

Lecture 23: Introduction to Filter Banks

Foundations of Digital Signal Processing

Outline

- Review of Filter Design
- Short Time Fourier Transform
- Inefficient Filter Banks
- DFT Filter Bank

■ Homework #9

- Due Today
- Submit via canvas

■ Coding Assignment #6

- Due on Tuesday, Nov. 20th
- Submit via canvas

Lecture 23: Introduction to Filter Banks

Foundations of Digital Signal Processing

Outline

- **Review of Filter Design**
- Short Time Fourier Transform
- Inefficient Filter Banks
- DFT Filter Bank

Filter Design

■ Problem

- Consider the ideal filter (assume ω is periodic with period 2π)

$$|H(\omega)| = \begin{cases} 1 & \text{for } |\omega| \leq \frac{\pi}{3} \\ 0 & \text{for otherwise} \end{cases} \quad w[n] = \frac{1}{2} \left(1 - \cos \left(\frac{2\pi n}{N-1} \right) \right)$$

- Compute a discrete-time FIR filter with a 12-point Hann window defined by $w[n]$.

Filter Design

■ Problem

- Consider the ideal filter (assume ω is periodic with period 2π)

$$|H(\omega)| = \begin{cases} 1 & \text{for } |\omega| \leq \frac{\pi}{3} \\ 0 & \text{for otherwise} \end{cases} \quad w[n] = \frac{1}{2} \left(1 - \cos \left(\frac{2\pi n}{N-1} \right) \right)$$

- Compute a discrete-time FIR filter with a 12-point Hann window defined by $w[n]$. **Need to add shift for causality.**

$$|H(\omega)| = \sum_{k=-\infty}^{\infty} u \left(\omega + \frac{\pi}{3} - 2\pi k \right) - u \left(\omega - \frac{\pi}{3} - 2\pi k \right)$$
$$h[n] = \frac{\sin \left(\left(\frac{\pi}{3} \right) \left(n - \frac{N-1}{2} \right) \right)}{\pi \left(n - \frac{N-1}{2} \right)}$$

Filter Design

■ Problem

- Consider the ideal filter (assume ω is periodic with period 2π)

$$|H(\omega)| = \begin{cases} 1 & \text{for } |\omega| \leq \frac{\pi}{3} \\ 0 & \text{for otherwise} \end{cases} \quad w[n] = \frac{1}{2} \left(1 - \cos \left(\frac{2\pi n}{N-1} \right) \right)$$

- Compute a discrete-time FIR filter with a 12-point Hann window defined by $w[n]$.

$$h[n]w[n] = \begin{cases} \frac{\sin \left(\left(\frac{\pi}{3} \right) (n - 5.5) \right)}{2\pi(n - 5.5)} \left(1 - \cos \left(\frac{2\pi n}{N-1} \right) \right) & \text{for } 0 \leq n \leq 11 \\ 0 & \text{for otherwise} \end{cases}$$

Filter Design

■ Problem

- Consider the ideal filter (assume ω is periodic with period 2π)

$$H(\omega) = \begin{cases} 1 & \text{for } |\omega| \leq \frac{\pi}{3} \\ 0 & \text{for otherwise} \end{cases}$$

- Determine a discrete-time FIR filter with a 12-point frequency sampling method

Filter Design

■ Problem

- Consider the ideal filter (assume ω is periodic with period 2π)

$$H(\omega) = \begin{cases} 1 & \text{for } |\omega| \leq \frac{\pi}{3} \\ 0 & \text{for otherwise} \end{cases}$$

- Determine a discrete-time FIR filter with a 12-point frequency sampling method

$$H[k] = [1 \ 1 \ 1 \ 0 \ 0 \ 0 \ 0]$$

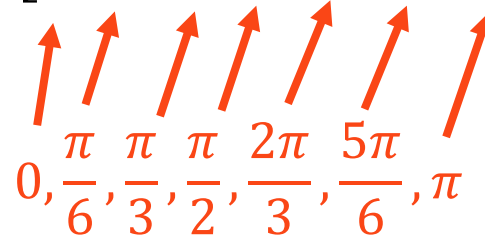


Diagram illustrating the frequency sampling method. Red arrows point from the frequency samples $0, \frac{\pi}{6}, \frac{\pi}{3}, \frac{\pi}{2}, \frac{2\pi}{3}, \frac{5\pi}{6}, \pi$ to the corresponding coefficients in the vector $H[k]$.

Filter Design

■ Problem

- Consider the ideal filter (assume ω is periodic with period 2π)

$$H(\omega) = \begin{cases} 1 & \text{for } |\omega| \leq \frac{\pi}{3} \\ 0 & \text{for otherwise} \end{cases}$$

- Determine a discrete-time FIR filter with a 12-point frequency sampling method

$$\text{Even } N: h[n] = H[0] + 2 \sum_{k=1}^{N/2} H[k] \cos\left(\frac{2\pi}{N} \left(n - \frac{N-1}{2}\right) k\right)$$

$$\text{Odd } N: h[n] = H[0] + 2 \sum_{k=1}^{(N-1)/2} H[k] \cos\left(\frac{2\pi}{N} \left(n - \frac{N-1}{2}\right) k\right)$$

Filter Design

■ Problem

- Consider the ideal filter (assume ω is periodic with period 2π)

$$H(\omega) = \begin{cases} 1 & \text{for } |\omega| \leq \frac{\pi}{3} \\ 0 & \text{for otherwise} \end{cases}$$

- Determine a discrete-time FIR filter with a 12-point frequency sampling method

$$h[n] = H[0] + 2 \sum_{k=1}^{N/2} H[k] \cos\left(\frac{2\pi}{12}(n - 5.5)k\right)$$

Filter Design

■ Problem

- Consider the ideal filter (assume ω is periodic with period 2π)

$$H(\omega) = \begin{cases} 1 & \text{for } |\omega| \leq \frac{\pi}{3} \\ 0 & \text{for otherwise} \end{cases}$$

- Determine a discrete-time FIR filter with a 12-point frequency sampling method

$$h[n] = H[0] + 2 \sum_{k=1}^{N/2} H[k] \cos\left(\frac{2\pi}{12}(n - 5.5)k\right)$$

$$h[n] = 1 + 2 \cos\left(\frac{2\pi}{12}(n - 5.5)\right) + 2 \cos\left(\frac{2\pi}{6}(n - 5.5)\right)$$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter by using the discrete approximation of the derivative.

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an $N=2$ order Butterworth filter by using the discrete approximation of the derivative.

$$s \rightarrow \frac{1}{T}(1 - z^{-1})$$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter by using the discrete approximation of the derivative and a sampling rate of T=1.

$$s \rightarrow \frac{1}{T}(1 - z^{-1})$$

$$|H(z)|^2 = \left| \frac{1}{1 - \frac{1}{T^4}(1 - z^{-1})^4} \right| = \left| \frac{1}{1 - (1 - z^{-1})^4} \right|$$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter by using the discrete approximation of the derivative and a sampling rate of T=1.

$$s \rightarrow \frac{1}{T}(1 - z^{-1})$$

$$|H(z)|^2 = \left| \frac{1}{1 - [z^{-4} - 4z^{-3} + 6z^{-2} - 4z^{-1} + 1]} \right|$$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter by using the discrete approximation of the derivative and a sampling rate of T=1.

$$s \rightarrow \frac{1}{T}(1 - z^{-1})$$

$$|H(z)|^2 = \left| \frac{1}{z^{-1}(2 - z^{-1})(1 - j - z^{-1})(1 + j - z^{-1})} \right|$$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter by using the discrete approximation of the derivative and a sampling rate of T=1.

Poles: $\infty, \frac{1}{2}, \frac{1}{2} - \frac{j}{2}, \frac{1}{2} + \frac{j}{2}$ **Zeros:** 0,0,0,0

$$|H(z)|^2 = \left| \frac{1}{z^{-1}(2 - z^{-1})(1 - j - z^{-1})(1 + j - z^{-1})} \right|$$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter with impulse invariance and T=1.

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an $N=2$ order Butterworth filter with impulse invariance and $T=1$.

Definition:

$$H(z) = \sum_{k=1}^K \frac{c_k}{1 - e^{p_k T} z^{-1}}$$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter with impulse invariance and T=1.

$$|H(s)|^2 = \frac{1}{1 - s^{2N}} = \frac{1}{1 - s^4}$$

Poles:

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter with impulse invariance and T=1.

$$|H(s)|^2 = \frac{1}{1 - s^{2N}} = \frac{1}{1 - s^4}$$

Poles: $s = e^{-j\frac{2\pi}{4}(0)}, e^{-j\frac{2\pi}{4}(1)}, e^{-j\frac{2\pi}{4}(2)}, e^{-j\frac{2\pi}{4}(3)}$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter with impulse invariance and T=1.

$$|H(s)|^2 = \frac{1}{1 - s^{2N}} = \frac{1}{1 - s^4}$$

Poles: $s = 1, -j, -1, j$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter with impulse invariance and T=1.

$$|H(s)|^2 = \frac{1}{(s-1)(s+1)(s^2+1)} = \frac{c_1}{(s-1)} + \frac{c_2}{(s+1)} + \frac{b}{(s^2+1)}$$

$$1 = c_1(s+1)(s^2+1) + c_2(s-1)(s^2+1) + b(s-1)(s+1)$$

$$s = 1$$

$$1 = 4c_1, \quad c_1 = 1/4$$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter with impulse invariance and T=1.

$$|H(s)|^2 = \frac{1}{(s-1)(s+1)(s^2+1)} = \frac{c_1}{(s-1)} + \frac{c_2}{(s+1)} + \frac{b}{(s^2+1)}$$

$$1 = c_1(s+1)(s^2+1) + c_2(s-1)(s^2+1) + b(s-1)(s+1)$$

$$s = -1$$

$$1 = -4c_2, \quad c_2 = -1/4$$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter with impulse invariance and T=1.

$$|H(s)|^2 = \frac{c_1}{(s-1)} + \frac{c_2}{(s+1)} + \frac{c_3}{(s+j)} + \frac{c_4}{(s-j)}$$

$$1 = c_1(s+1)(s^2+1) + c_2(s-1)(s^2+1) + c_3(s-1)(s+1)(s-j) + c_4(s-1)(s+1)(s+j)$$

$$s = -j$$

$$1 = (-j-1)(-j+1)(-2j)c_3 = 4jc_3, \quad c_3 = -j/4$$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter with impulse invariance and T=1.

$$|H(s)|^2 = \frac{c_1}{(s-1)} + \frac{c_2}{(s+1)} + \frac{c_3}{(s+j)} + \frac{c_4}{(s-j)}$$

$$1 = c_1(s+1)(s^2+1) + c_2(s-1)(s^2+1) + c_3(s-1)(s+1)(s-j) + c_4(s-1)(s+1)(s+j)$$

$$s = j$$

$$1 = (j-1)(j+1)(2j)c_4 = -4jc_4, \quad c_4 = j/4$$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter with impulse invariance and T=1.

$$|H(s)|^2 = \frac{1/4}{(s-1)} - \frac{1/4}{(s+1)} - \frac{j/4}{(s+j)} + \frac{j/4}{(s-j)}$$

Poles: $s = 1, -j, -1, j$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an $N=2$ order Butterworth filter with impulse invariance and $T=1$.

Poles: $s = 1, -j, -1, j$

$$H(z) = \sum_{k=1}^K \frac{c_k}{1 - e^{p_k T} z^{-1}} = \frac{1/4}{1 - e^1 z^{-1}} - \frac{1/4}{1 - e^{-1} z^{-1}} - \frac{j/4}{1 - e^{-j} z^{-1}} + \frac{j/4}{1 - e^{+j} z^{-1}}$$

Poles: $z = e, e^{-1}, e^{-j}, e^j$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter with the bilinear transform and T=1.

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter w/ the bilinear transform and T=1.

$$S \rightarrow \frac{2}{T} \frac{1 - z^{-1}}{1 + z^{-1}}$$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter w/ the bilinear transform and T=1.

$$S \rightarrow \frac{2}{T} \frac{1 - z^{-1}}{1 + z^{-1}}$$

$$|H(z)|^2 = \frac{1}{1 - s^4} = \frac{1}{1 - \left(2 \frac{1 - z^{-1}}{1 + z^{-1}}\right)^4} = \frac{(1 + z^{-1})^4}{(1 + z^{-1})^4 - 16(1 - z^{-1})^4}$$

Filter Design

■ Problem

- The squared magnitude of a Butterworth filter is defined by

$$|H(s)|^2 = \frac{1}{1 - s^{2N}}$$

- Determine the squared magnitude for a discrete-time IIR filter of an N=2 order Butterworth filter w/ the bilinear transform and T=1.

$$S \rightarrow \frac{2}{T} \frac{1 - z^{-1}}{1 + z^{-1}}$$

$$|H(z)|^2 = \frac{(1 + z^{-1})^4}{(3z^{-1} - 1)(3 - z^{-1})(5z^{-2} - 6z^{-1} + 5)}$$

Poles: $z = \frac{1}{3}, 3, \frac{3}{5} - \frac{4}{5}j, \frac{3}{5} + \frac{4}{5}j$

Lecture 23: Introduction to Filter Banks

Foundations of Digital Signal Processing


Outline

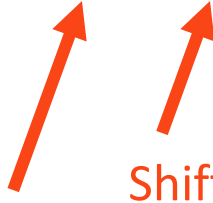
- Review of Filter Design
- **Short Time Fourier Transform**
- Inefficient Filter Banks
- DFT Filter Bank

The Short-Time Fourier Transform

■ The Definition:

$$X[Mm, k] = \text{DFT}_{n \rightarrow k}(w[n]x[n - Mm])$$


Time Frequency


Window Shifted Data

The Short-Time Fourier Transform

■ The Definition:

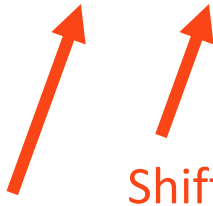
$$X[Mm, k] = \text{DFT}_{n \rightarrow k}(w[n]x[n - Mm])$$

Time



Frequency

Window



Shifted Data

$$X[m, k] = \sum_{n=0}^{N-1} w[n]x[n - Mm] e^{-j\frac{2\pi}{N}nk}$$

The Short-Time Fourier Transform

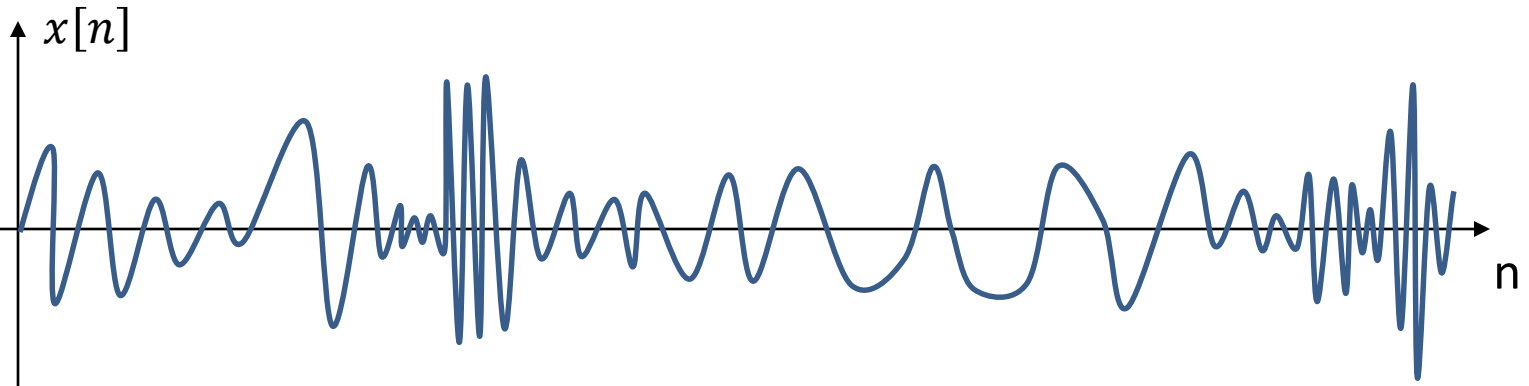
■ The Definition:

