikt} + e^{-2\pi ikt}}{2}$$

and

$$\sin(2\pi kt) = i \frac{e^{-2\pi ikt} - e^{2\pi ikt}}{2}$$

- Pure cosine or sine wave of frequency k is equivalent to sum or difference of complex exponentials of half amplitude and frequencies k and $-k$

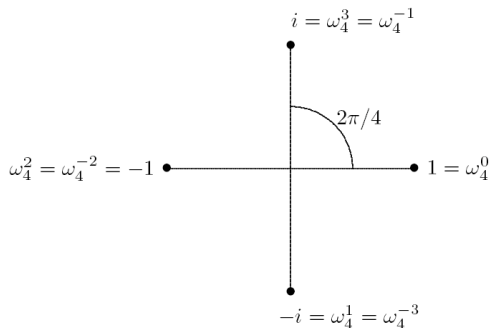


Roots of Unity

- For given integer n , *primitive n th root of unity* is given by

$$\omega_n = \cos(2\pi/n) - i \sin(2\pi/n) = e^{-2\pi i/n}$$

- n th roots of unity, called *twiddle factors* in this context, are given by ω_n^k or by ω_n^{-k} , $k = 0, \dots, n-1$



< interactive example >



Discrete Fourier Transform

- *Discrete Fourier transform*, or *DFT*, of sequence $\mathbf{x} = [x_0, \dots, x_{n-1}]^T$ is sequence $\mathbf{y} = [y_0, \dots, y_{n-1}]^T$ given by

$$y_m = \sum_{k=0}^{n-1} x_k \omega_n^{mk}, \quad m = 0, 1, \dots, n-1$$

- In matrix notation, $\mathbf{y} = \mathbf{F}_n \mathbf{x}$, where entries of *Fourier matrix* \mathbf{F}_n are given by

$$\{\mathbf{F}_n\}_{mk} = \omega_n^{mk}$$

- For example,

$$\mathbf{F}_4 = \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & \omega^1 & \omega^2 & \omega^3 \\ 1 & \omega^2 & \omega^4 & \omega^6 \\ 1 & \omega^3 & \omega^6 & \omega^9 \end{bmatrix} = \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & -i & -1 & i \\ 1 & -1 & 1 & -1 \\ 1 & i & -1 & -i \end{bmatrix}$$



Inverse DFT

- Note that

$$\frac{1}{n} \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & \omega^{-1} & \omega^{-2} & \omega^{-3} \\ 1 & \omega^{-2} & \omega^{-4} & \omega^{-6} \\ 1 & \omega^{-3} & \omega^{-6} & \omega^{-9} \end{bmatrix} \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & \omega^1 & \omega^2 & \omega^3 \\ 1 & \omega^2 & \omega^4 & \omega^6 \\ 1 & \omega^3 & \omega^6 & \omega^9 \end{bmatrix} = \begin{bmatrix} 1 & 0 & 0 & 0 \\ 0 & 1 & 0 & 0 \\ 0 & 0 & 1 & 0 \\ 0 & 0 & 0 & 1 \end{bmatrix}$$

- In general,

$$\mathbf{F}_n^{-1} = (1/n)\mathbf{F}_n^H$$

- Inverse DFT* is therefore given by

$$x_k = \frac{1}{n} \sum_{m=0}^{n-1} y_m \omega_n^{-mk} \quad k = 0, 1, \dots, n-1$$

- DFT gives trigonometric interpolant using only matrix-vector multiplication, which costs only $\mathcal{O}(n^2)$



DFT, continued

- DFT of sequence, even purely real sequence, is in general complex
- Components of DFT y of real sequence x of length n are *conjugate symmetric*: y_k and y_{n-k} are complex conjugates for $k = 1, \dots, (n/2) - 1$
- Two components of special interest are
 - y_0 , whose value is sum of components of x , is sometimes called *DC component*, corresponding to zero frequency (i.e., constant function)
 - $y_{n/2}$, corresponding to *Nyquist frequency*, which is highest frequency representable at given sampling rate
- Components of y beyond Nyquist frequency correspond to frequencies that are negatives of those below Nyquist frequency



Example: DFT

- For randomly chosen sequence x ,

$$\mathbf{F}_8 \mathbf{x} = \mathbf{F}_8 \begin{bmatrix} 4 \\ 0 \\ 3 \\ 6 \\ 2 \\ 9 \\ 6 \\ 5 \end{bmatrix} = \begin{bmatrix} 35 \\ -5.07 + 8.66i \\ -3 + 2i \\ 9.07 + 2.66i \\ -5 \\ 9.07 - 2.66i \\ -3 - 2i \\ -5.07 - 8.66i \end{bmatrix} = \mathbf{y}$$

