

■ 1.4 Fast Fourier Transform (FFT) Algorithm

Fast Fourier Transform, or FFT, is any algorithm for computing the N -point DFT with a computational complexity of $\mathcal{O}(N \log N)$. It is *not* a new transform, but simply an efficient method of calculating the DFT of $x(n)$.

If we assume that N is even, we can write the N -point DFT of $x(n)$ as

$$\begin{aligned}
 X^{(N)}(k) &= \sum_{\substack{n \text{ is even: } n=2m, m=0, \dots, \frac{N}{2}-1}} x(n) e^{-j \frac{2\pi k}{N} n} \\
 &\quad + \sum_{\substack{n \text{ is odd: } n=2l+1, l=0, \dots, \frac{N}{2}-1}} x(n) e^{-j \frac{2\pi k}{N} n} \\
 &= \sum_{m=0}^{\frac{N}{2}-1} x(2m) e^{-j \frac{2\pi k}{N} 2m} + \sum_{l=0}^{\frac{N}{2}-1} x(2l+1) e^{-j \frac{2\pi k}{N} (2l+1)} \quad (1.31)
 \end{aligned}$$

We make the following substitutions:

$$\begin{aligned}
 x_0(m) &= x(2m), \text{ where } m = 0, \dots, \frac{N}{2} - 1, \\
 x_1(l) &= x(2l+1), \text{ where } l = 0, \dots, \frac{N}{2} - 1.
 \end{aligned}$$

Rewriting Eq. (1.31), we get

$$\begin{aligned}
 X^{(N)}(k) &= \sum_{m=0}^{\frac{N}{2}-1} x_0(m) e^{-j \frac{2\pi k}{N} m} + e^{-j \frac{2\pi k}{N}} \sum_{l=0}^{\frac{N}{2}-1} x_1(l) e^{-j \frac{2\pi k}{N} l} \\
 &= X_0^{(\frac{N}{2})}(k) + e^{-j \frac{2\pi k}{N}} X_1^{(\frac{N}{2})}(k), \quad (1.32)
 \end{aligned}$$

where $X_0^{(\frac{N}{2})}(k)$ is the $\frac{N}{2}$ -point DFT of the even-numbered samples of $x(n)$ and $X_1^{(\frac{N}{2})}(k)$ is the $\frac{N}{2}$ -point DFT of the odd-numbered samples of $x(n)$. Note that both of them are $\frac{N}{2}$ -periodic discrete-time functions.

We have the following algorithm to compute $X^{(N)}(k)$ for $k = 0, \dots, (N-1)$:

1. Compute $X_0^{(\frac{N}{2})}(k)$ for $k = 0, \dots, \frac{N}{2} - 1$.
2. Compute $X_1^{(\frac{N}{2})}(k)$ for $k = 0, \dots, \frac{N}{2} - 1$.
3. Perform the computation (1.32) with N complex multiplications and N complex additions.