$$X[Mm, k] = \text{DFT}_{n \rightarrow k}(w[n]x[n - Mm])$$

Time
Frequency

Window
Shifted Data

$$X[m, k] = \sum_{n=0}^{N-1} w[n]x[n - Mm] e^{-j\frac{2\pi}{N}nk}$$



The Short-Time Fourier Transform

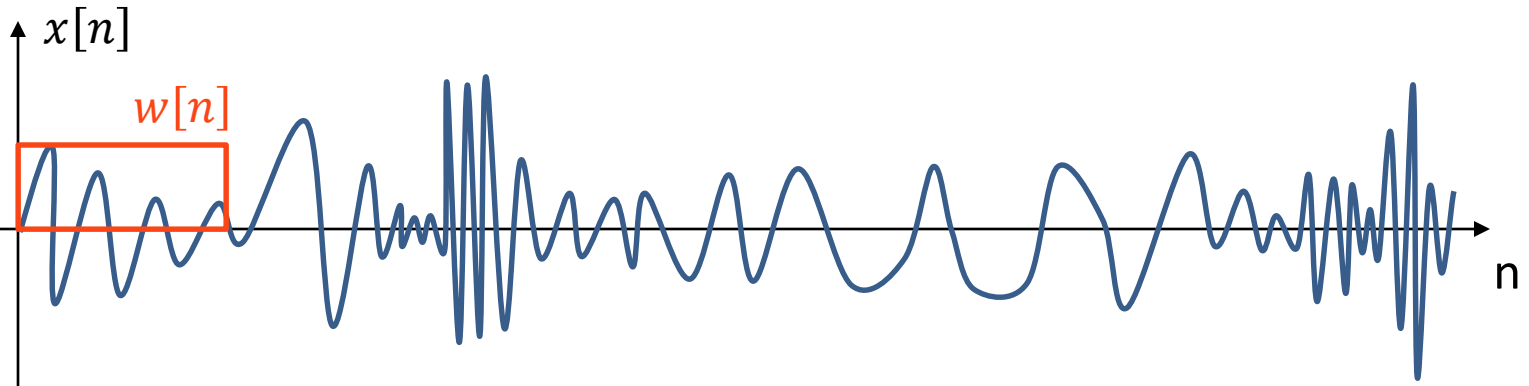
■ The Definition:

$$X[Mm, k] = \text{DFT}_{n \rightarrow k}(w[n]x[n - Mm])$$

Time
Frequency

Window
Shifted Data

$$X[m, k] = \sum_{n=0}^{N-1} w[n]x[n - Mm] e^{-j\frac{2\pi}{N}nk}$$



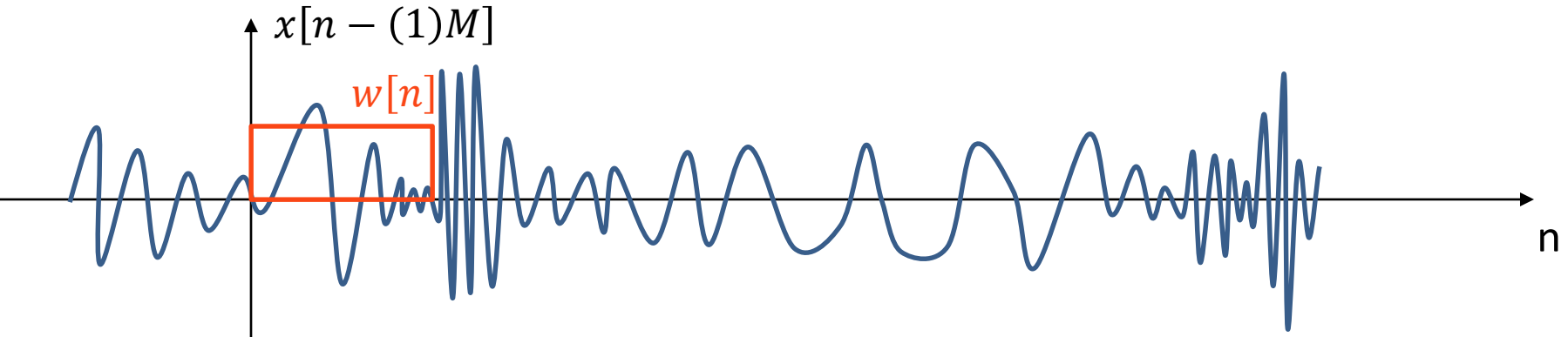
The Short-Time Fourier Transform

■ The Definition:

$$X[Mm, k] = \text{DFT}_{n \rightarrow k}(w[n]x[n - Mm])$$

Frequency
Time
Window
Shifted Data

$$X[m, k] = \sum_{n=0}^{N-1} w[n]x[n - Mm] e^{-j\frac{2\pi}{N}nk}$$



The Short-Time Fourier Transform

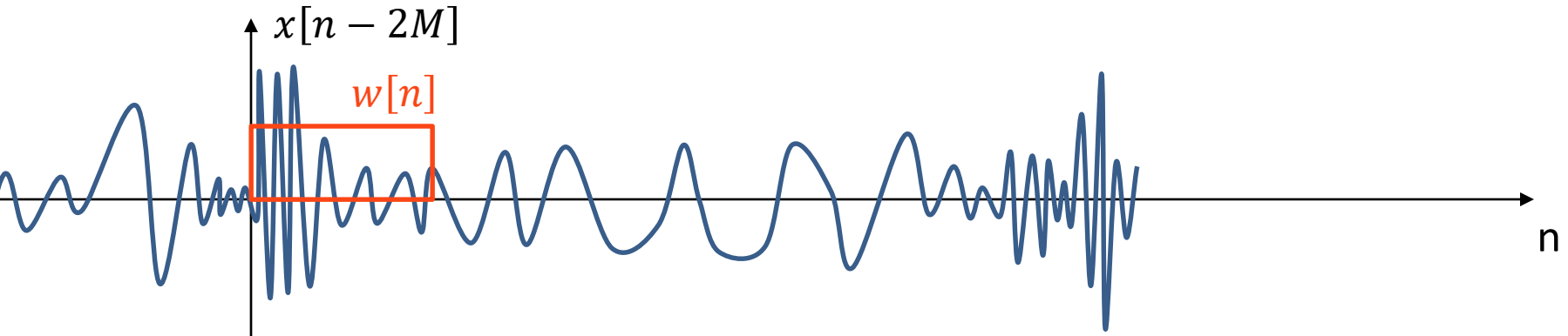
■ The Definition:

$$X[Mm, k] = \text{DFT}_{n \rightarrow k}(w[n]x[n - Mm])$$

Time
Frequency

Window
Shifted Data

$$X[m, k] = \sum_{n=0}^{N-1} w[n]x[n - Mm] e^{-j\frac{2\pi}{N}nk}$$



The Short-Time Fourier Transform

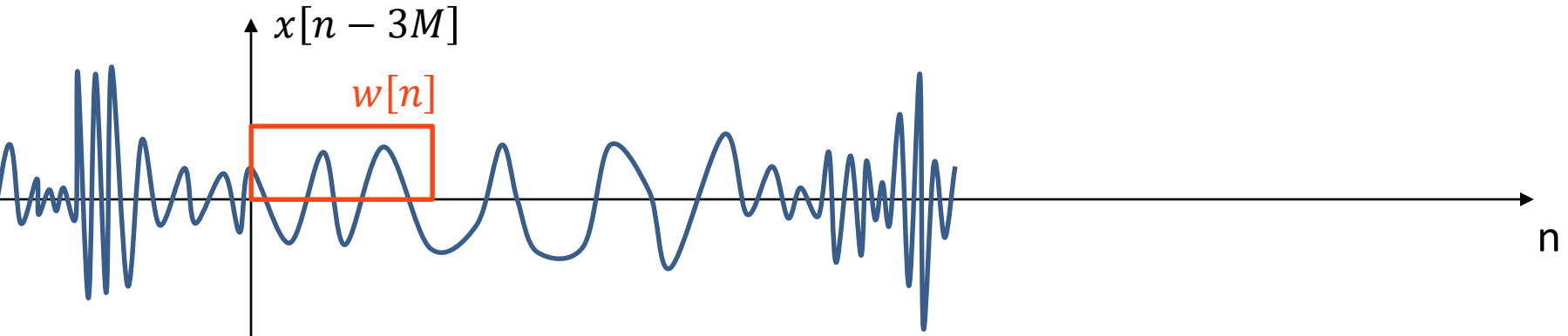
■ The Definition:

$$X[Mm, k] = \text{DFT}_{n \rightarrow k}(w[n]x[n - Mm])$$

Time
Frequency

Window
Shifted Data

$$X[m, k] = \sum_{n=0}^{N-1} w[n]x[n - Mm] e^{-j\frac{2\pi}{N}nk}$$



The Short-Time Fourier Transform

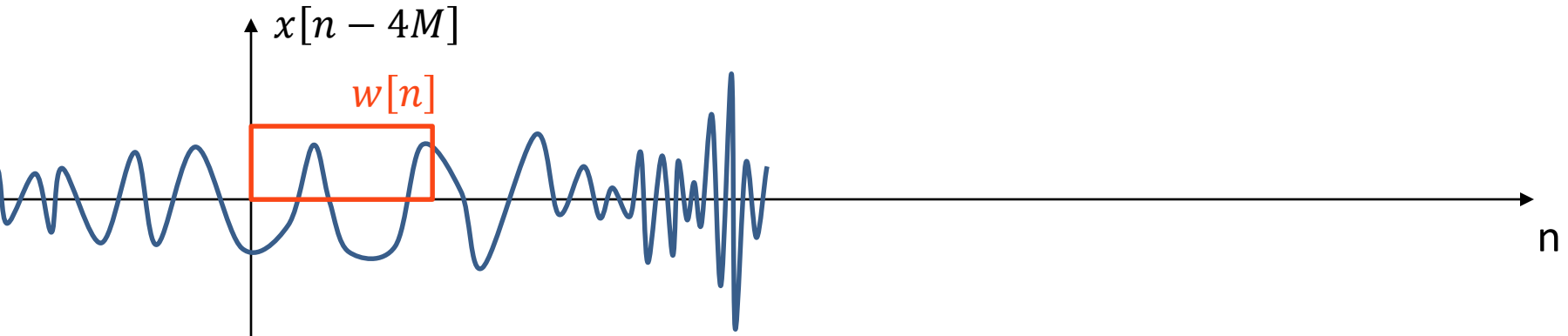
■ The Definition:

$$X[Mm, k] = \text{DFT}_{n \rightarrow k}(w[n]x[n - Mm])$$

Time
Frequency

Window
Shifted Data

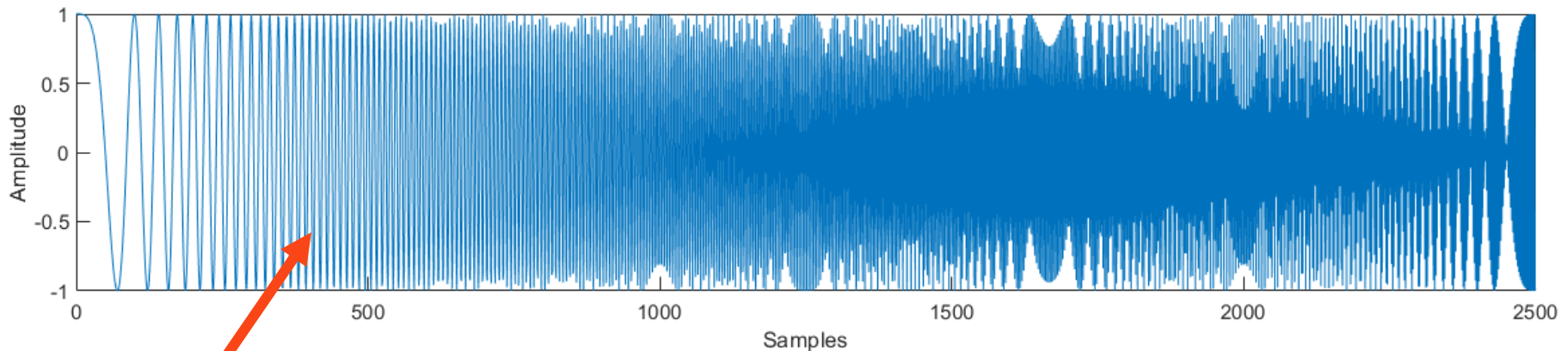
$$X[m, k] = \sum_{n=0}^{N-1} w[n]x[n - Mm] e^{-j\frac{2\pi}{N}nk}$$



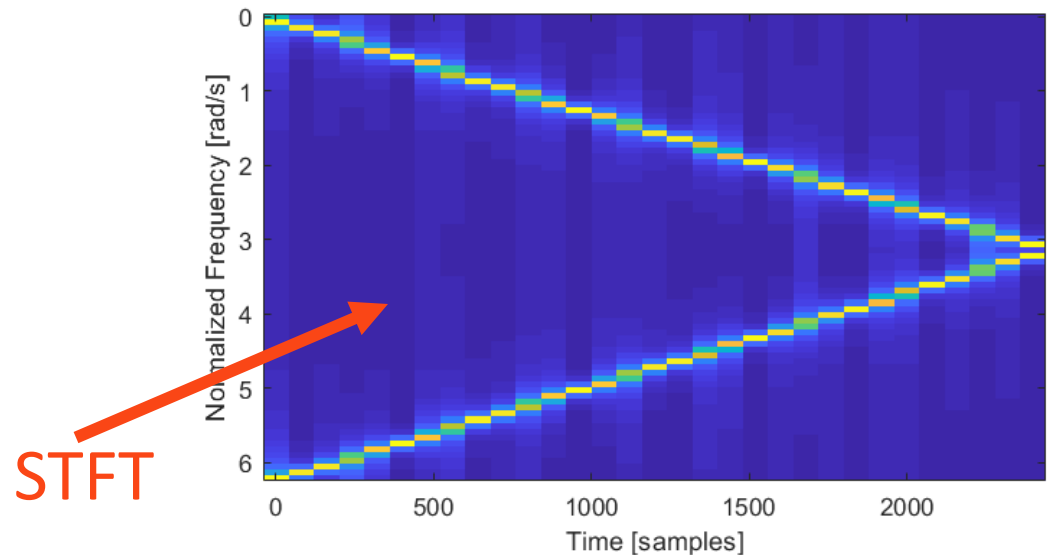
The Short-Time Fourier Transform

■ The Definition:

$$X[Mm, k] = \text{DFT}_{n \rightarrow k}(w[n]x[n - Mm])$$



Chirp

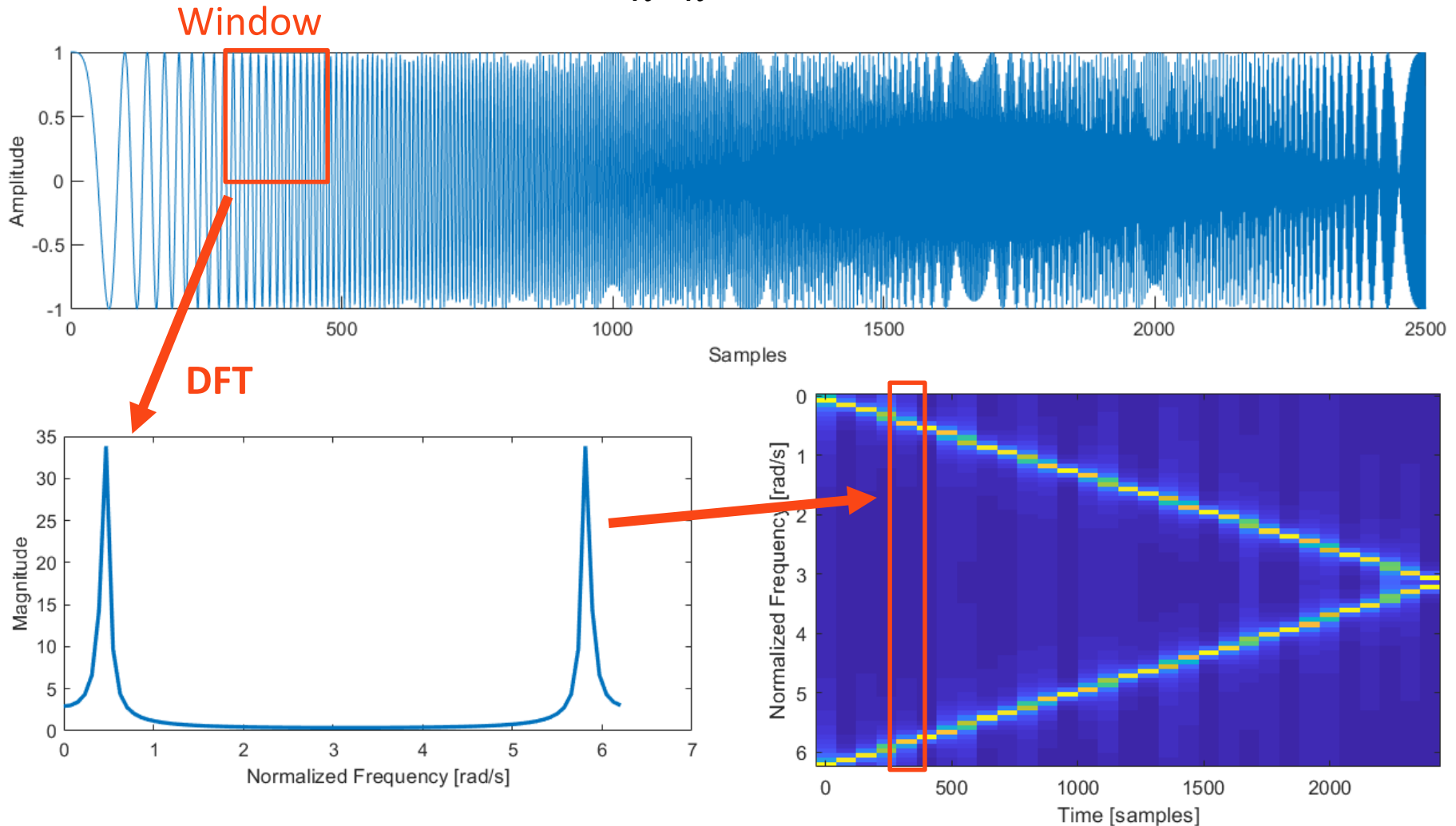


STFT

The Short-Time Fourier Transform

■ The Definition:

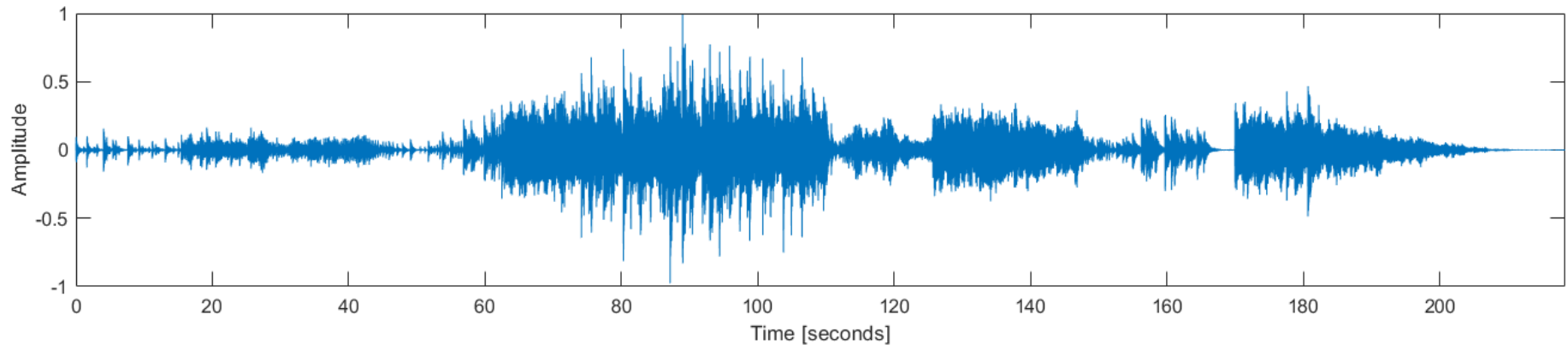
$$X[Mm, k] = \text{DFT}_{n \rightarrow k}(w[n]x[n - Mm])$$



The Short-Time Fourier Transform

■ The Definition:

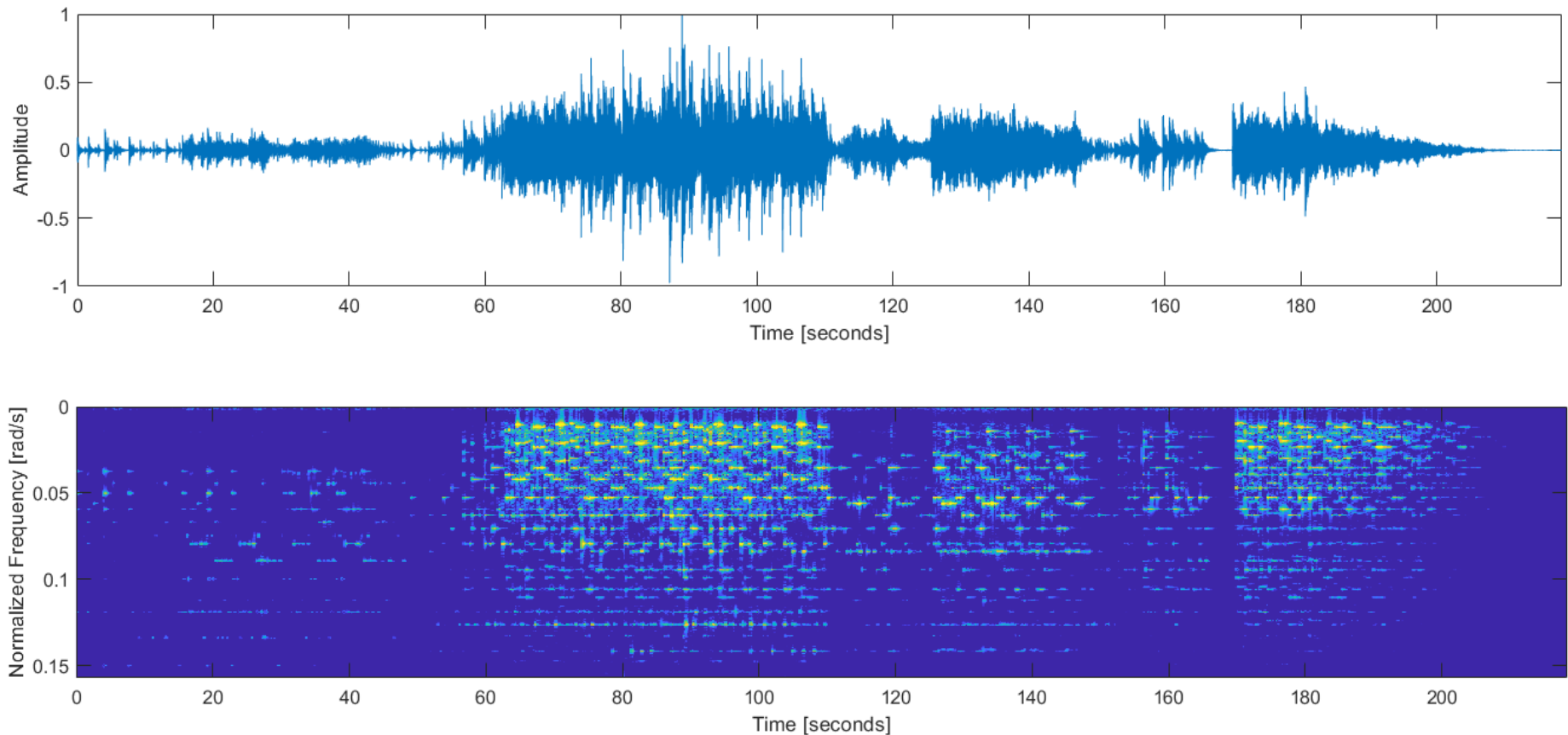
$$X[Mm, k] = \text{DFT}_{n \rightarrow k}(w[n]x[n - Mm])$$



The Short-Time Fourier Transform

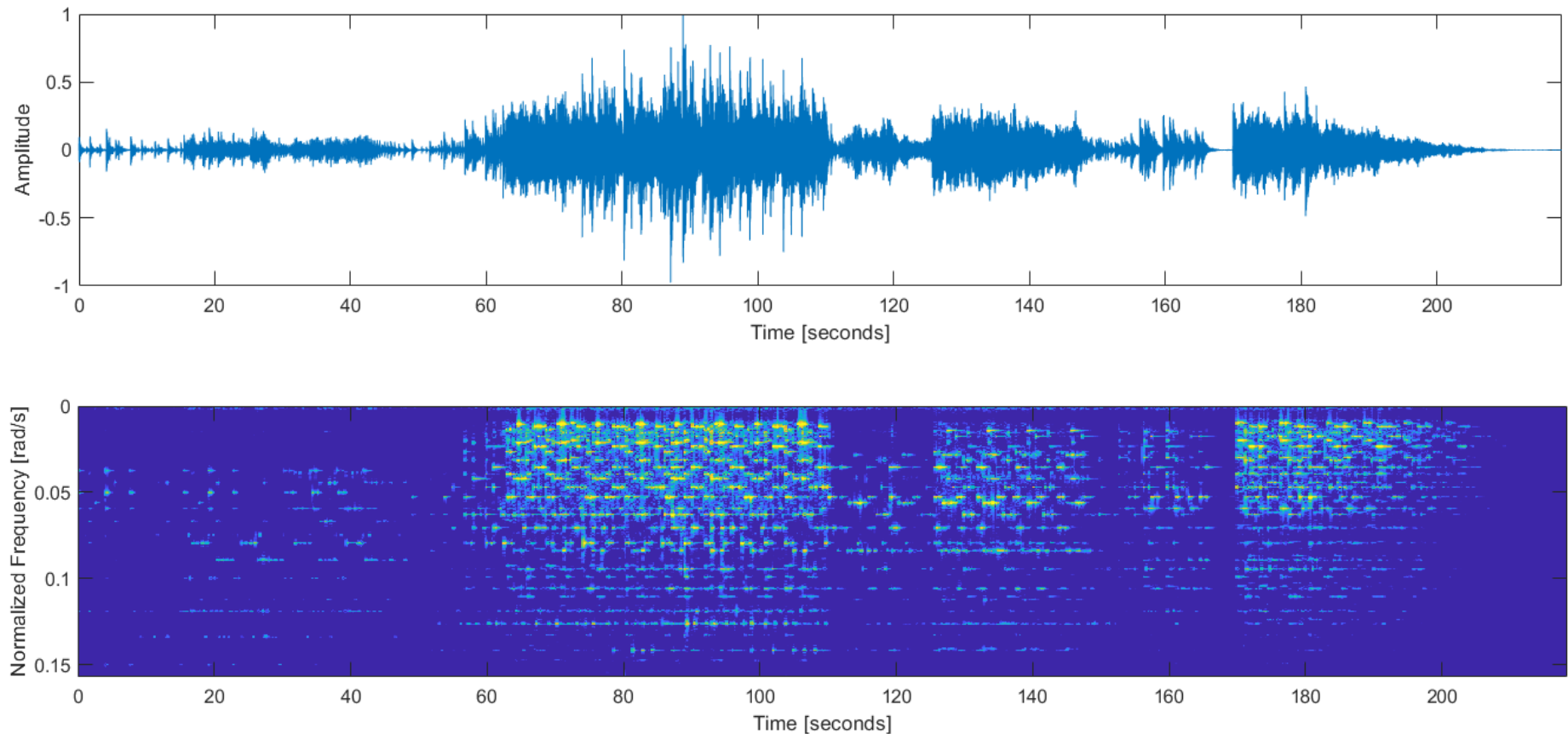
■ The Definition:

$$X[Mm, k] = \text{DFT}_{n \rightarrow k}(w[n]x[n - Mm])$$



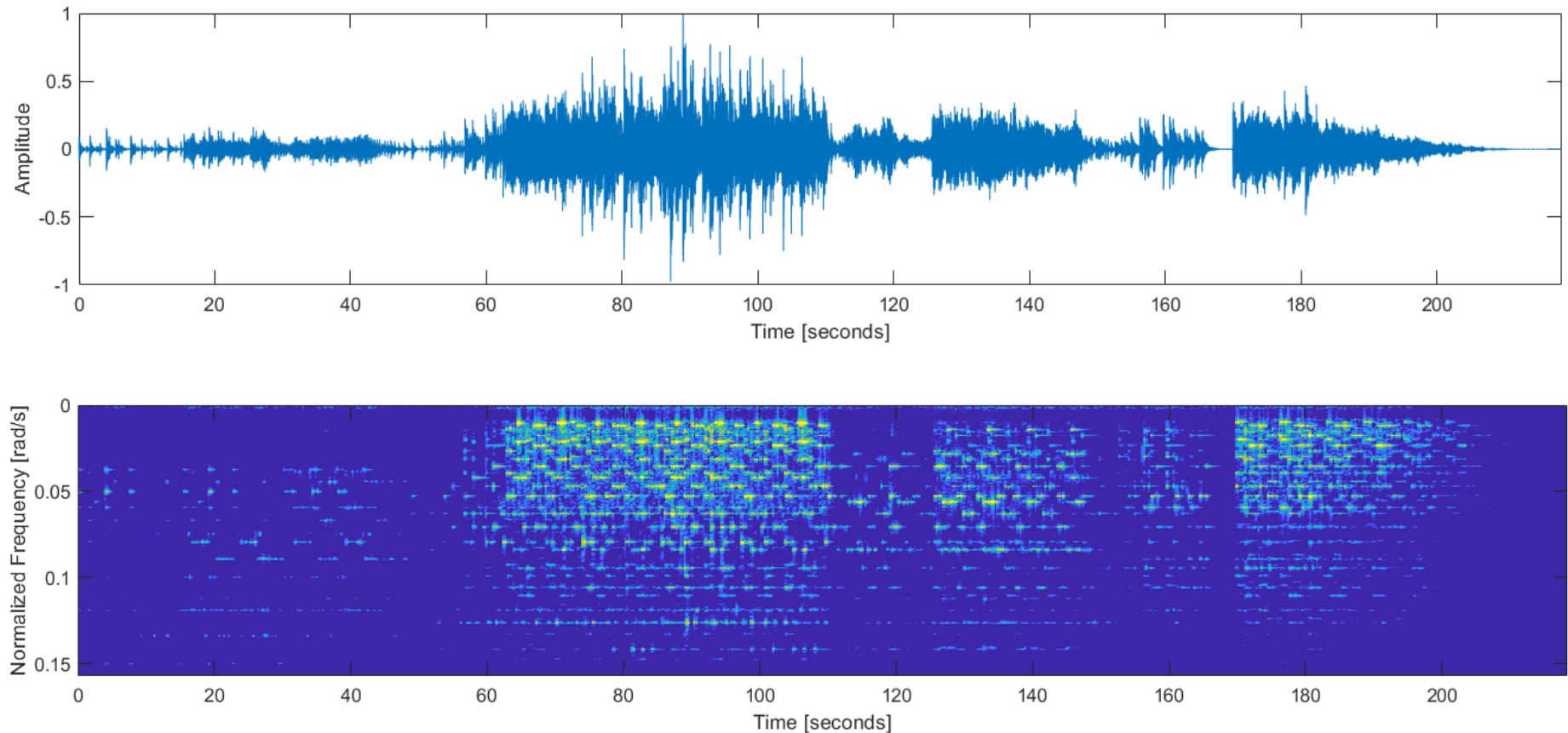
The Short-Time Fourier Transform

■ Question: Why do I care about the STFT?



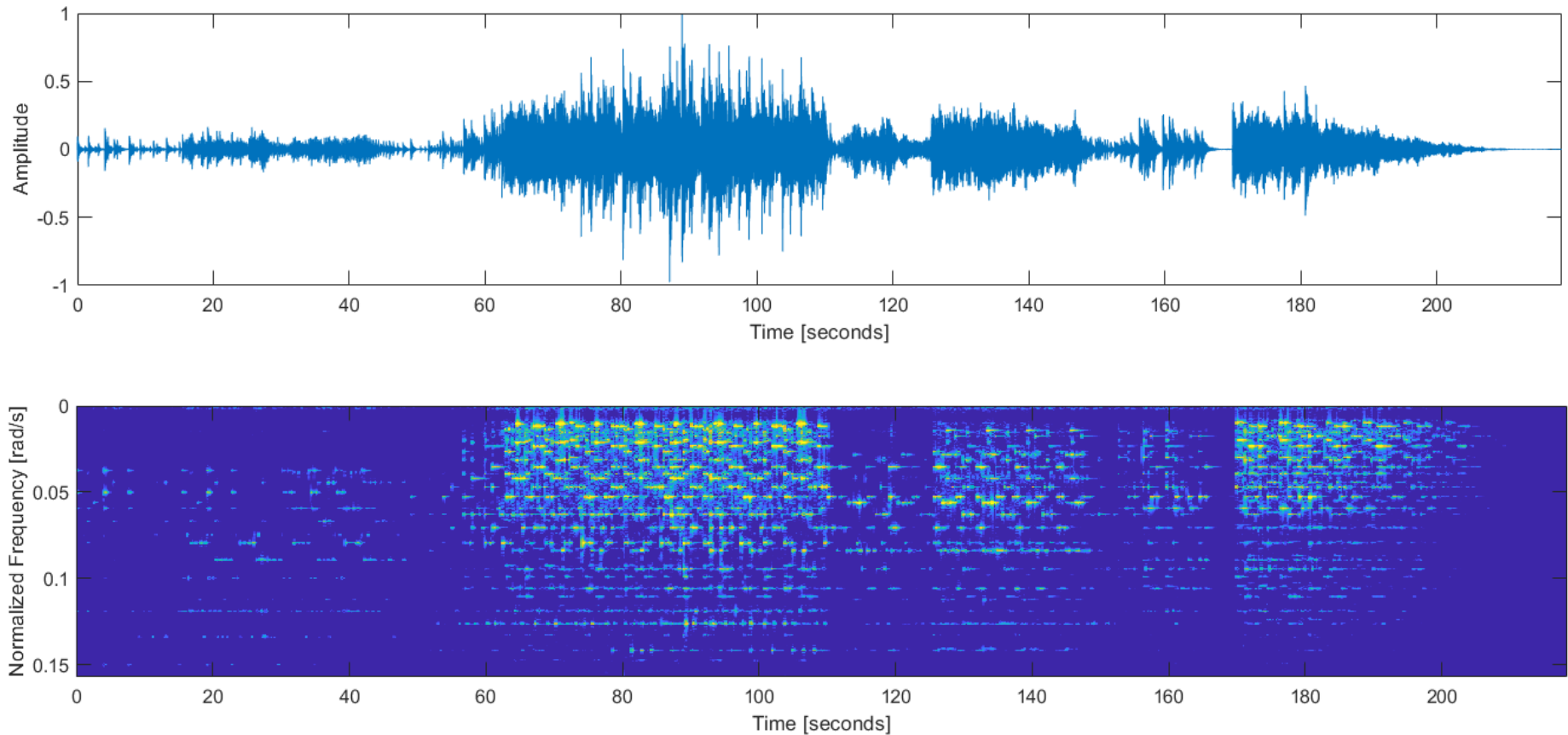
The Short-Time Fourier Transform

■ **Question:** How is this problematic for a real-time system?