- Transformed sequence is complex, but y_0 and y_4 are real, while y_5 , y_6 , and y_7 are complex conjugates of y_3 , y_2 , and y_1 , respectively
- There appears to be no discernible pattern to frequencies present, and y_0 is indeed equal to sum of components of x



Example: DFT

- For cyclic sequence x ,

$$\mathbf{F}_8 \mathbf{x} = \mathbf{F}_8 \begin{bmatrix} 1 \\ -1 \\ 1 \\ -1 \\ 1 \\ -1 \\ 1 \\ -1 \end{bmatrix} = \begin{bmatrix} 0 \\ 0 \\ 0 \\ 0 \\ 8 \\ 0 \\ 0 \\ 0 \end{bmatrix} = \mathbf{y}$$

- Sequence has highest possible rate of oscillation (between 1 and -1) for this sampling rate
- In transformed sequence, only nonzero component is at Nyquist frequency (in this case y_4)



Computing DFT

- By taking advantage of symmetries and redundancies in definition of DFT, shortcut algorithm can be developed for evaluating DFT very efficiently
- For illustration, consider case $n = 4$
- From definition of DFT

$$y_m = \sum_{k=0}^3 x_k \omega_n^{mk}, \quad m = 0, \dots, 3$$

- Writing out four equations in full

$$y_0 = x_0\omega_n^0 + x_1\omega_n^0 + x_2\omega_n^0 + x_3\omega_n^0$$

$$y_1 = x_0\omega_n^0 + x_1\omega_n^1 + x_2\omega_n^2 + x_3\omega_n^3$$

$$y_2 = x_0\omega_n^0 + x_1\omega_n^2 + x_2\omega_n^4 + x_3\omega_n^6$$

$$y_3 = x_0\omega_n^0 + x_1\omega_n^3 + x_2\omega_n^6 + x_3\omega_n^9$$



Computing DFT, continued

- Noting that

$$\omega_n^0 = \omega_n^4 = 1, \quad \omega_n^2 = \omega_n^6 = -1, \quad \omega_n^9 = \omega_n^1$$

and regrouping, we obtain four equations

$$y_0 = (x_0 + \omega_n^0 x_2) + \omega_n^0 (x_1 + \omega_n^0 x_3)$$

$$y_1 = (x_0 - \omega_n^0 x_2) + \omega_n^1 (x_1 - \omega_n^0 x_3)$$

$$y_2 = (x_0 + \omega_n^0 x_2) + \omega_n^2 (x_1 + \omega_n^0 x_3)$$

$$y_3 = (x_0 - \omega_n^0 x_2) + \omega_n^3 (x_1 - \omega_n^0 x_3)$$

- DFT can now be computed with only 8 additions or subtractions and 6 multiplications, instead of expected $(4 - 1) * 4 = 12$ additions and $4^2 = 16$ multiplications
- Actually, even fewer multiplications are required for this small case, since $\omega_n^0 = 1$, but we have tried to illustrate how algorithm works in general



Computing DFT, continued

- Main point is that computing DFT of original 4-point sequence has been reduced to computing DFT of its two 2-point even and odd subsequences
- This property holds in general: DFT of n -point sequence can be computed by breaking it into two DFTs of half length, provided n is even
- General pattern becomes clearer when viewed in terms of first few Fourier matrices

$$\mathbf{F}_1 = 1, \quad \mathbf{F}_2 = \begin{bmatrix} 1 & 1 \\ 1 & -1 \end{bmatrix}, \quad \mathbf{F}_4 = \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & -i & -1 & i \\ 1 & -1 & 1 & -1 \\ 1 & i & -1 & -i \end{bmatrix}, \dots$$



Computing DFT, continued

- Let P_4 be permutation matrix

$$P_4 = \begin{bmatrix} 1 & 0 & 0 & 0 \\ 0 & 0 & 1 & 0 \\ 0 & 1 & 0 & 0 \\ 0 & 0 & 0 & 1 \end{bmatrix}$$

and D_2 be diagonal matrix

$$D_2 = \text{diag}(1, \omega_4) = \begin{bmatrix} 1 & 0 \\ 0 & -i \end{bmatrix}$$

- Then

$$F_4 P_4 = \left[\begin{array}{cc|cc} 1 & 1 & 1 & 1 \\ 1 & -1 & -i & i \\ \hline 1 & 1 & -1 & -1 \\ 1 & -1 & i & -i \end{array} \right] = \begin{bmatrix} F_2 & D_2 F_2 \\ F_2 & -D_2 F_2 \end{bmatrix}$$