Actually, it is possible to use fewer than N complex multiplications. Let

$$W_N = e^{-j \frac{2\pi}{N}}.$$

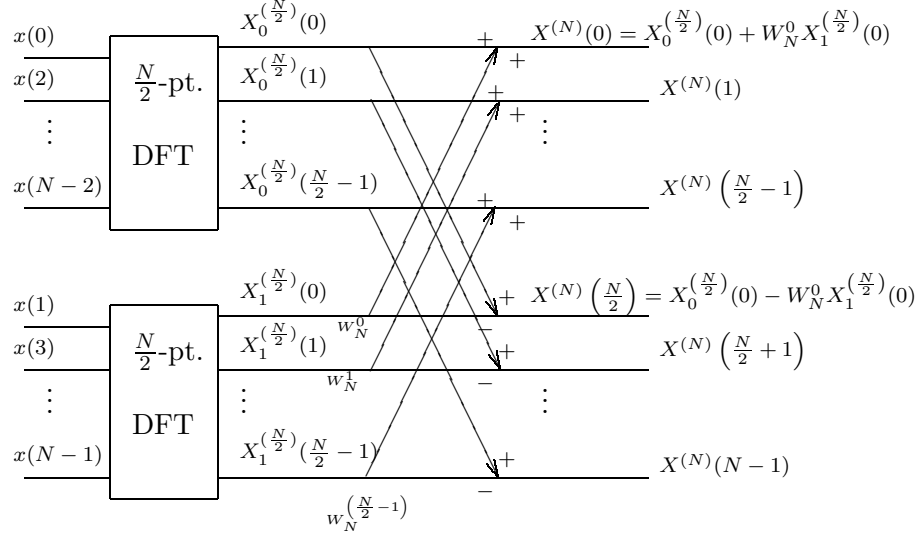


Figure 1.36. The FFT algorithm.

Then

$$\begin{aligned} W_N^{k+\frac{N}{2}} &= e^{-j(\frac{2\pi k}{N} + \pi)} \\ &= -e^{-j\frac{2\pi k}{N}} \\ &= -W_N^k \end{aligned}$$

Therefore,

$$\begin{aligned} X^{(N)}(k) &= X_0^{(\frac{N}{2})}(k) + W_N^k X_1(k) \quad \text{for } k = 0, \dots, \frac{N}{2} - 1, \\ X^{(N)}\left(k + \frac{N}{2}\right) &= X_0^{(\frac{N}{2})}(k) - W_N^k X_1(k) \quad \text{for } k = 0, \dots, \frac{N}{2} - 1, \end{aligned}$$

as illustrated in Fig. 1.36. This shows that we do not need to actually perform N complex multiplications, but only $\frac{N}{2}$.⁸

Fig. 1.37 illustrates the recursive implementation of the FFT supposing that $N = 2^M$. There is a total of $M = \log_2 N$ stages of computation, each requiring $\frac{3}{2}N$ complex operations. Hence, the total computational complexity is $\mathcal{O}(N \log N)$. We see that the process ends at a 1-point DFT. A 1-point DFT is the sample of the original signal:

$$X(0) = \sum_{n=0}^0 x(n) e^{-j(\frac{2\pi \cdot 0}{1})n} = x(0).$$

The following remarks apply to the FFT:

⁸Actually, slightly fewer if we do not count multiplications by ± 1 and $\pm j$.

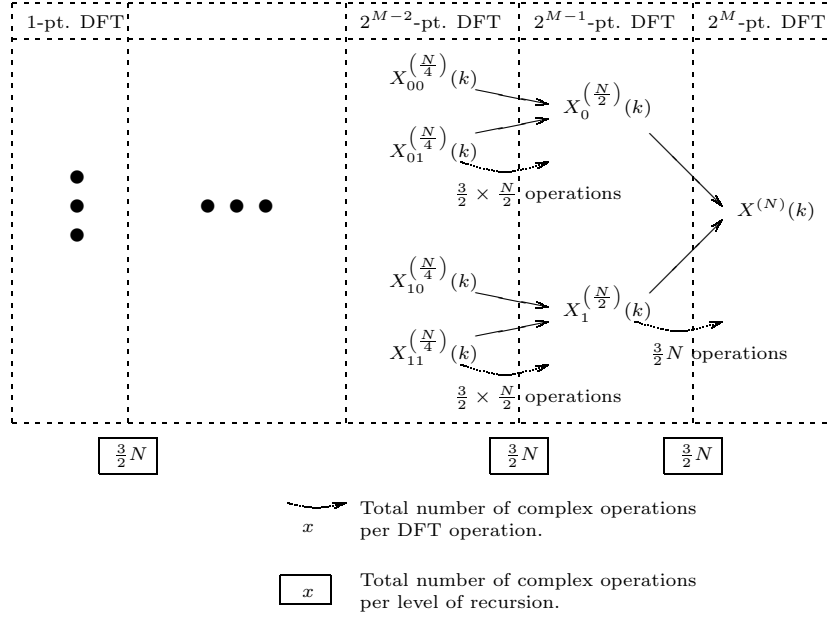


Figure 1.37. The recursive implementation of the FFT supposing that $N = 2^M$. There is a total of $M = \log_2 N$ stages of computation, each requiring $\frac{3}{2}N$ complex operations. Hence, the total computational complexity is $\mathcal{O}(N \log N)$.