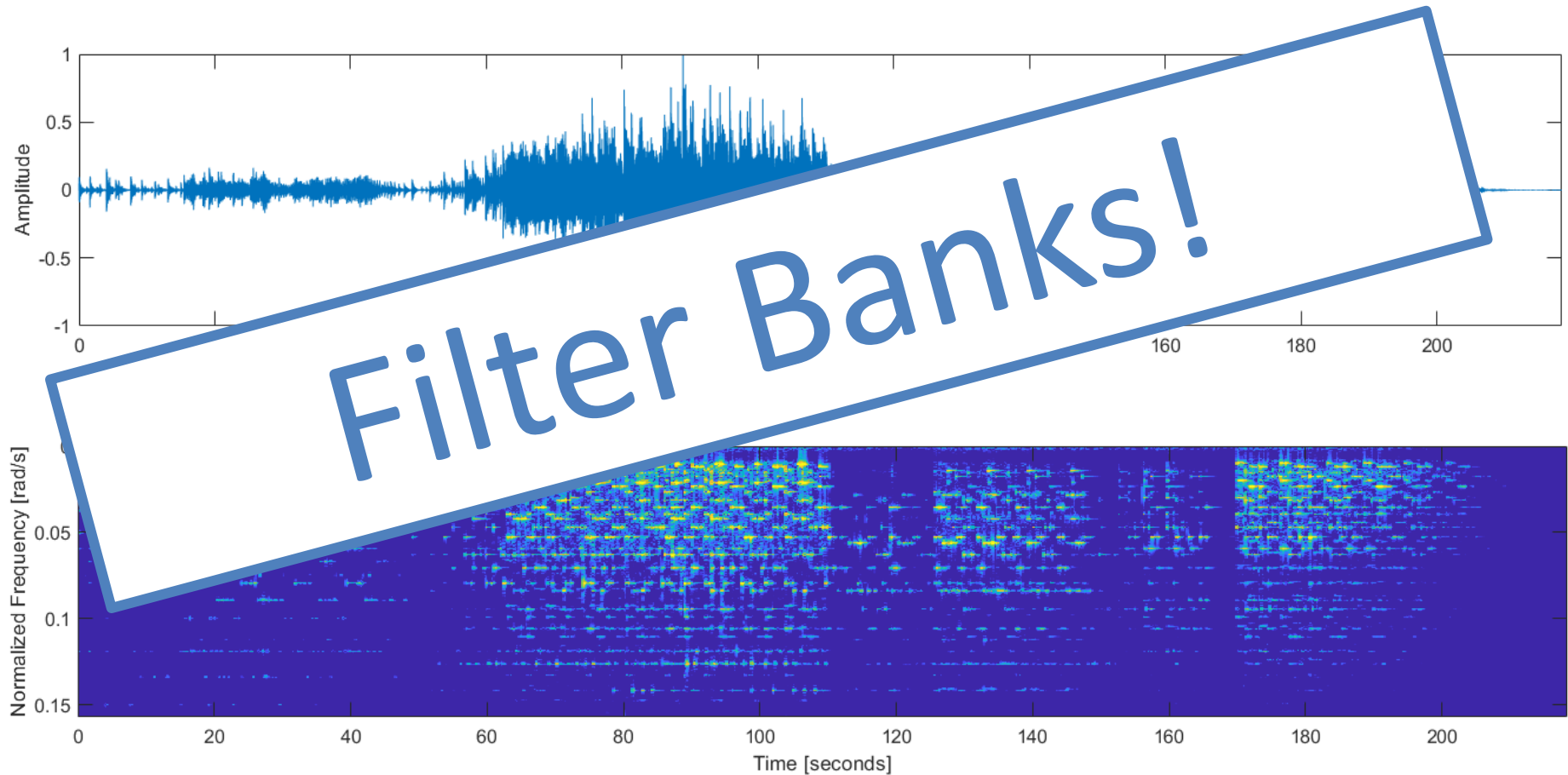
The Short-Time Fourier Transform

■ **Question:** How do I solve this problem??



The Short-Time Fourier Transform

■ **Question:** How do I solve this problem??



Lecture 23: Introduction to Filter Banks

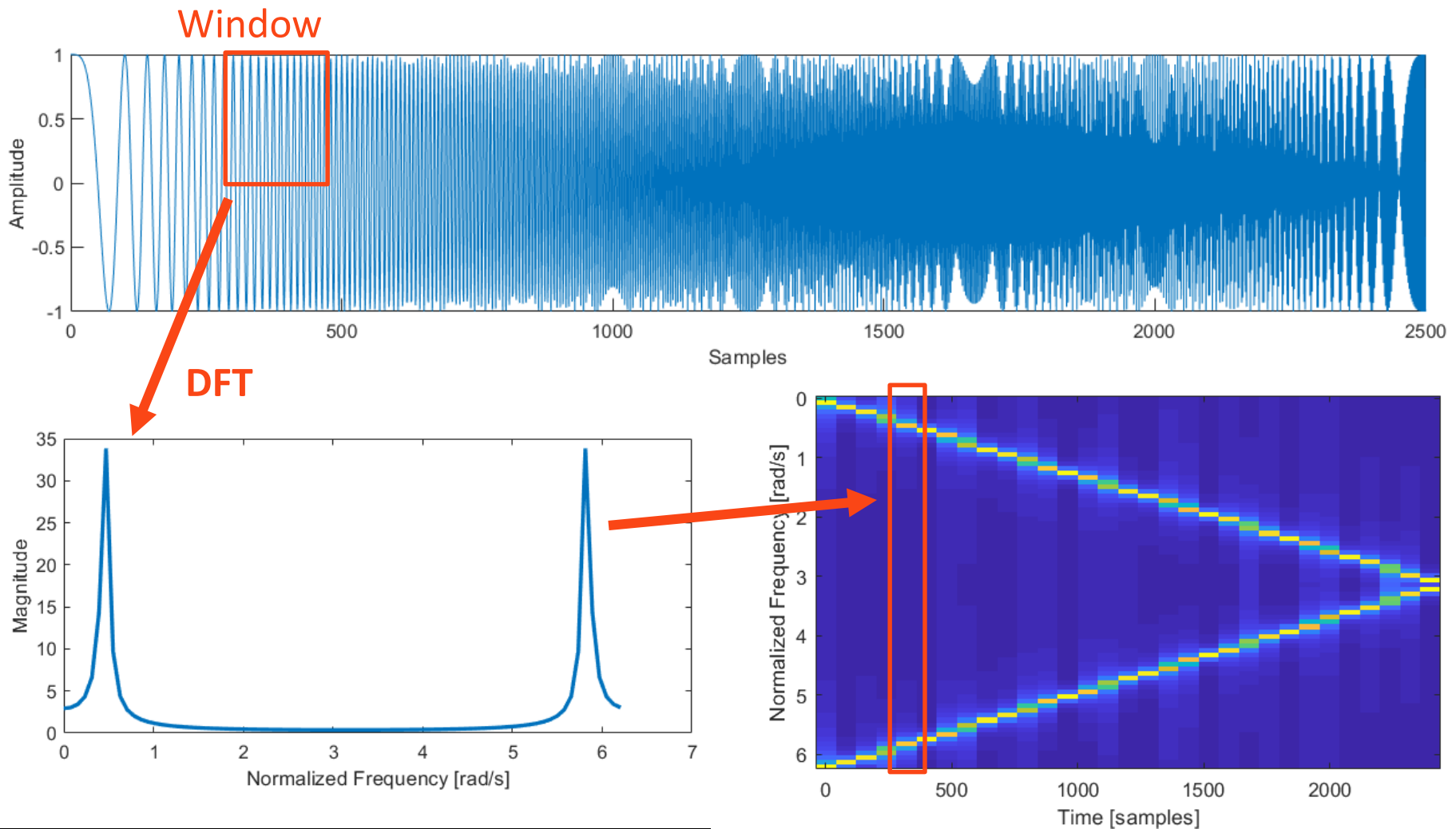
Foundations of Digital Signal Processing

Outline

- Review of Filter Design
- Short Time Fourier Transform
- **Inefficient Filter Banks**
- DFT Filter Bank