Computing DFT, continued

- Thus, F_4 can be rearranged so that each block is diagonally scaled version of F_2
- Such hierarchical splitting can be carried out at each level, provided number of points is even
- In general, P_n is permutation that groups even-numbered columns of F_n before odd-numbered columns, and

$$D_{n/2} = \text{diag} \left(1, \omega_n, \dots, \omega_n^{(n/2)-1} \right)$$

- To apply F_n to sequence of length n , we need merely apply $F_{n/2}$ to its even and odd subsequences and scale results, where necessary, by $\pm D_{n/2}$
- Resulting recursive divide-and-conquer algorithm for computing DFT is called *fast Fourier transform*, or *FFT*



FFT Algorithm

```
procedure  $\text{fft}(x, y, n, \omega)$   
  if  $n = 1$  then  
     $y[0] = x[0]$                                 { bottom of recursion }  
  else  
    for  $k = 0$  to  $(n/2) - 1$   
       $p[k] = x[2k]$                                 { split into even and  
       $s[k] = x[2k + 1]$                             odd subsequences }  
    end  
     $\text{fft}(p, q, n/2, \omega^2)$                        { call  $\text{fft}$  procedure  
     $\text{fft}(s, t, n/2, \omega^2)$                        recursively }  
    for  $k = 0$  to  $n - 1$   
       $y[k] = q[k \bmod (n/2)] +$                        { combine results }  
         $\omega^k t[k \bmod (n/2)]$   
    end  
end
```



FFT Algorithm, continued

- There are $\log_2 n$ levels of recursion, each of which involves $\mathcal{O}(n)$ arithmetic operations, so total cost is $\mathcal{O}(n \log_2 n)$
- For clarity, separate arrays were used for subsequences, but transform can be computed in place using no additional storage
- Input sequence is assumed complex; if input sequence is real, then additional symmetries in DFT can be exploited to reduce storage and operation count by half
- Output sequence is not produced in natural order, but either input or output sequence can be rearranged at cost of $\mathcal{O}(n \log_2 n)$, analogous to sorting



FFT Algorithm, continued

- FFT algorithm can be formulated using iteration rather than recursion, which is often desirable for greater efficiency or when using programming language that does not support recursion
- Despite its name, fast Fourier transform is an *algorithm*, not a transform
- It is particular way of computing DFT of sequence in efficient manner

< interactive example >



Complexity of FFT

- DFT is defined in terms of matrix-vector product, whose straightforward evaluation would appear to require $\mathcal{O}(n^2)$ arithmetic operations
- Use of FFT algorithm reduces work to only $\mathcal{O}(n \log_2 n)$, which makes enormous practical difference in time required to transform large sequences

n	$n \log_2 n$	n^2
64	384	4096
128	896	16384
256	2048	65536
512	4608	262144
1024	10240	1048576



Inverse Transform

- Due to similar form of DFT and its inverse (only sign of exponent differs), FFT algorithm can also be used to compute inverse DFT efficiently
- Ability to transform back and forth quickly between time and frequency domains makes it practical to perform computations or analysis that may be required in whichever domain is more convenient and efficient



Limitations of FFT

- FFT algorithm is not always applicable or maximally efficient
- Input sequence is assumed to be
 - Equally spaced
 - Periodic
 - Power of two in length
- First two of these follow from definition of DFT, while third is required for maximal efficiency of FFT algorithm
- Care must be taken in applying FFT algorithm to produce most meaningful results as efficiently as possible
- For example, transforming sequence that is not really periodic or padding sequence to make its length power of two may introduce spurious noise and complicate interpretation of results [< interactive example >](#)



Mixed-Radix FFT

- *Mixed-radix* FFT algorithm does not require number of points n to be power of two
- More general algorithm is still based on divide and conquer, with sequence being split at each level by smallest prime factor of length of remaining sequence
- Efficiency depends on whether n is product of small primes (ideally power of two)
- If not, then much of computational advantage of FFT may be lost
- For example, if n itself is prime, then sequence cannot be split at all, and “fast” algorithm becomes standard $\mathcal{O}(n^2)$ matrix-vector multiplication



Applications of DFT

- DFT is often of direct interest itself and is also useful as computational tool that provides efficient means for computing other quantities
- By its nature, DFT can be used to detect periodicities or cycles in discrete data, and to *remove* unwanted periodicities
- For example, to remove high-frequency noise, compute DFT of sequence, set high-frequency components of transformed sequence to zero, then compute inverse DFT of modified sequence to get back into original domain