1. For large N , the FFT is much faster than the direct application of the definition of DFT, which is of complexity $\mathcal{O}(N^2)$.
2. The particular implementation of the FFT described above is called *decimation-in-time radix-2 FFT*.
3. The number of operations required by an FFT algorithm can be approximated as $CN \log N$, where C is a constant. There are many variations of FFT aimed at reducing this constant—e.g., if $N = 3^M$, it may be better to use a radix-3 FFT.
4. Note that

$$\begin{aligned} \left\{ \frac{1}{N} \text{DFT}[x^*(n)] \right\}^* &= \left\{ \frac{1}{N} \sum_{n=0}^{N-1} x^*(n) e^{-j \left(\frac{2\pi k}{N} \right) n} \right\}^* \\ &= \frac{1}{N} \sum_{n=0}^{N-1} x(n) e^{j \left(\frac{2\pi k}{N} \right) n} \end{aligned}$$

which is the IDFT of $x(n)$. Thus, the FFT can also be used to compute the IDFT.

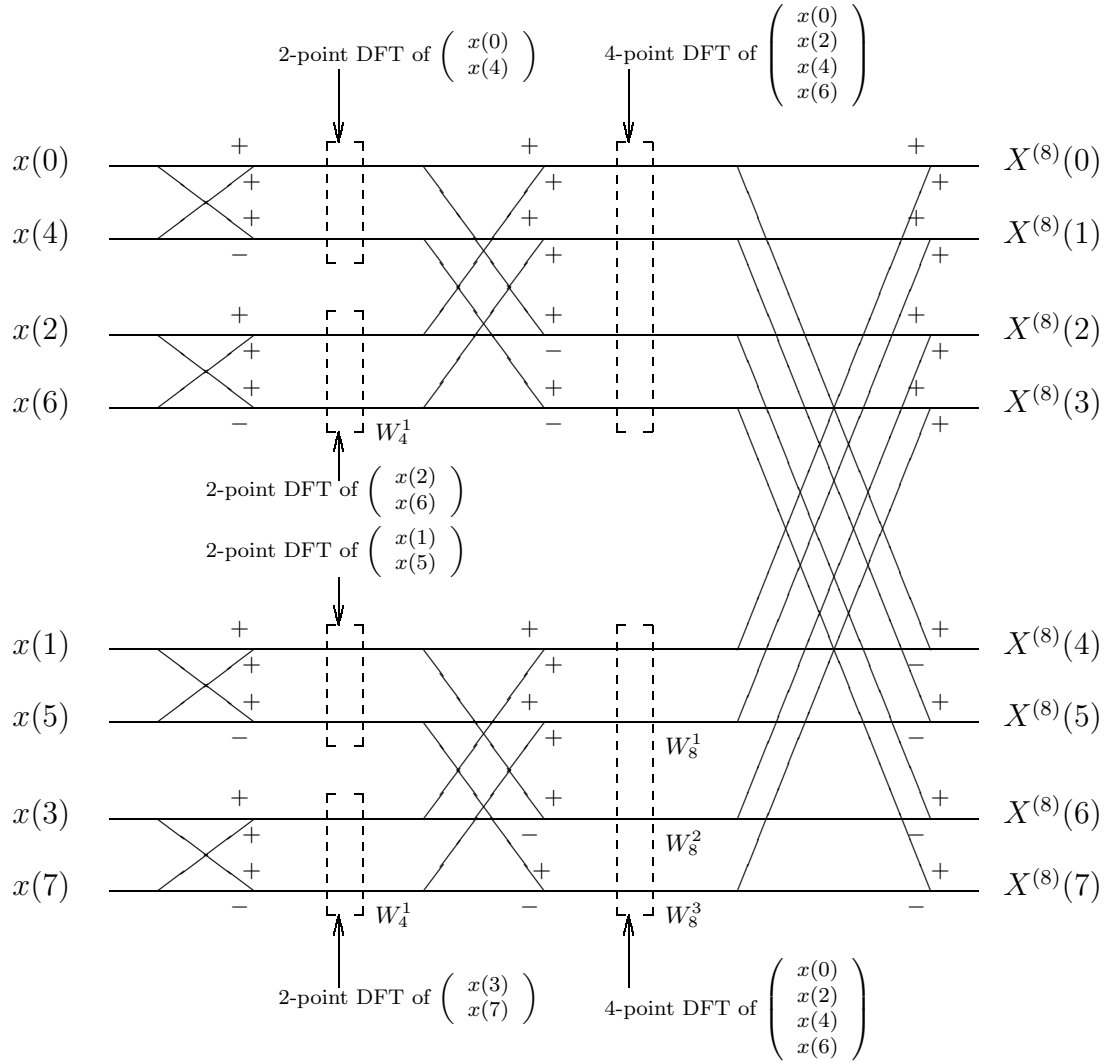


Figure 1.38. The 8-point FFT.

Example 1.26. The 8-point FFT is depicted in Fig. 1.38. The values of the twiddle factors are:

$$\begin{aligned} W_2 &= e^{-j\frac{2\pi}{2}} = -1, \\ W_4 &= e^{-j\frac{2\pi}{4}} = -j, \\ W_8 &= e^{-j\frac{2\pi}{8}}. \quad \blacksquare \end{aligned}$$

$$\begin{aligned}
& \begin{array}{c} \boxed{\mathbf{X}} \\ N \times 1 \end{array} = \begin{array}{c} \boxed{A^{(N)}} \\ N \times N \end{array} \begin{array}{c} \boxed{\mathbf{x}} \\ N \times 1 \end{array} \\