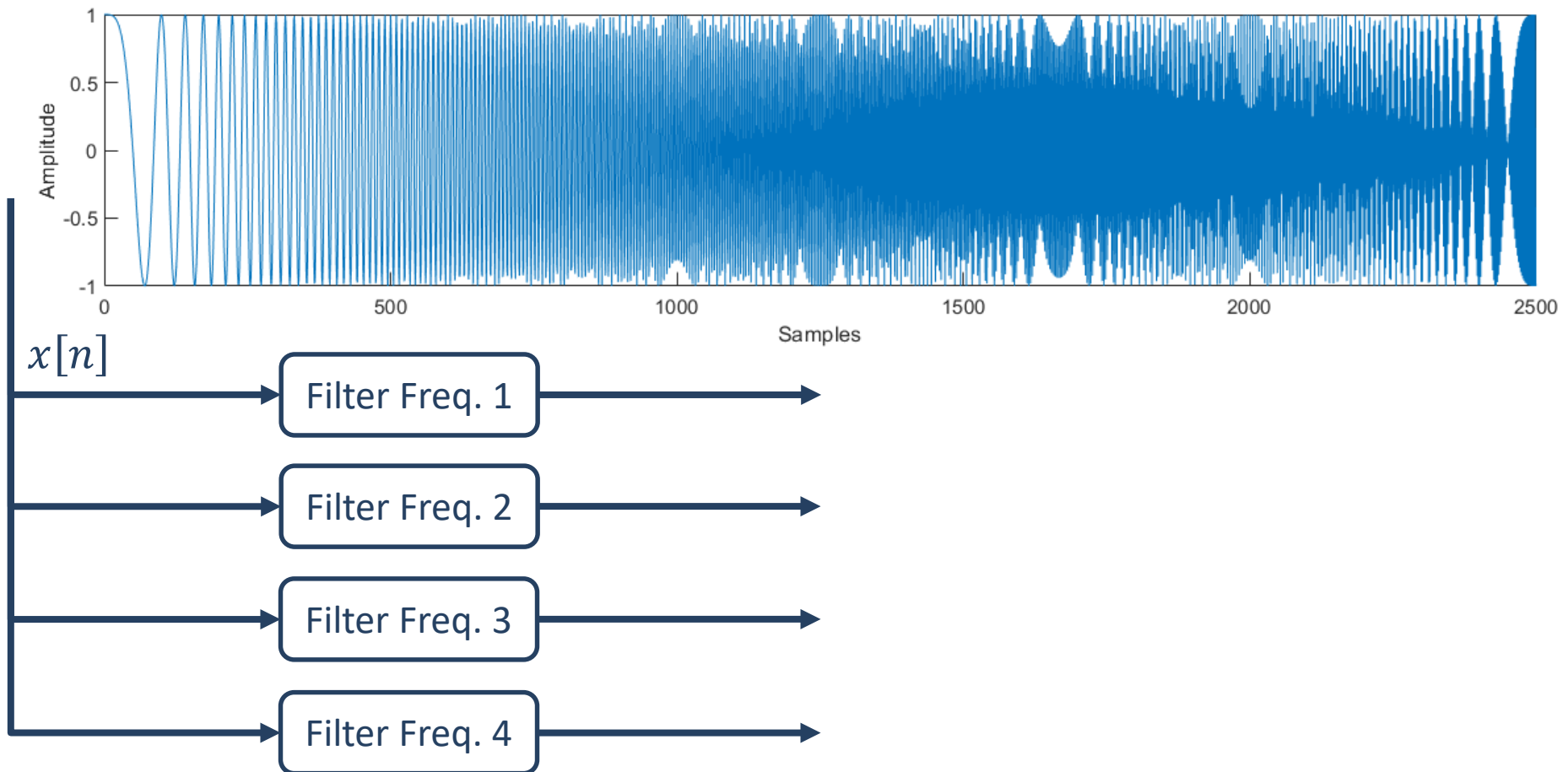
Filter Banks

■ The Short Time Fourier Transform Process



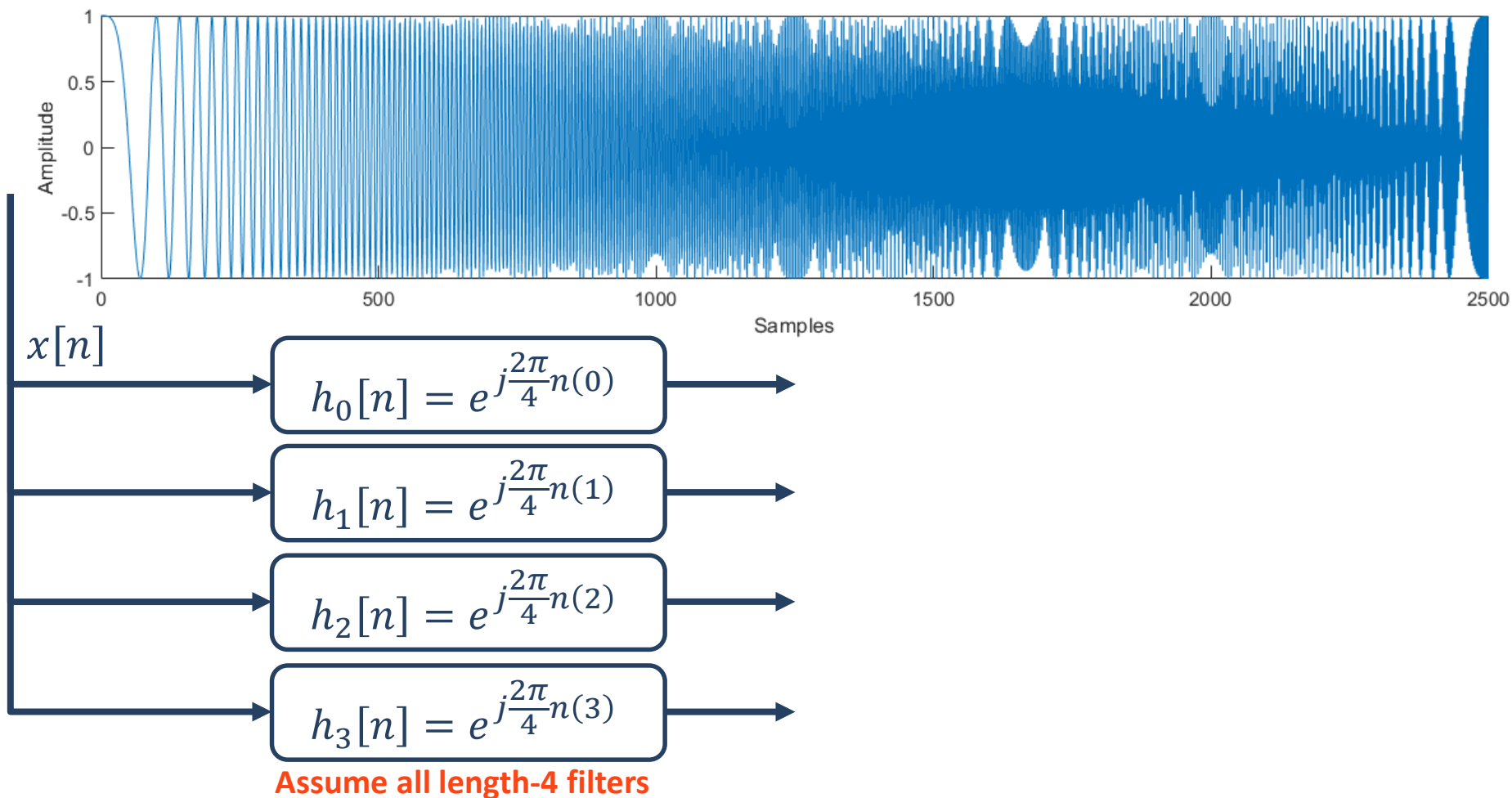
Filter Banks

■ Inefficient Filter Bank Process



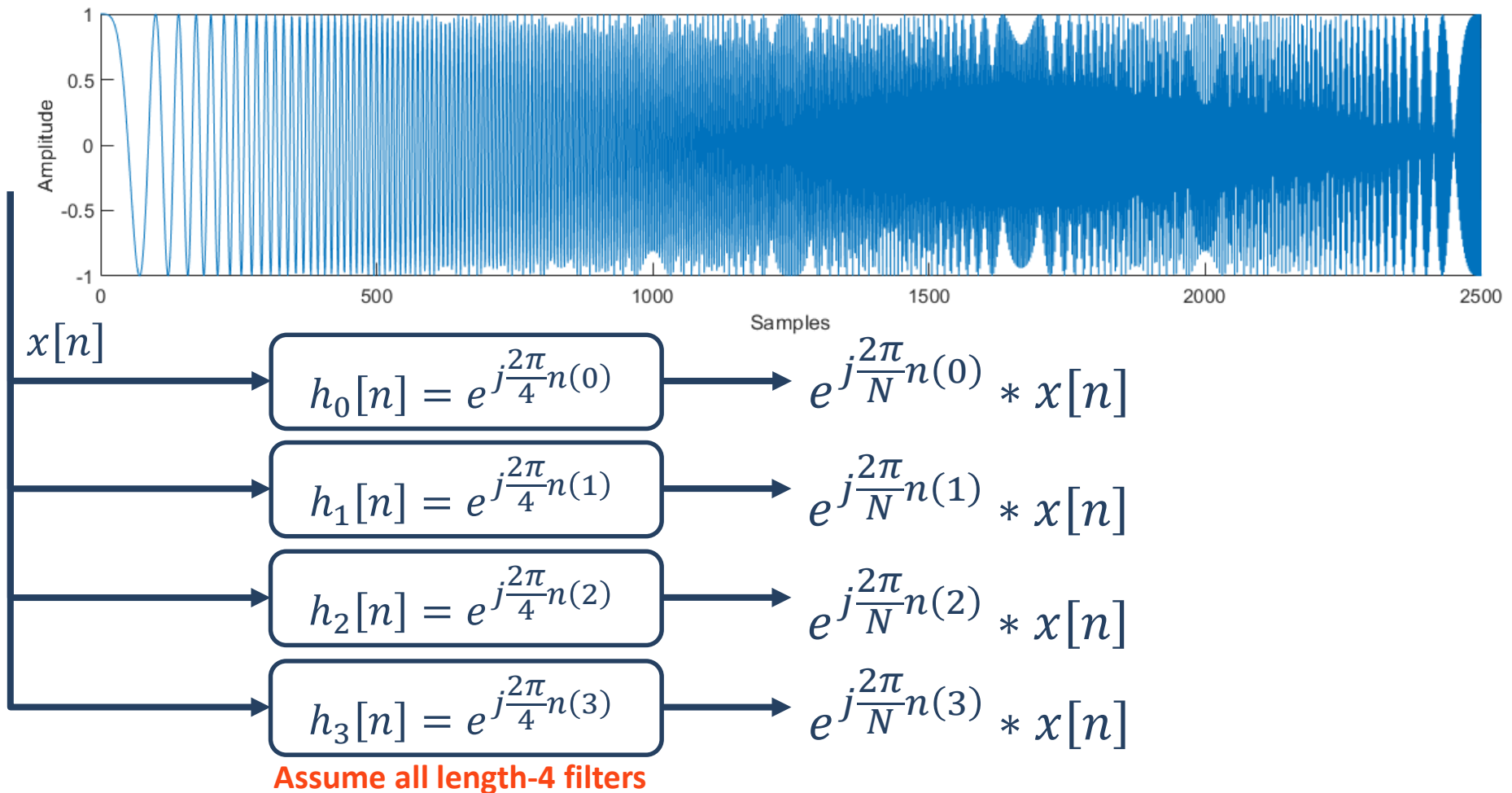
Filter Banks

■ Inefficient Filter Bank Process



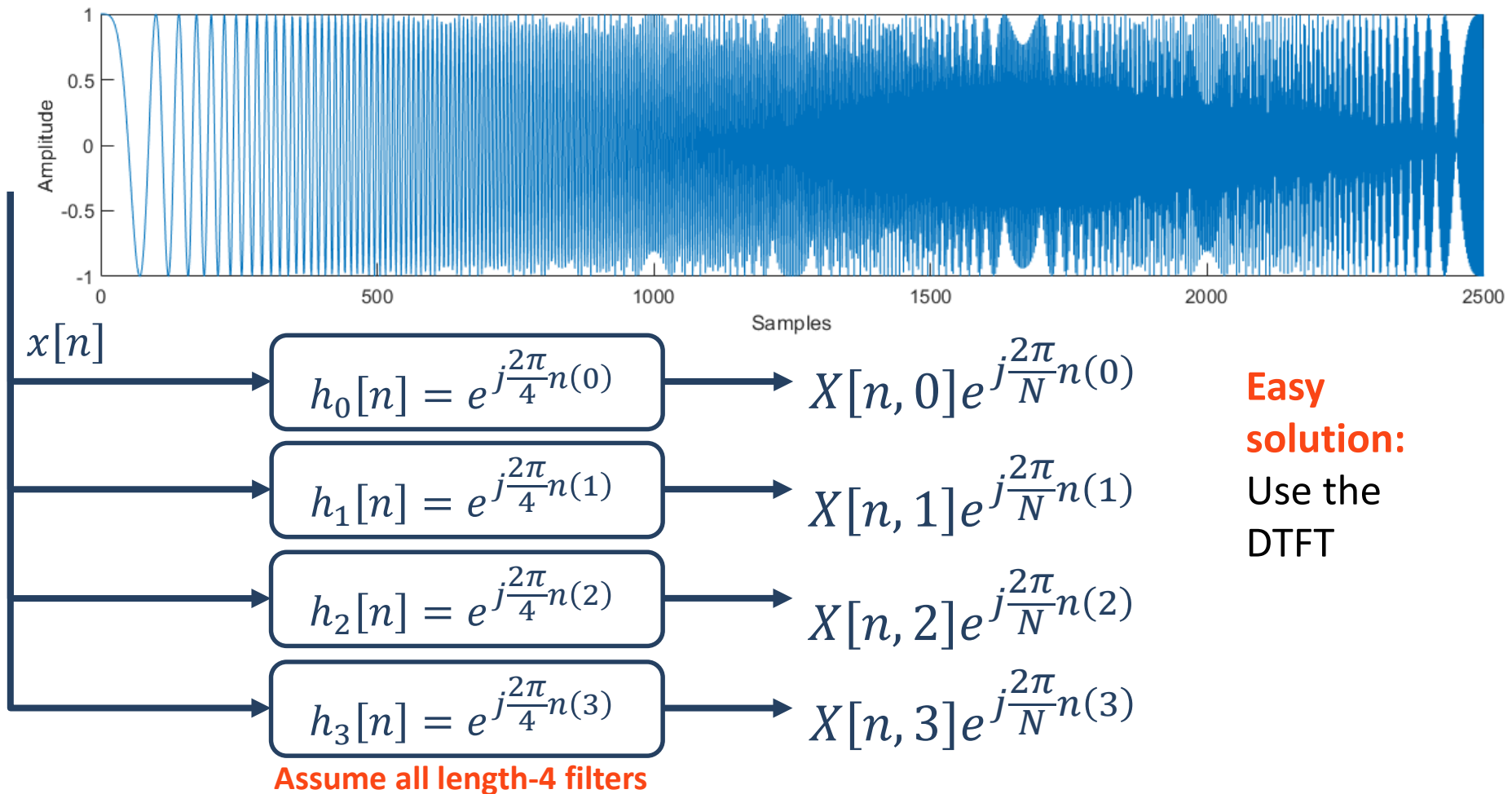
Filter Banks

■ Inefficient Filter Bank Process



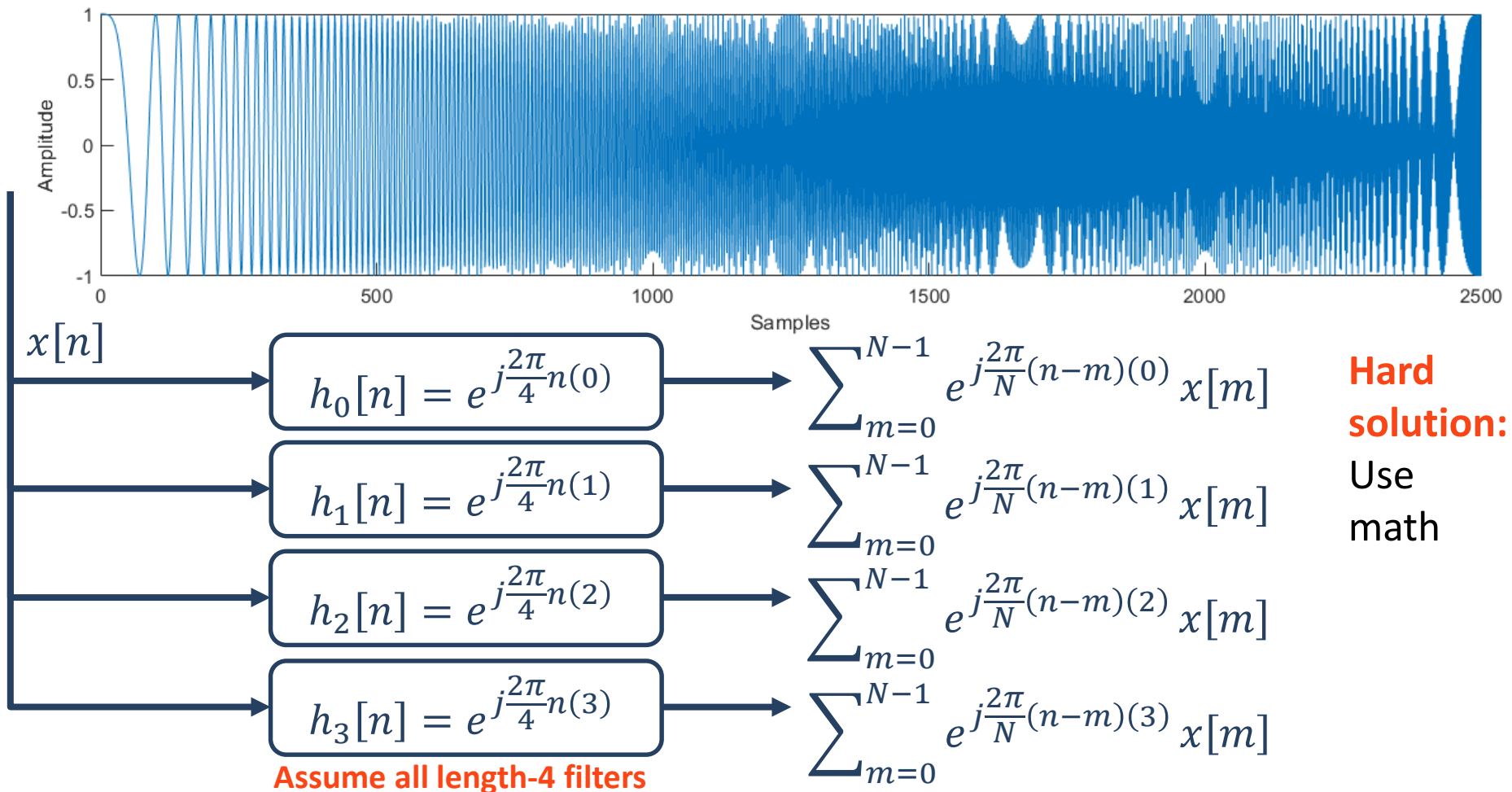
Filter Banks

■ Inefficient Filter Bank Process



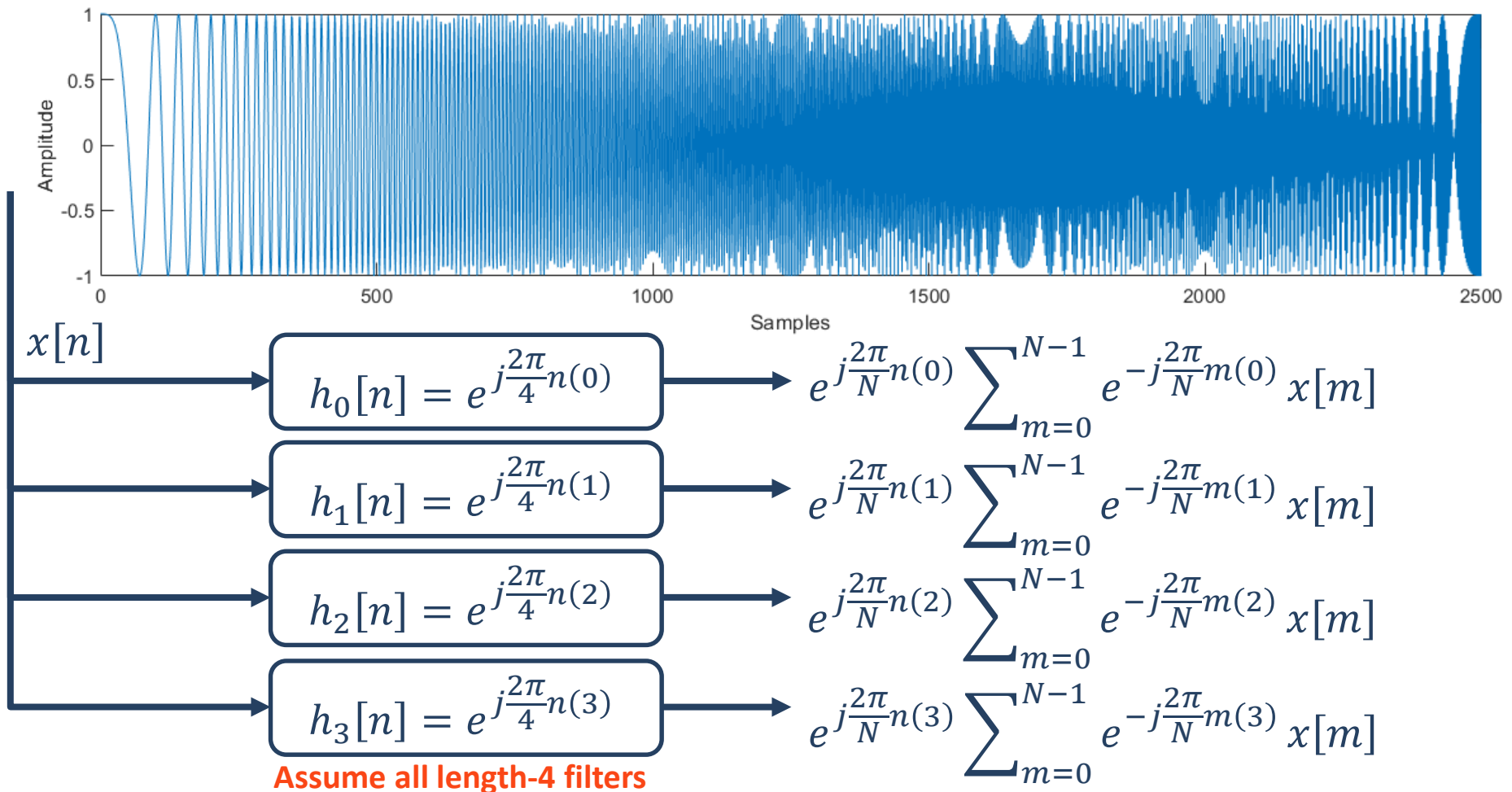
Filter Banks

■ Inefficient Filter Bank Process



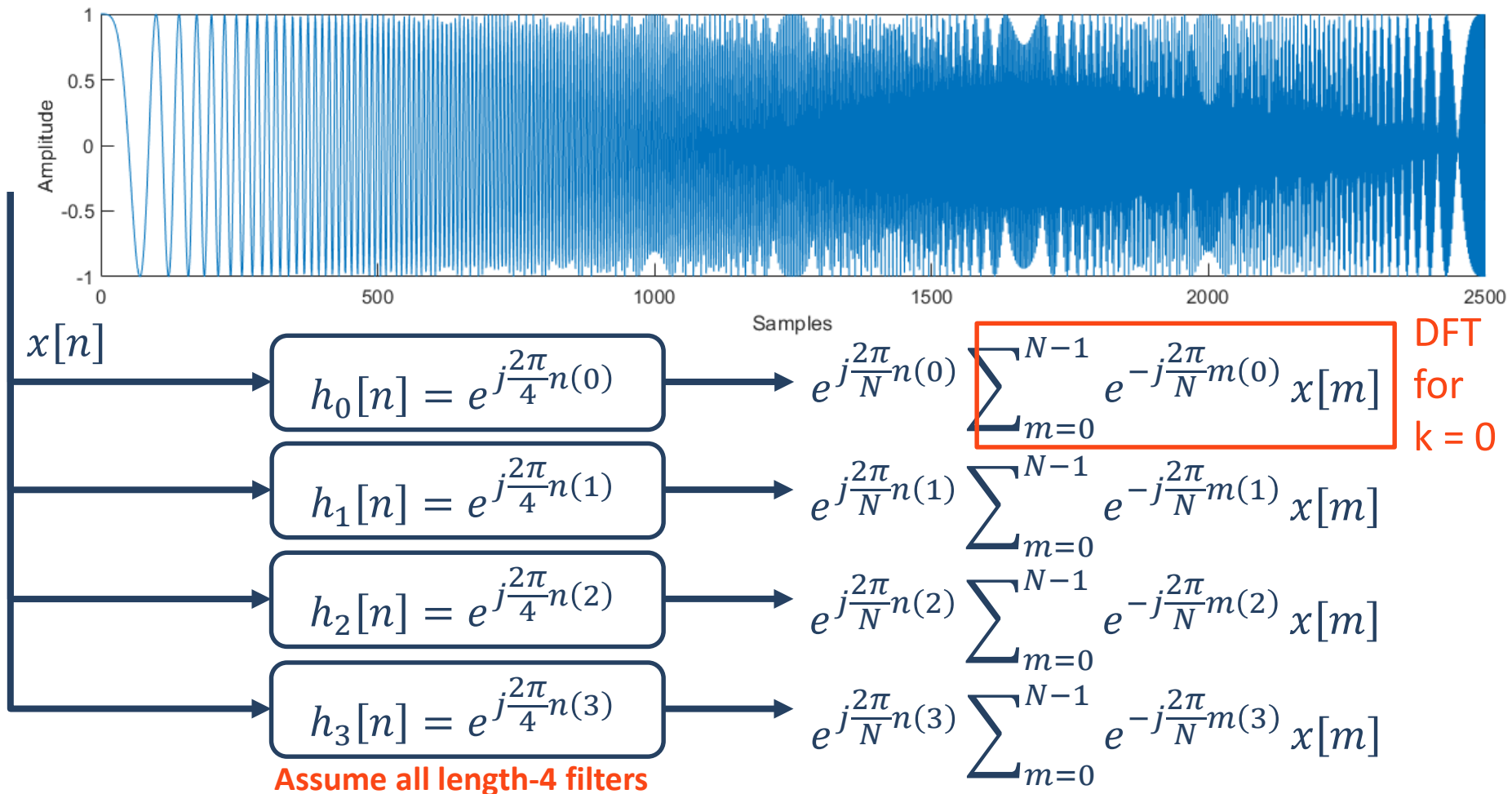
Filter Banks

■ Inefficient Filter Bank Process



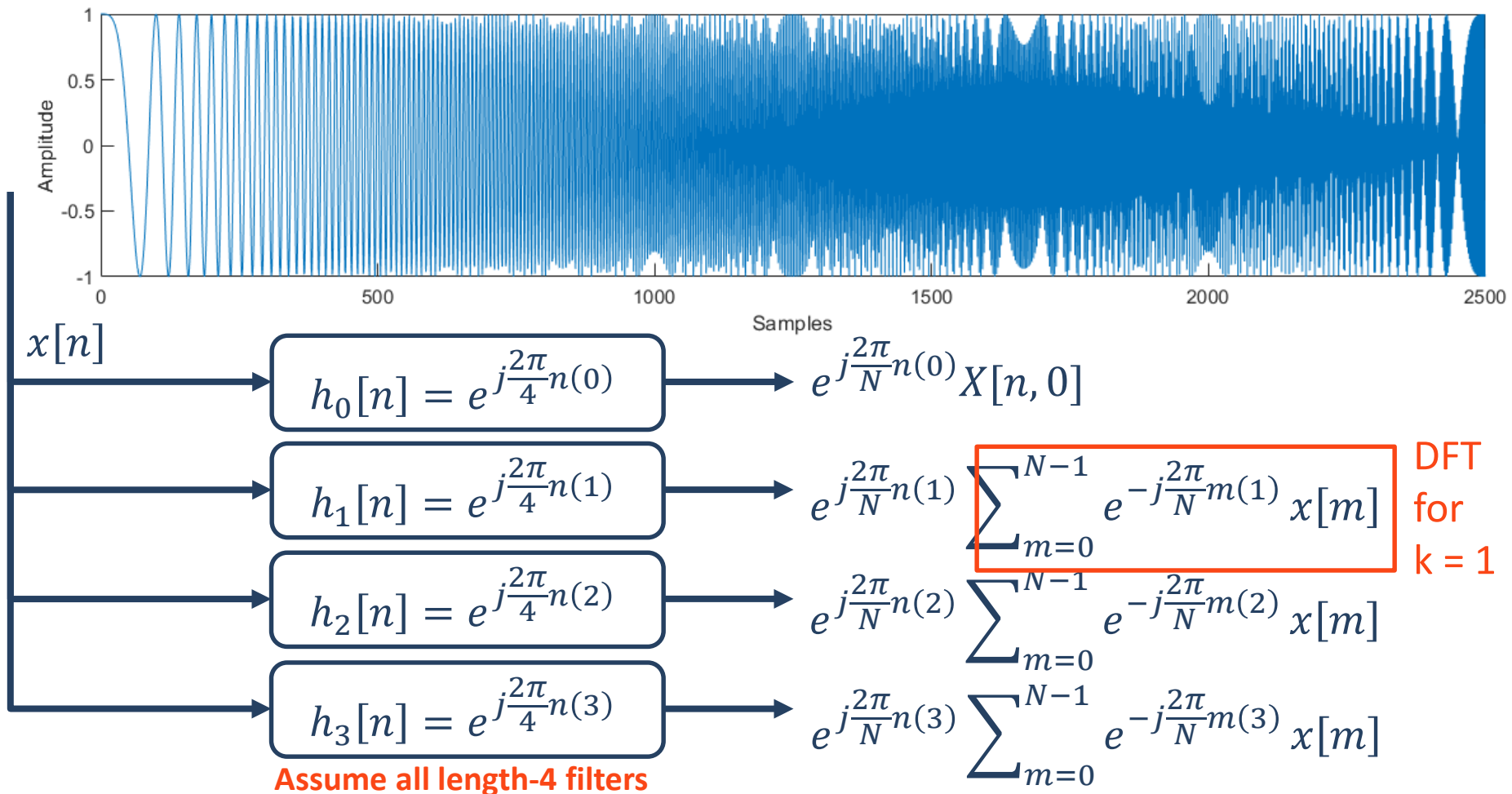
Filter Banks

■ Inefficient Filter Bank Process



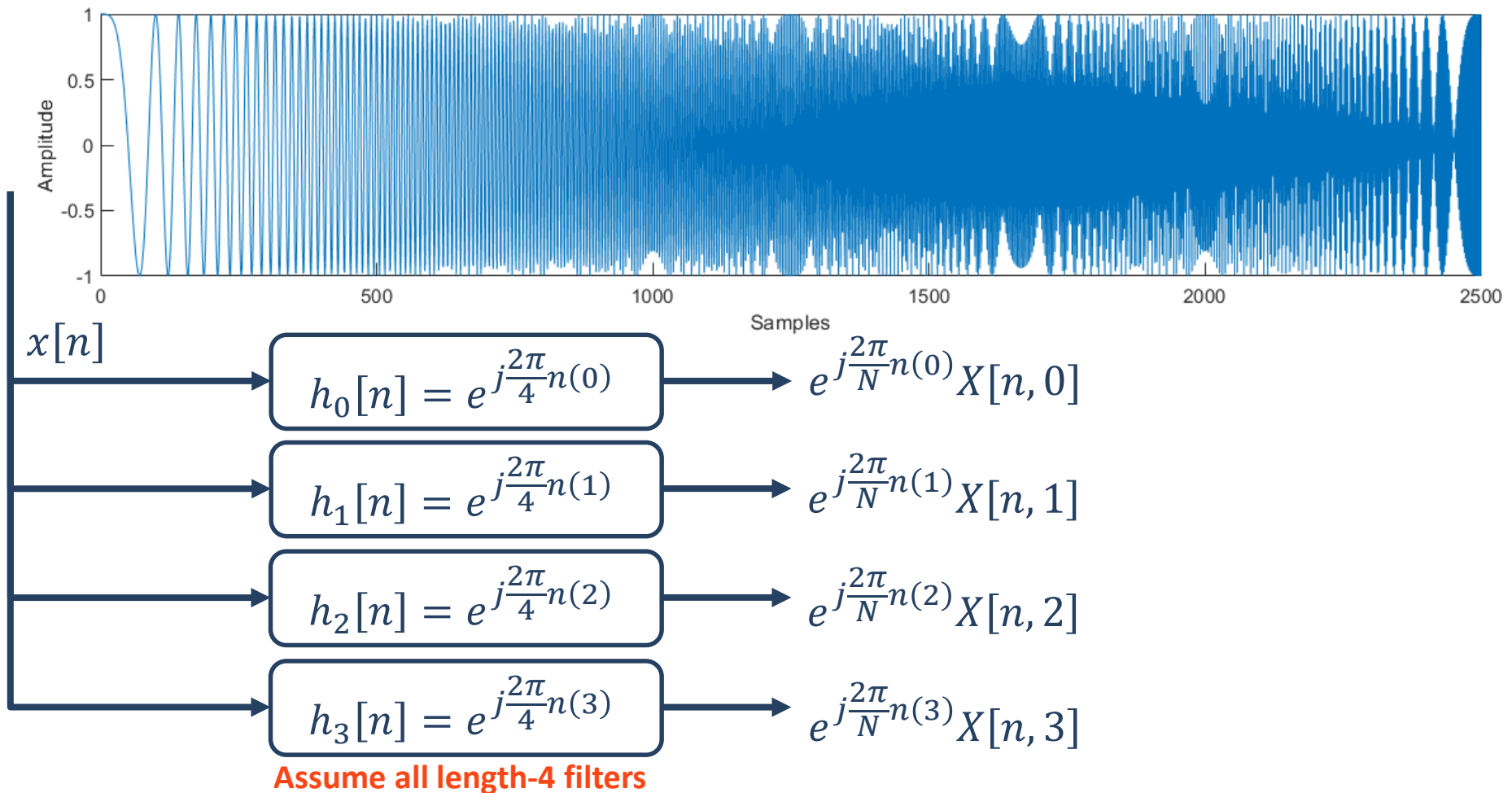
Filter Banks

■ Inefficient Filter Bank Process