Applications of DFT, continued

- As another example, weather data often contain two distinct cycles, diurnal (daily) and annual (yearly), and one might want to remove one to study the other in isolation
- Economic data are also often “seasonally adjusted” by removing unwanted periodicities to reveal “secular” trends
- Because of such uses, DFT is of vital importance in many aspects of signal processing, such as digital filtering

< interactive example >



Applications of DFT, continued

- Some computations are simpler or more efficient in frequency domain than in time domain
- Examples include discrete convolution of two sequences u and v of length n

$$\{u \star v\}_m = \sum_{k=0}^{n-1} v_k u_{m-k}, \quad m = 0, 1, \dots, n-1$$

and related quantities such as cross-correlation of two sequences or autocorrelation of a sequence with itself

- Equivalent operation in frequency domain is simply pointwise multiplication (plus complex conjugation in some cases)



Applications of DFT, continued

- If DFT and its inverse can be computed efficiently, then it may be advantageous to transform to frequency domain to compute such convolutions, then transform back to time domain
- For example, to compute convolution or correlation of two sequences, it is often advantageous to use FFT algorithm to compute DFT of both sequences, compute pointwise product in frequency domain, then inverse DFT back to time domain, again via FFT algorithm
- FFT algorithm also forms basis for exceptionally efficient methods for solving certain periodic boundary value problems, such as Poisson's equation on regular domain with periodic boundary conditions



Fast Polynomial Multiplication

- FFT algorithm also provides fast methods for some computations that might not seem related to it
- For example, complexity of straightforward multiplication of two polynomials is proportional to product of their degrees
- However, polynomial of degree $n - 1$ is uniquely determined by its values at n distinct points
- Thus, product polynomial can be determined by interpolation from pointwise product of factor polynomials evaluated at n points
- Both polynomial evaluation and interpolation using n points would normally require $\mathcal{O}(n^2)$ operations, but by choosing points to be n th roots of unity, FFT algorithm can be used to reduce complexity to $\mathcal{O}(n \log_2 n)$



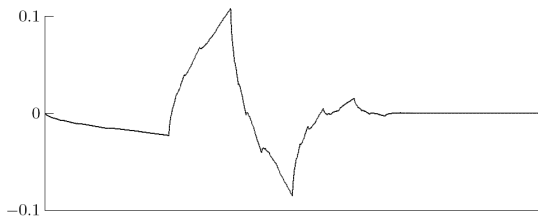
Wavelets

- Sine and cosine functions used in Fourier analysis are very smooth (infinitely differentiable), and very broad (nonzero almost everywhere on real line)
- They are not very effective for representing functions that change abruptly or have highly localized support
- Gibbs phenomenon in Fourier representation of square wave (“ringing” at corners) is one manifestation of this
- In response to this shortcoming, there has been intense interest in recent years in new type of basis functions called *wavelets*



Wavelets, continued

- Wavelet basis is generated from single function $\phi(x)$, called *mother wavelet* or *scaling function*, by dilation and translation, $\phi((x - b)/a)$, where $a, b \in \mathbb{R}$ with $a \neq 0$
- There are many choices for mother wavelet, with choice trading off smoothness vs compactness
- Commonly used family of wavelets is due to Daubechies, example of which is shown below



Wavelets, continued

- Typical choices for dilation and translation parameters are $a = 2^{-j}$ and $b = k2^j$, where j and k are integers, so that $\phi_{jk}(x) = \phi(2^j x - k)$

- If mother wavelet $\phi(x)$ has sufficiently localized support, then

$$\int \phi_{jk} \phi_{mn} = 0$$

whenever indices do not both match, so doubly-indexed basis functions $\phi_{jk}(x)$ are orthogonal

- By replicating mother wavelet at many different scales, it is possible to mimic behavior of any function at many different scales; this property of wavelets is called *multiresolution*



Wavelets, continued

- Fourier basis functions are localized in frequency but not in time: small changes in frequency produce changes everywhere in time domain
- Wavelets are localized in both frequency (by dilation) and time (by translation)
- This localization tends to make wavelet representation of function very sparse



Discrete Wavelet Transform

- As with Fourier transform, there is analogous discrete wavelet transform, or DWT
- DWT and its inverse can be computed very efficiently by pyramidal, hierarchical algorithm
- Sparsity of wavelet basis makes computation of DWT even faster than FFT
- DWT requires only $\mathcal{O}(n)$ work for sequence of length n , instead of $\mathcal{O}(n \log n)$
- Because of their efficiency, both in computation and in compactness of representation, wavelets are playing an increasingly important role in many areas of signal and image processing, such as data compression, noise removal, and computer vision