& = \begin{array}{c} \boxed{\begin{array}{cc|cc} 1 & & w_N^0 & \\ 1 & 0 & w_N^1 & 0 \\ & \ddots & & \ddots \\ 0 & & 1 & 0 \\ 0 & & & w_N^{\frac{N}{2}-1} \end{array}} & \boxed{\begin{array}{c|c} A^{(\frac{N}{2})} & 0 \\ \hline 0 & A^{(\frac{N}{2})} \end{array}} & \boxed{\begin{array}{cccc|c} 1 & 0 & 0 & 0 & 0 \\ 0 & 0 & 1 & 0 & 0 \\ 0 & 0 & 0 & 0 & 1 \\ \vdots & \vdots & \vdots & \vdots & \vdots \\ 0 & 0 & 0 & 0 & 0 \\ \vdots & \vdots & \vdots & \vdots & \vdots \\ 0 & 1 & 0 & 0 & 0 \\ 0 & 0 & 0 & 1 & 0 \\ \vdots & \vdots & \vdots & \vdots & \vdots \\ 0 & 0 & 0 & 0 & 0 \\ \vdots & \vdots & \vdots & \vdots & \vdots \\ 0 & 0 & 0 & 0 & 0 \end{array}} & \boxed{\mathbf{x}} \\
& = \begin{array}{c} \boxed{\begin{array}{cc|cc} 1 & & w_N^0 & \\ 1 & 0 & w_N^1 & 0 \\ & \ddots & & \ddots \\ 0 & & 1 & 0 \\ 0 & & & w_N^{\frac{N}{2}-1} \end{array}} & \boxed{\begin{array}{c|c} A^{(\frac{N}{2})} & 0 \\ \hline 0 & A^{(\frac{N}{2})} \end{array}} & \boxed{\begin{array}{c} x(0) \\ x(2) \\ \vdots \\ x(N-2) \\ x(1) \\ x(3) \\ \vdots \\ x(N-1) \end{array}} \\
& = \begin{array}{c} \boxed{\begin{array}{cc|cc} 1 & & w_N^0 & \\ 1 & 0 & w_N^1 & 0 \\ & \ddots & & \ddots \\ 0 & & 1 & 0 \\ 0 & & & w_N^{\frac{N}{2}-1} \end{array}} & \boxed{\begin{array}{c} X_0^{(\frac{N}{2})}(0) \\ \vdots \\ X_0^{(\frac{N}{2})}(\frac{N}{2}-1) \\ X_1^{(\frac{N}{2})}(0) \\ \vdots \\ X_1^{(\frac{N}{2})}(\frac{N}{2}-1) \end{array}}
\end{aligned}$$