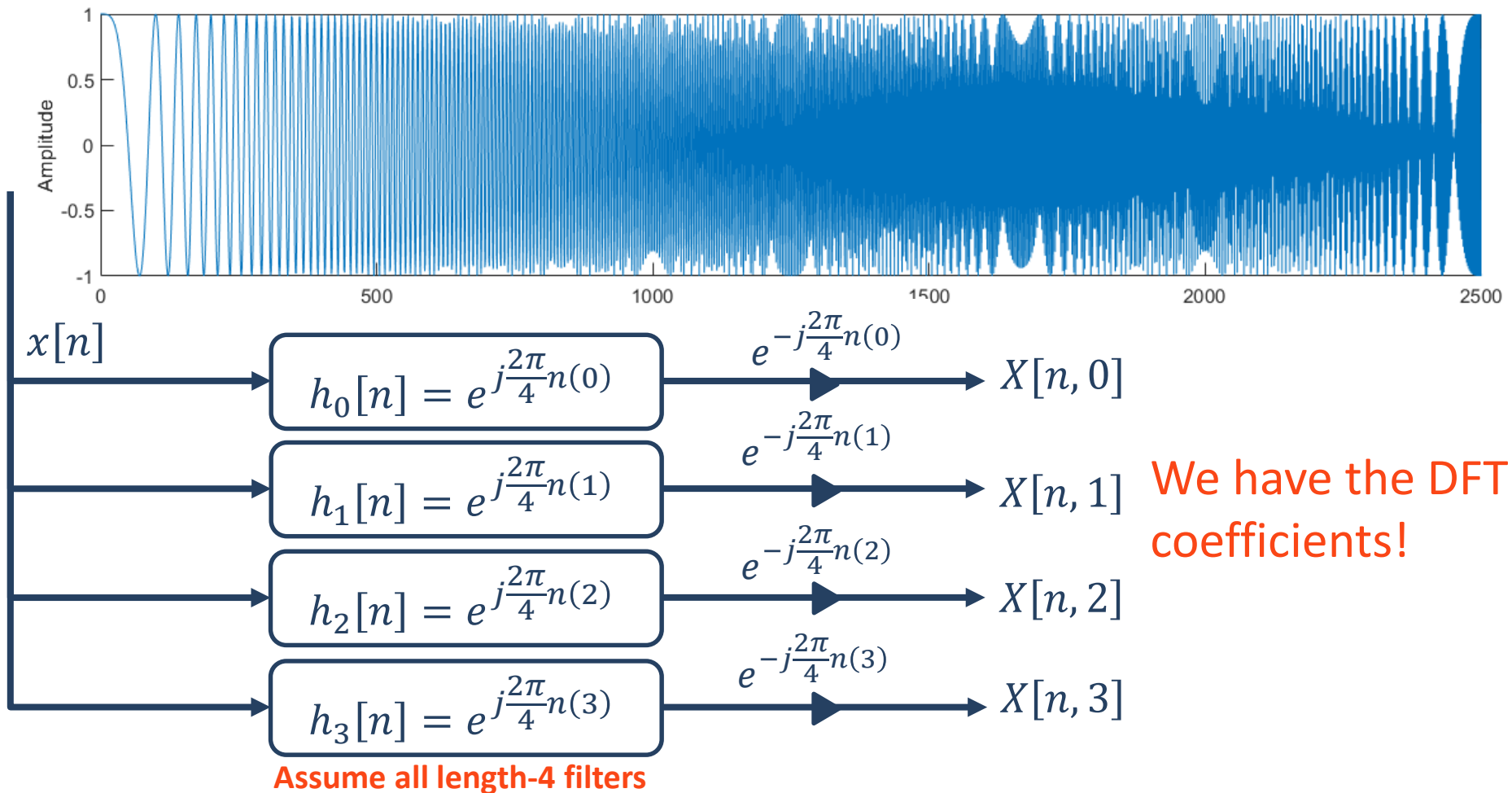
Filter Banks

■ Inefficient Filter Bank Process



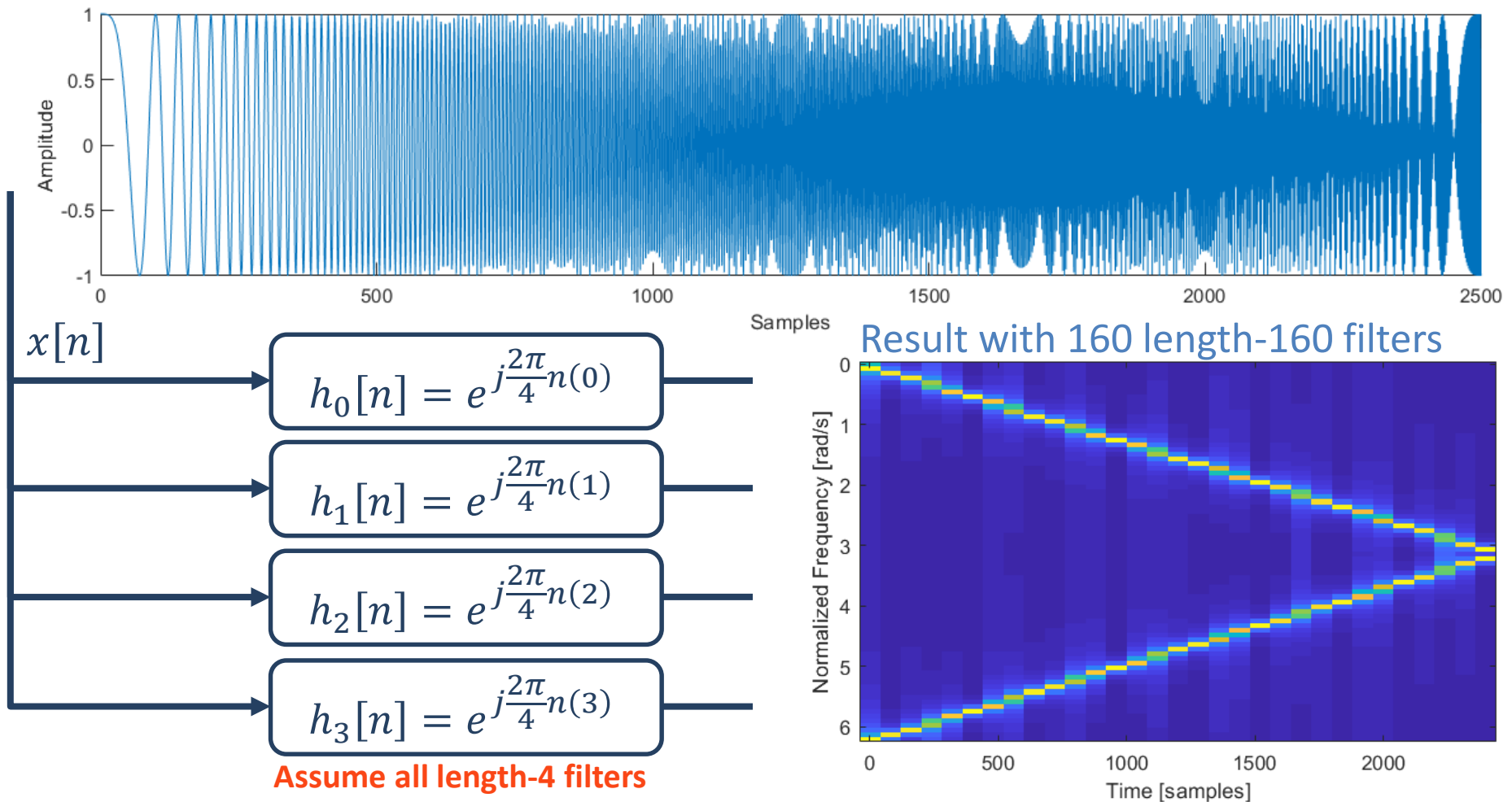
Filter Banks

■ Inefficient Filter Bank Process



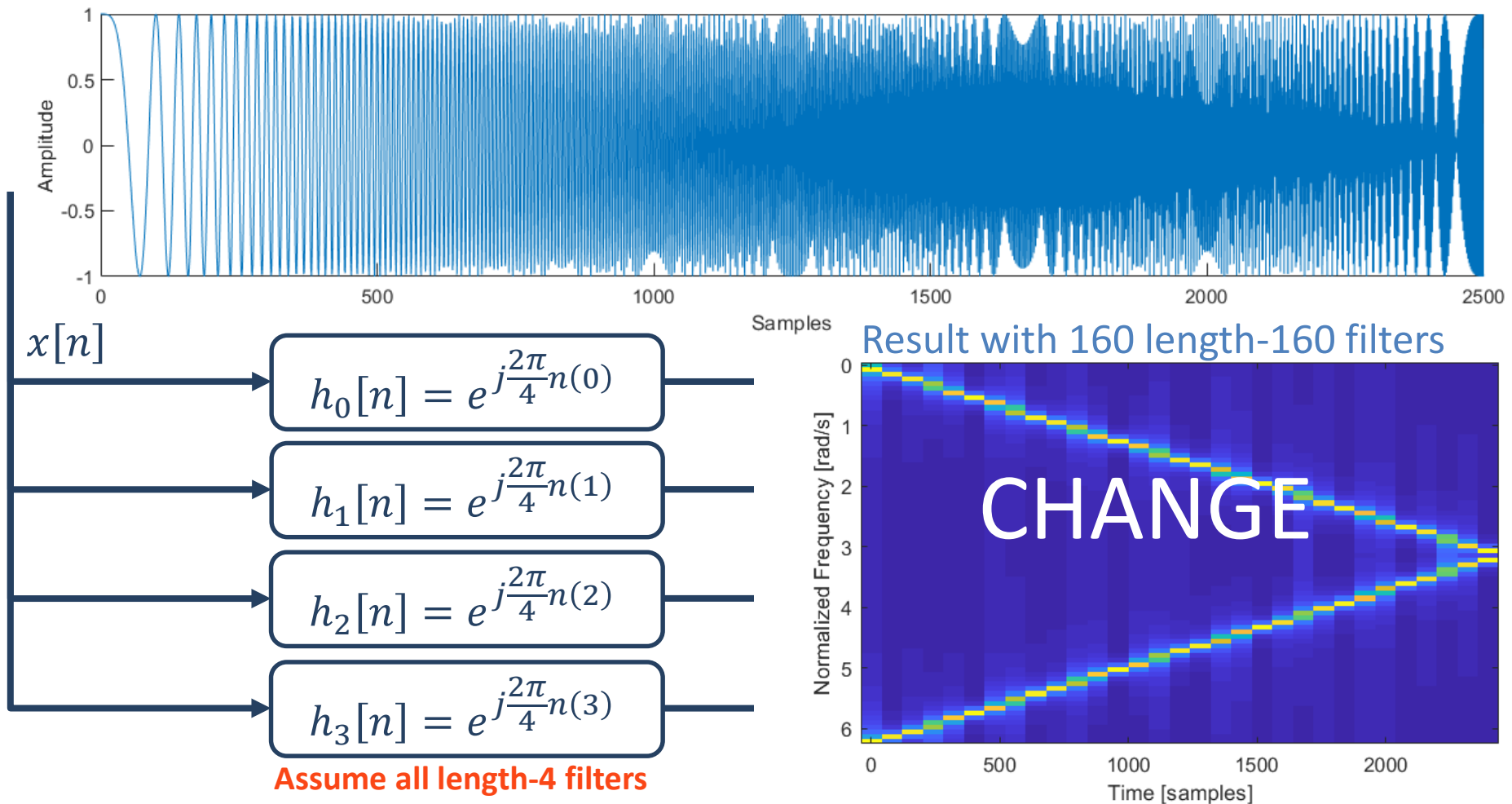
Filter Banks

■ Inefficient Filter Bank Process



Filter Banks

■ **Question:** Why is this not a preferred approach?



Filter Banks

■ Question: Why is this not a preferred approach?

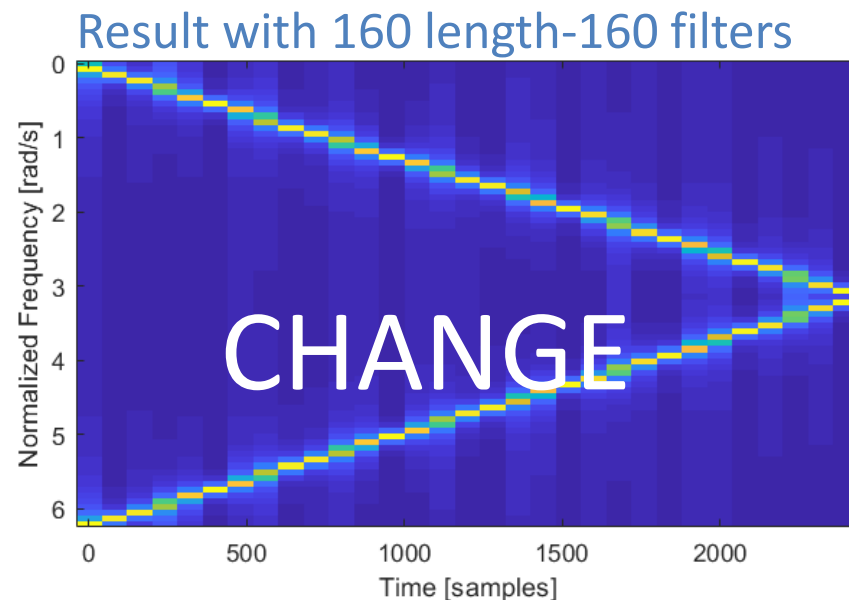
- It is really expensive!

■ STFT Approach

- W^2 multiplications for every W samples
- $W^2 = 160^2 = 25,600$

■ Filter Bank Approach

- W^3 multiplications for every W samples
- $W^3 = 160^3 = 4,086,000$



Lecture 23: Introduction to Filter Banks

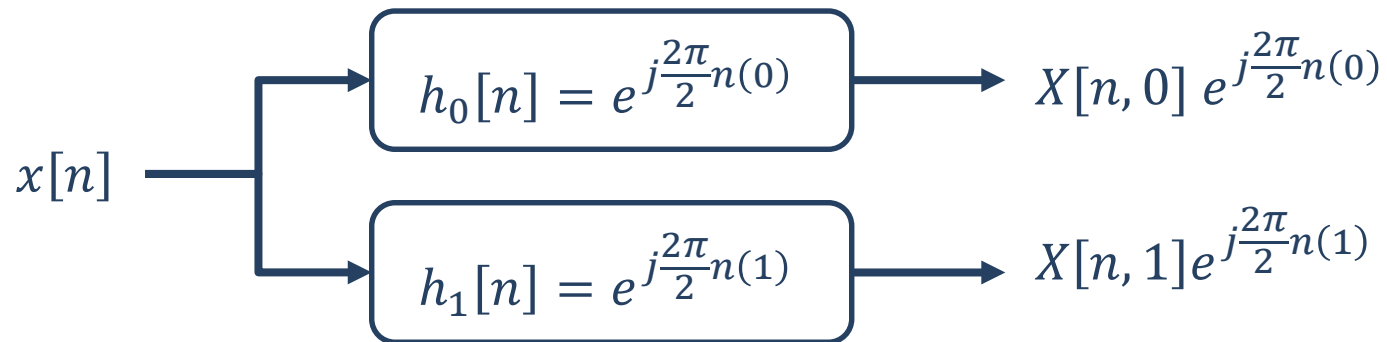
Foundations of Digital Signal Processing

Outline

- Review of Filter Design
- Short Time Fourier Transform
