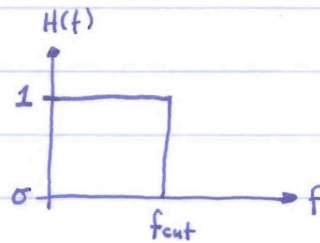


LINEAR FILTERS

Ideal filter

frequency response function
FFT of filter weights

$$Y(f) = H(f) \cdot X(f)$$

filtered output filter raw input

↓

$$y(t) = \int_{-\infty}^{\infty} h(\tau) x(t-\tau) d\tau = h(t) * x(t)$$

note that for $x(t-\tau) = \delta(t-\tau)$.

$$y(t) = \int_{-\infty}^{\infty} h(\tau) \delta(t-\tau) d\tau = h(t)$$

↓ $H(f)$ is the FFT of the filtered output due to a unit impulse function.

Filter Design is to come up with "good" filter weights $h(t)$

- | | |
|--|---------------------------------------|
| 1. sharp cut-off | causes ringing |
| 2. good transient response | minimize ringing |
| 3. minimize "phase shift" | $H(f)$ real |
| 4. leave low frequency content undisturbed | flat pass-band response |
| 5. short filter span | $h(t)$ drops to zero fast |
| 6. efficient computationally | → the minimize filter loss |

Some of these criteria are contradictory, i.e., a sharp cut off requires a long filter span (Gibbs phenomenon)

Ideal Filter continued:

(4) What does the ideal filter do in the time domain?

$$\mathcal{F}^{-1}(\Pi(f, f_{cut})) = f_{cut} \text{sinc}(f_{cut}t)$$

what was leakage before is now called ringing *

as the data $x(t)$ is convolved with the sinc func.

→ smooth the edges of the filter, i.e., taper the FFT coefficients, i.e., apply windowing / tapering techniques.

A Time Domain Non-recursive filters

1. Running Mean, Box Car Filter

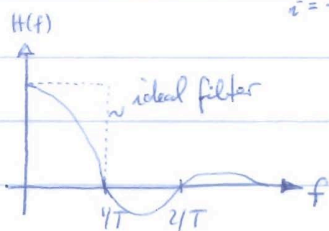
$$h_i = 1/(2N+1) \quad \text{if } 2N+1 \text{ is \# of weights}$$

i.e.,

$$y_t = \sum_{k=-N}^N h_k x_{t-k} \quad \int_{-\infty}^{\infty} h(\tau) x(t-\tau) d\tau$$

T is the filter length

$$\sum_{i=-N}^N h_i = 1$$



kills off variance at $1/T, 2/T, 3/T$ etc.
but lets through at $1.5/T, 2.5/T, 3.5/T$ etc.

(2) Godin Filter

one of the first digital filters
used to remove tides

3 box cars applied successively with

1. $T = 25$ hrs $h_1(t)$

2. $T = 25$ hrs $h_2(t)$

3. $T = 24$ hrs $h_3(t)$

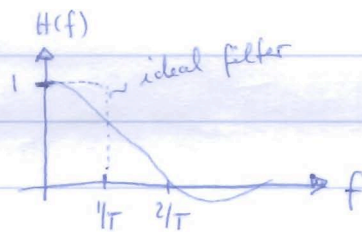
length

$$F(h_1(t) * h_2(t) * h_3(t)) = H_1(f) \cdot H_2(f) \cdot H_3(f)$$

real smooth response, too smooth as it also removes frequencies
~~below~~ below the cut-off

Full

(3) Raised Cosine

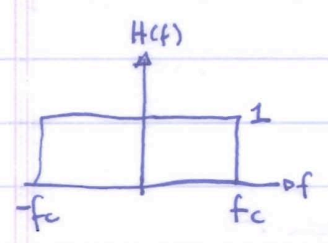


$$h_n = \frac{1}{2N} \left[\frac{1}{2} \left(1 + \cos \left(\frac{2\pi n}{2N+1} \right) \right) \right]$$

$$n = -N, \dots, +N$$

Lanczos filter

- attempt to mimic the "ideal" filter with a few ~~low~~ sine waves



$$H(f, f_c) = \begin{cases} 1 & f \leq f_c \\ 0 & f > f_c \end{cases}$$

inverse Fourier transform gives the filter weights

$$h(t = n \Delta t) = \int_{-\infty}^{\infty} H(f) e^{j2\pi f n \Delta t} df =$$

$$\begin{cases} \mathcal{F}(\Pi(t, T)) = T \text{sinc}(fT) \\ \mathcal{F}^{-1}(\Pi(f, f_c)) = f_c \text{sinc}(f_c t) \end{cases}$$

$$= 2 f_c \text{sinc}(2\pi f_c n \Delta t)$$

this is from the Lanczos filter program

$2\pi f_c n \Delta t$
length of window in frequency
 $\approx 2f_c$ sinc

infinite # of filter weights here
→ truncation

- (1) $h_0 = 2 f_c / \Delta t$
- (2) $h_n = h_{-n}$
- (3) need to normalize filter weights
- (4) finite number of filter weights: $2S + 1$

given ~~$h_{-s}, h_{-s+1}, \dots, h_0, h_1, \dots, h_s$~~ $h_{-s}, h_{-s+1}, \dots, h_0, h_1, \dots, h_s$

what is $\tilde{F}(h) = \hat{H}$

symmetric, $H(f)$ is real (no imaginary part, no phase shift by design)

$$\hat{H}(f) = \sum_{n=-s}^s h_n e^{-j2\pi f n \Delta t} = h_0 + 2 \sum_{n=1}^s h_n \cos(2\pi f n \Delta t)$$

$\hat{H}(f)$ is the approximation of the step fct by a limited # of cosine fct.

→ Gibbs's phenomena