Figure 1.39. The FFT reduces the number of operations required to calculate the DFT by reducing $A^{(N)}$ to two $A^{(\frac{N}{2})}$ that is only half the size of $A^{(N)}$. This operation is repeated with every recursion until we reach the 1-point DFT.

Recall that the DFT is a matrix multiplication (Fig. 1.35). One stage of the FFT essentially reduces the multiplication by an $N \times N$ matrix to two multiplications by $\frac{N}{2} \times \frac{N}{2}$ matrices. This reduces the number of operations required to calculate the DFT by almost a factor of two (Fig. 1.39).

Another interpretation of FFT involves analyzing the matrix

$$A_{k,L} = \frac{1}{\sqrt{2}} \begin{pmatrix} 1 & e^{-j\frac{2\pi k}{2^L}} \\ 1 & -e^{-j\frac{2\pi k}{2^L}} \end{pmatrix},$$

where k and L are nonnegative integers such that $k < 2^L$. Note that

$$\begin{aligned} \langle A_{k,L}\mathbf{x}, A_{k,L}\mathbf{y} \rangle &= (A_{k,L}\mathbf{y})^H (A_{k,L}\mathbf{x}) \\ &= \mathbf{y}^H A_{k,L}^H A_{k,L} \mathbf{x} \\ &= \mathbf{y}^H \frac{1}{\sqrt{2}} \begin{pmatrix} 1 & 1 \\ e^{j\frac{2\pi k}{N}} & -e^{j\frac{2\pi k}{N}} \end{pmatrix} \frac{1}{\sqrt{2}} \begin{pmatrix} 1 & e^{-j\frac{2\pi k}{N}} \\ 1 & -e^{-j\frac{2\pi k}{N}} \end{pmatrix} \mathbf{x} \\ &= \mathbf{y}^H \frac{1}{2} \begin{pmatrix} 2 & 0 \\ 0 & 2 \end{pmatrix} \mathbf{x} = \mathbf{y}^H \mathbf{x} = \langle \mathbf{x}, \mathbf{y} \rangle, \end{aligned}$$

i.e., multiplication by $A_{k,L}$ preserves distances and angles — roughly speaking, it is a rotation or reflection. Continuing the matrix decomposition of Fig. 1.39 further until we get the full FFT, it can be shown that FFT consists of $\frac{N}{2} \log N$ multiplications by 2×2 matrices of the form $\sqrt{2}A_{k,L}$, each operating on a pair of coordinates.⁹ Therefore, FFT breaks down the multiplication by the DFT matrix A into elementary planar transformations.

■ 1.4.1 Fast Computation of Convolution

Consider a linear system described by

$$\mathbf{y} = S\mathbf{x}, \tag{1.33}$$

where \mathbf{x} is the $N \times 1$ input vector, representing an N -periodic input signal; S is an $N \times N$ matrix; and \mathbf{y} is the $N \times 1$ output vector, representing an N -periodic output signal. What conditions must the matrix S satisfy in order for the system to be time-invariant, i.e., invariant to circular shifts of the input vector?