- Inefficient Filter Banks
- **DFT Filter Bank**

Filter Banks

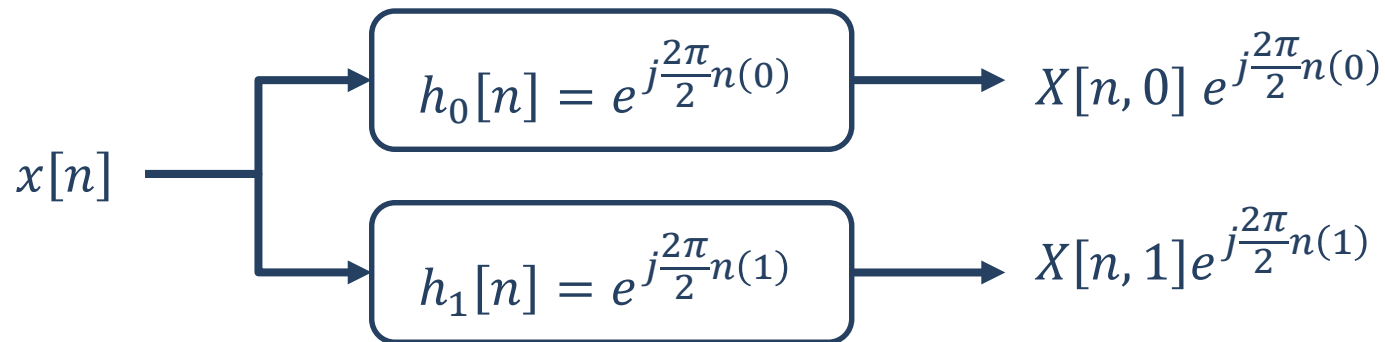
- Consider the following filter bank



Filter Banks

■ Consider the following filter bank

- **Question:** How do I make this like the STFT????

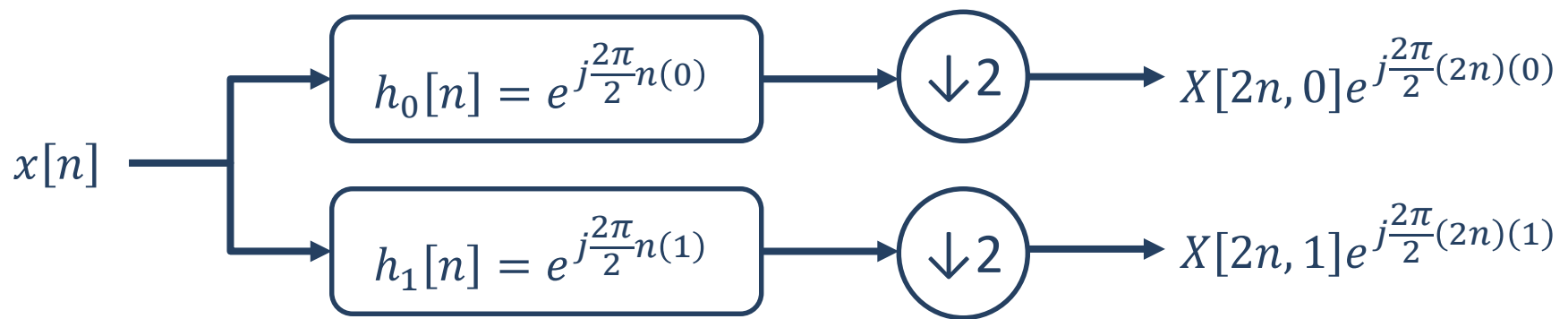


- **Now recall:** The short-time Fourier Transform gave us
 - ◇ $X[Mn, 0]$ <- M = shift amount (often window length)

Filter Banks

■ Consider the following filter bank

- **Question:** How do I make this like the STFT????

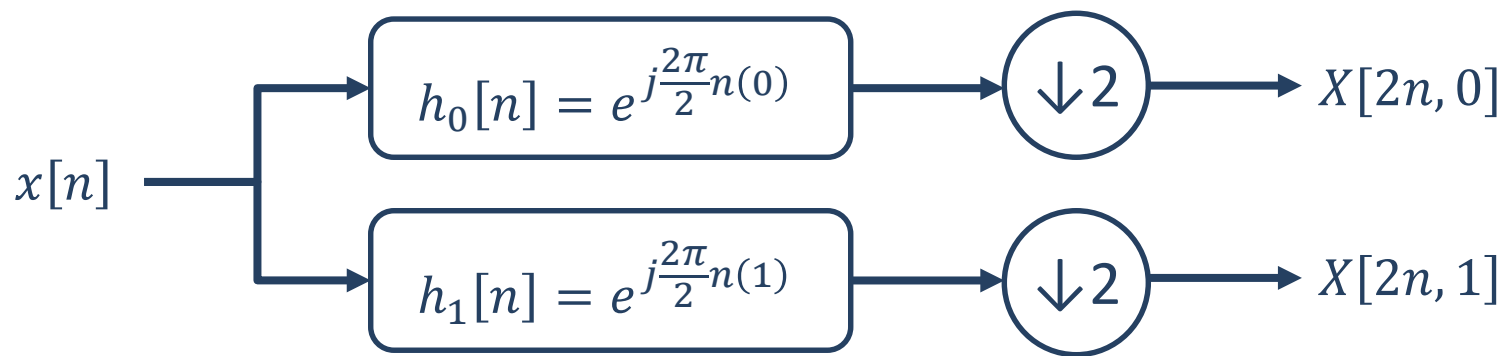


- **Now recall:** The short-time Fourier Transform gave us
 - ◇ $X[Mn, 0]$ <- M = shift amount (often window length)

Filter Banks

■ Consider the following filter bank

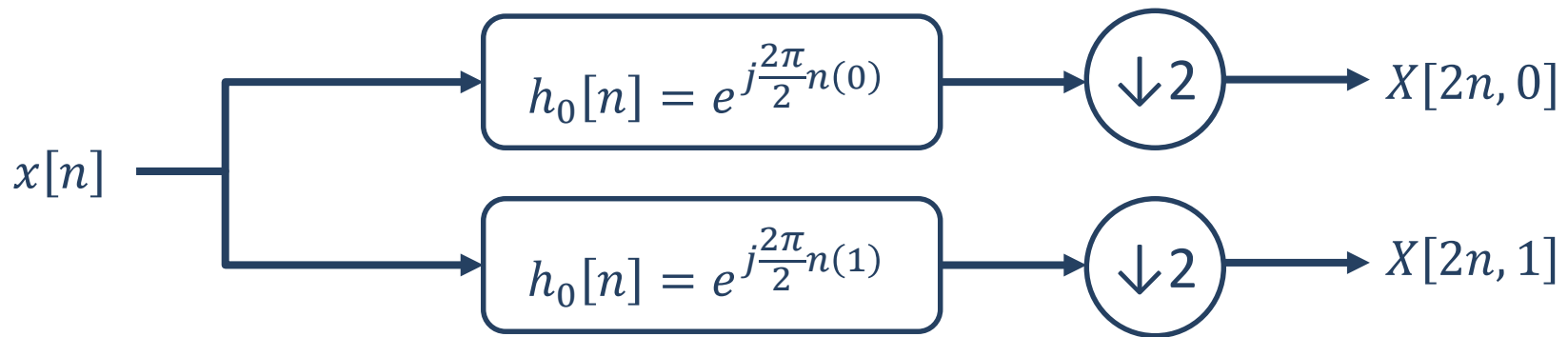
- **Question:** How do I make this like the STFT????