Note that a circular shift by one sample is

$$\begin{pmatrix} x(0) \\ x(1) \\ x(2) \\ \vdots \\ x(N-1) \end{pmatrix} \rightarrow \begin{pmatrix} x(-1) = x(N-1) \\ x(0) \\ x(1) \\ \vdots \\ x(N-2) \end{pmatrix}.$$

⁹The same conclusion can be reached by examining an FFT diagram such as Fig. 1.38.

Let the first column of S be

$$\mathbf{h} = \begin{pmatrix} h(0) \\ h(1) \\ h(2) \\ \vdots \\ h(N-1) \end{pmatrix}.$$

Note that when

$$\mathbf{x} = \begin{pmatrix} 1 \\ 0 \\ 0 \\ \vdots \\ 0 \end{pmatrix}, \text{ then } \mathbf{y} = \mathbf{h},$$

and when

$$\mathbf{x} = \begin{pmatrix} 0 \\ 1 \\ 0 \\ \vdots \\ 0 \end{pmatrix},$$

then \mathbf{y} is the second column of S , which therefore, in order for S to be invariant to circular shifts, must be equal to:

$$\begin{pmatrix} h(N-1) \\ h(0) \\ h(1) \\ \vdots \\ h(N-2) \end{pmatrix}.$$

Similarly, when

$$\mathbf{x} = \begin{pmatrix} 0 \\ 0 \\ 1 \\ 0 \\ \vdots \\ 0 \end{pmatrix}, \text{ then } \mathbf{y} \text{ is the third column of } S, \text{ etc.}$$

Thus, the matrix S must have the following structure:

$$S = \begin{pmatrix} h(0) & h(N-1) & h(N-2) & \cdots & h(1) \\ h(1) & h(0) & h(N-1) & \cdots & h(2) \\ h(2) & h(1) & h(0) & \cdots & h(3) \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ h(N-1) & h(N-2) & h(N-3) & \cdots & h(0) \end{pmatrix}.$$

This is called a *circulant matrix*. We can then write Eq. (1.33) as

$$\begin{aligned} y(n) &= \sum_{m=0}^{N-1} x(m)h(n-m) \\ &= \sum_{m=0}^{N-1} x(m)h((n-m) \bmod N) \end{aligned} \quad (1.34)$$

$$= x \circledast h(n) = x \circledast h \quad (1.35)$$

Eq. (1.35) is called a *circular convolution* or a *periodic convolution*. Note that formula (1.34) works even when x or h are non-periodic. Observe the following:

- For $y(0)$, the sum of the indices of x and h is always 0 mod N for every term.

$$y(0) = x(0)h(0) + x(1)h(N-1) + x(2)h(N-2) + \cdots + x(N-1)h(1)$$

- For $y(1)$, the sum of the indices of x and h is always 1 mod N for every term.

$$y(1) = x(0)h(1) + x(1)h(0) + x(2)h(N-1) + \cdots + x(N-1)h(2)$$

This is true for all $y(k)$, $k = 0, 1, \dots, N-1$.

What are the eigenvectors of S ? Let us try

$$\mathbf{g}_k = \begin{pmatrix} \frac{1}{N} \\ \frac{1}{N}e^{j\frac{2\pi k}{N} \cdot 1} \\ \frac{1}{N}e^{j\frac{2\pi k}{N} \cdot 2} \\ \vdots \\ \frac{1}{N}e^{j\frac{2\pi k}{N} \cdot (N-1)} \end{pmatrix}, \text{ where } k = 0, 1, \dots, N-1.$$

We have:

$$\begin{aligned} y(n) &= h(n) \circledast \mathbf{g}_k \\ &= \sum_{m=0}^{N-1} h(m)g_k(n-m) \\ &= \sum_{m=0}^{N-1} h(m) \frac{1}{N} e^{j\frac{2\pi k}{N}(n-m)} \\ &= \left\{ \sum_{m=0}^{N-1} h(m) e^{-j\frac{2\pi k}{N}m} \right\} \frac{1}{N} e^{j\frac{2\pi k}{N}n} \\ &= \underbrace{H(k)}_{\text{DFT of } h} \frac{1}{N} e^{j\frac{2\pi k}{N}n} \end{aligned}$$

Hence we have that

$$S\mathbf{g}_k = H(k)\mathbf{g}_k$$

where \mathbf{g}_k is the k -th eigenvector and $H(k)$ gives the corresponding eigenvalue. Therefore,

$$S \left(\underbrace{\mathbf{g}_0 \ \mathbf{g}_1 \ \cdots \ \mathbf{g}_{N-1}}_{\text{The IDFT matrix } B} \right) = \left(\mathbf{g}_0 \ \mathbf{g}_1 \ \cdots \ \mathbf{g}_{N-1} \right) \begin{pmatrix} H(0) & & & 0 \\ & H(1) & & \\ & & \ddots & \\ 0 & & & H(N-1) \end{pmatrix}.$$

Then S can be written as:

$$S = B \begin{pmatrix} H(0) & & & 0 \\ & H(1) & & \\ & & \ddots & \\ 0 & & & H(N-1) \end{pmatrix} A,$$

where the DFT matrix A is:

$$A = NB^H = \begin{pmatrix} \mathbf{g}_0^H \\ \mathbf{g}_1^H \\ \vdots \\ \mathbf{g}_{N-1}^H \end{pmatrix}.$$

Complex exponentials are the eigenvectors of circulant matrices. They *diagonalize* circulant matrices. Thus, for any $\mathbf{x} \in \mathbb{C}^N$,

$$S\mathbf{x} = B \begin{pmatrix} H(0) & & & 0 \\ & H(1) & & \\ & & \ddots & \\ 0 & & & H(N-1) \end{pmatrix} A\mathbf{x}.$$

Let us compare two algorithms for computing the circular convolution of \mathbf{x} and \mathbf{h} .

Algorithm 1 Directly perform the multiplication $S\mathbf{x}$. This has computational complexity $\mathcal{O}(N^2)$.

Algorithm 2 1. Represent \mathbf{x} in the eigenbasis of S , i.e., the Fourier basis,

$$\mathbf{X} = A\mathbf{x}.$$

This step can be done with FFT whose complexity is $\mathcal{O}(N \log N)$.

Step 1	Step 2	Step 3
$x(n) \xrightarrow{\text{N-point DFT}} X(k)$ $h(n) \xrightarrow{\text{N-point DFT}} H(k)$	$Y(k) = X(k)H(k)$	$Y(k) \xrightarrow{\text{N-point IDFT}} y(n) = x \circledast h(n)$

Figure 1.40. An illustration of the FFT implementation of the circular convolution.

2. Compute the representation of \mathbf{y} in the eigenbasis of S :

$$\mathbf{Y} = \begin{pmatrix} H(0) & & 0 \\ & H(1) & \\ 0 & & \ddots \\ & & & H(N-1) \end{pmatrix} \mathbf{X}.$$

This computation has complexity $\mathcal{O}(N)$.

3. Reconstruct \mathbf{y} from its Fourier coefficients:

$$\mathbf{y} = B\mathbf{Y}.$$

This has complexity $\mathcal{O}(N \log N)$, if done using the FFT.

This algorithm is summarized in Fig. 1.40. Its total complexity is $\mathcal{O}(N \log N)$.

(Note that the second algorithm does not necessarily perform better for *any* matrix.)