- **Now recall:** The short-time Fourier Transform gave us
 - ◇ $X[Mn, 0]$ <- M = shift amount (often window length)

Filter Banks

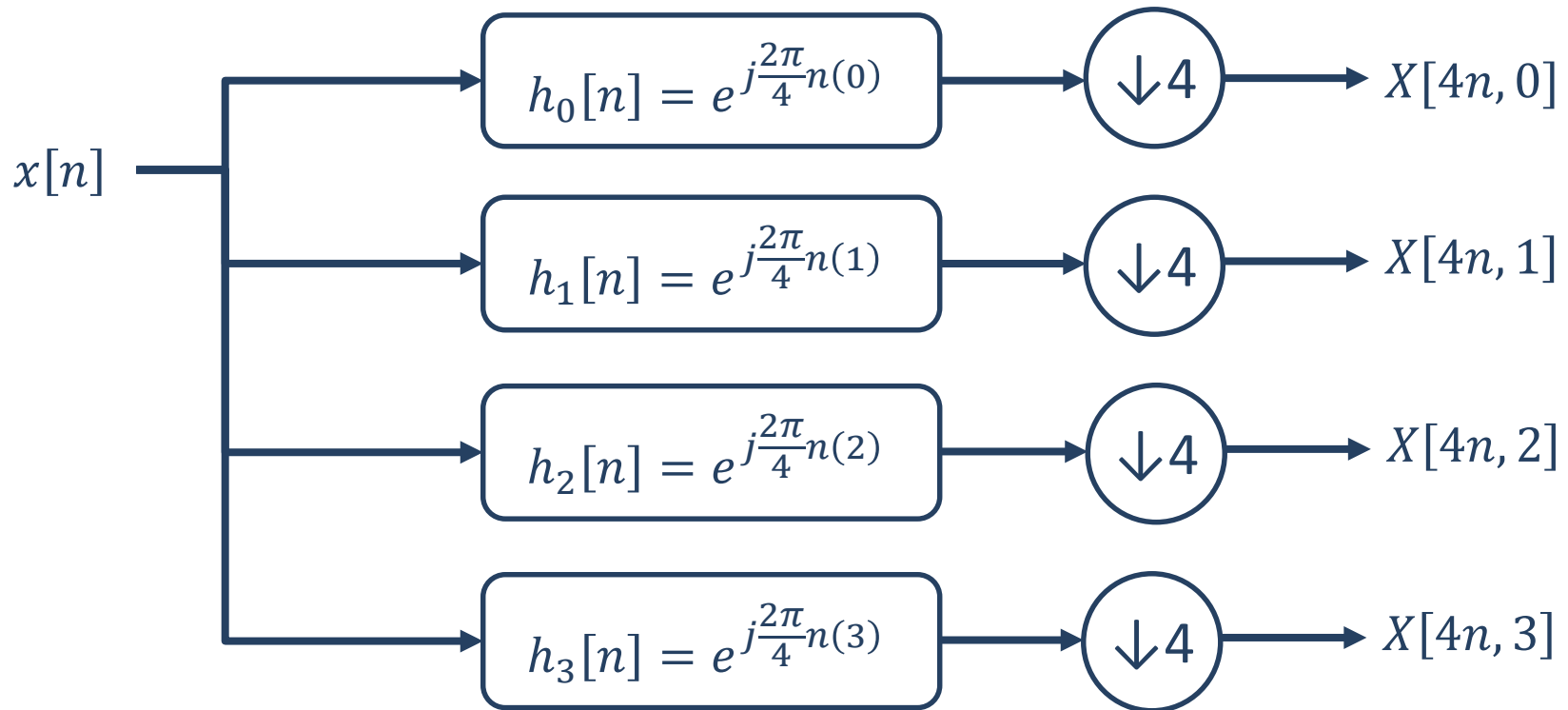
- Consider the following filter bank



- **Result:** It is now exactly the same as the STFT with a window of length 2 and shift of 2 between windows
- **But,** we do not need to buffer $x[n]$

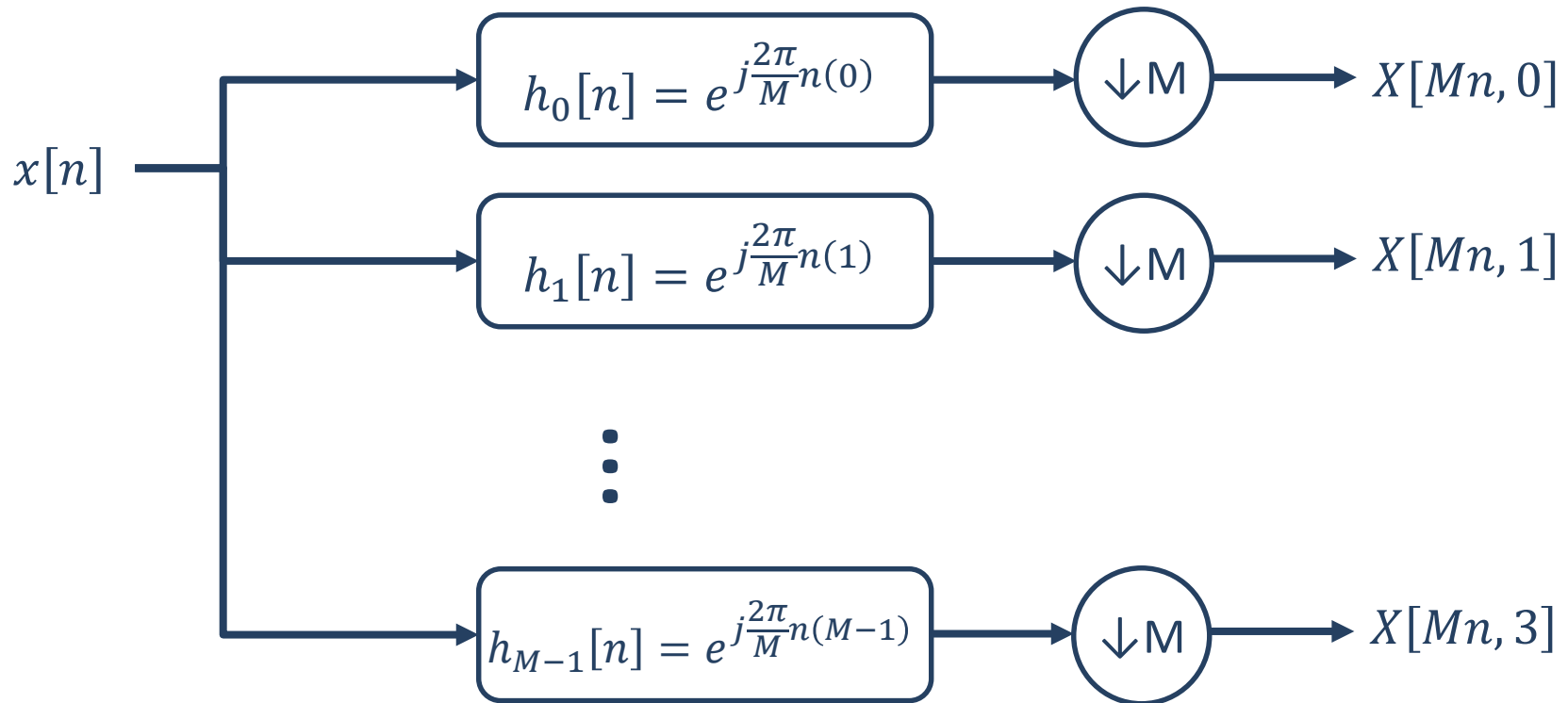
Filter Banks

■ Consider the following filter bank



Filter Banks

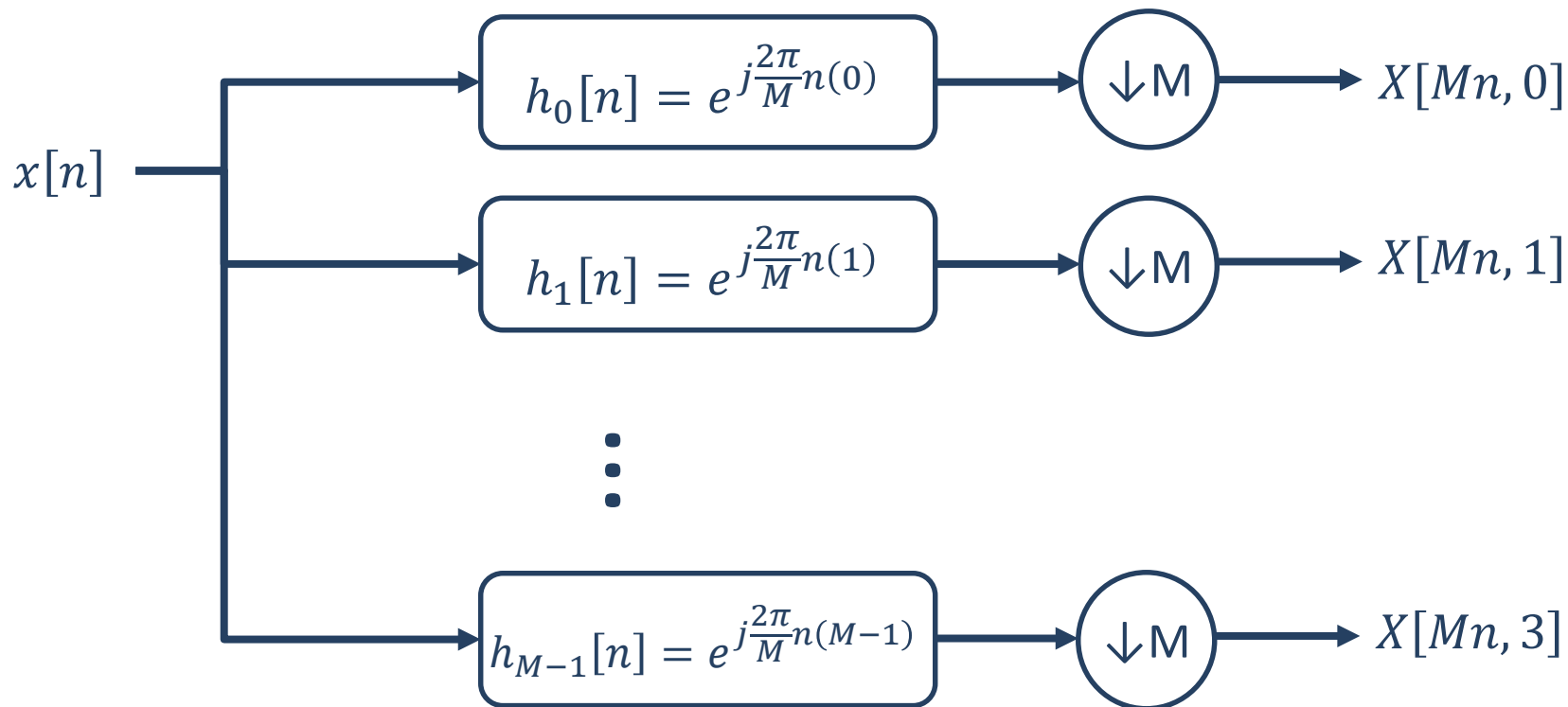
- Hence, this is an M-point DFT



- So, I can implement the STFT as a filter bank...
 - Can I do more?

Filter Banks

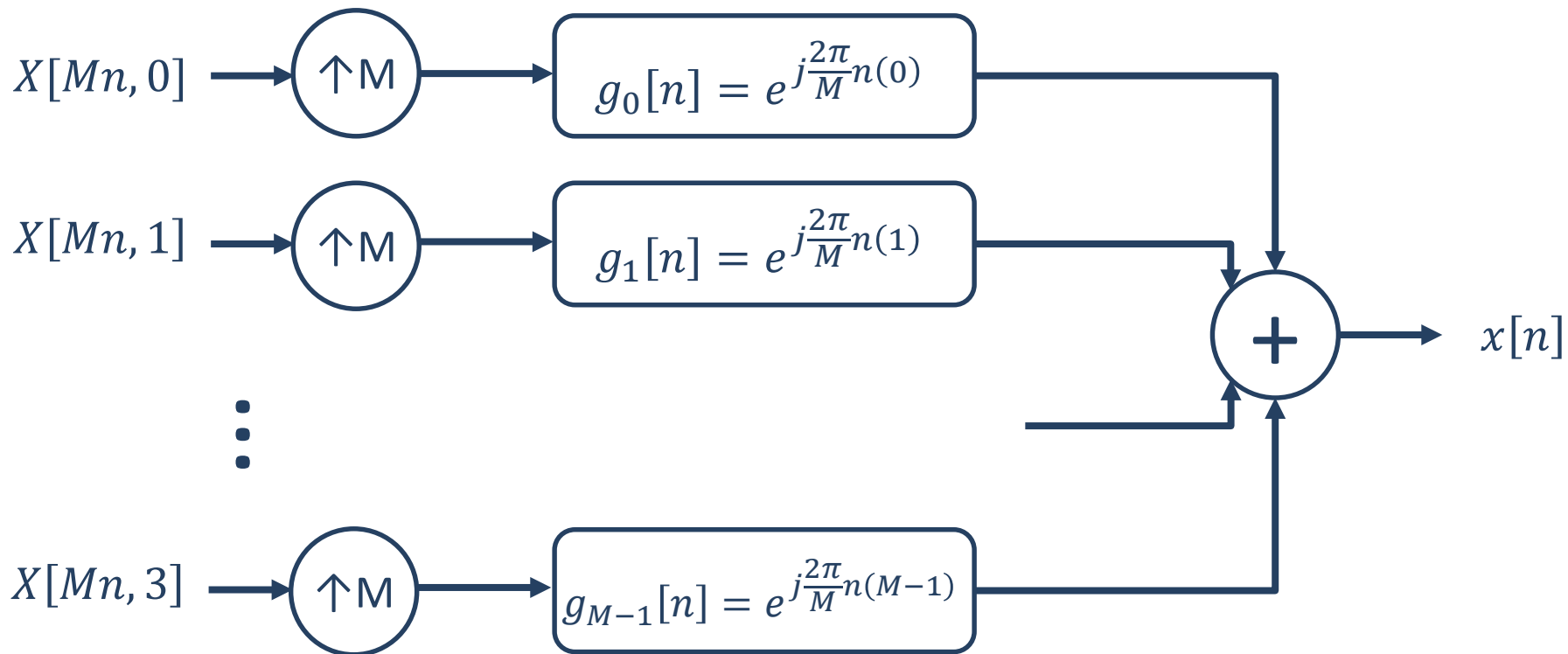
- Hence, this is an M-point DFT



- So, I can implement the STFT as a filter bank...
 - **Question:** How do I get back into the time domain?

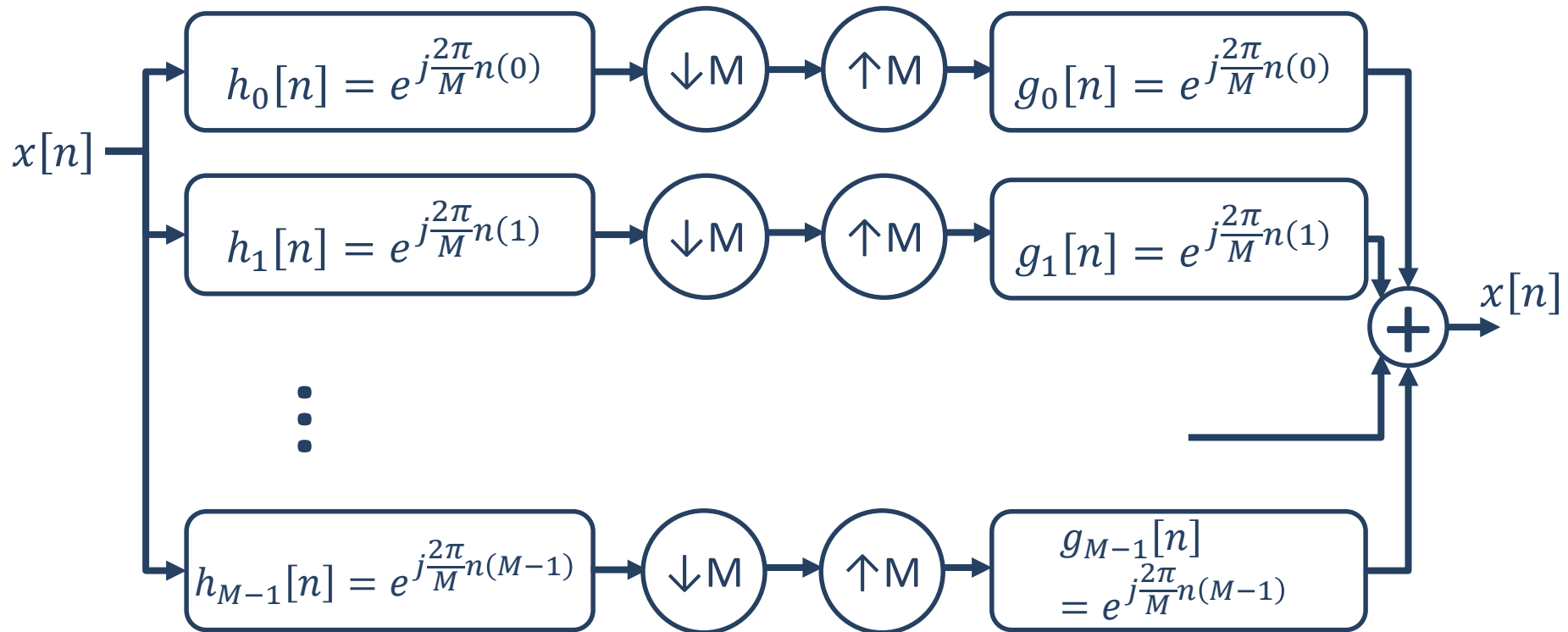
Filter Banks

■ Hence, this is an M-point IDFT



Filter Banks

- Hence, this is an M-point DFT and IDFT



Filter Banks

- Hence, this is an $M=2$ point DFT and IDFT