Example 1.27. *This example explores the relationship between the convolution and the circular convolution. Let x and h be N -periodic signals, and let*

$$x_z = \begin{cases} x(n), & 0 \leq n \leq N-1 \\ 0, & \text{otherwise} \end{cases}$$

$$h_z = \begin{cases} h(n), & 0 \leq n \leq N-1 \\ 0, & \text{otherwise} \end{cases}$$

If we let

$$y_z(n) = x_z * h_z(n)$$

$$y(n) = x \circledast h(n)$$

then $y(n)$ can be expressed as

$$y(n) = \begin{cases} y_z(n) + y_z(N+n), & n = 0, 1, \dots, N-2 \\ y(N-1), & n = N-1 \end{cases}$$

Note that the overlap of $y_z(n)$ and $y_z(N + n)$ causes temporal aliasing in the resulting $y(n)$. This is the main difference between convolution and circular convolution.

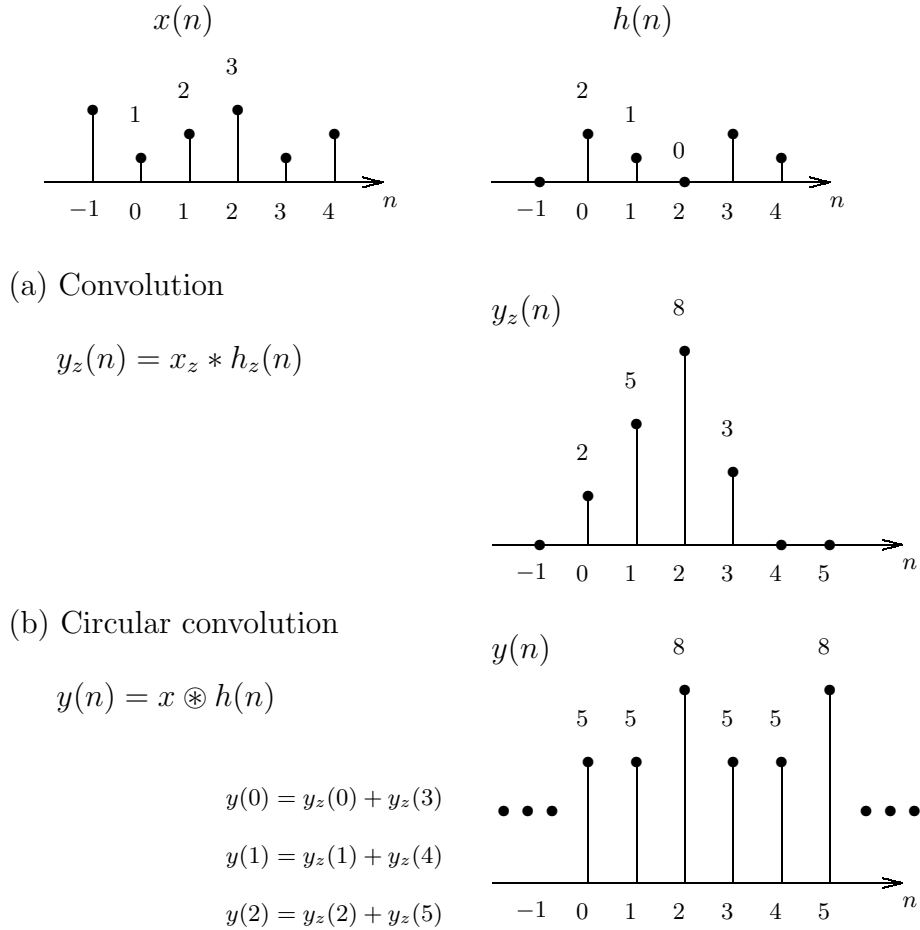


Figure 1.41. A comparison between circular convolution and convolution.

Fig. 1.41 illustrates the effect of temporal aliasing. To remove or minimize the effect of temporal aliasing, we could zero-pad x and h so that the temporal replicas are spread further apart, and thus, overlapping would not occur.