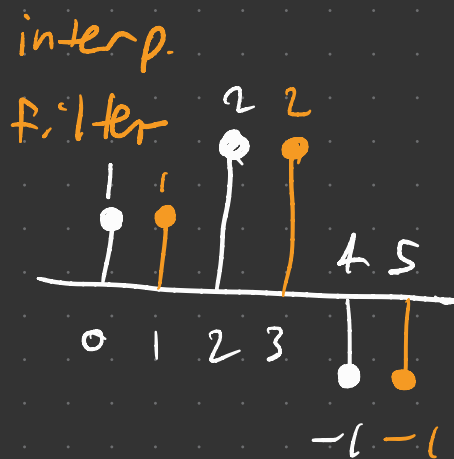
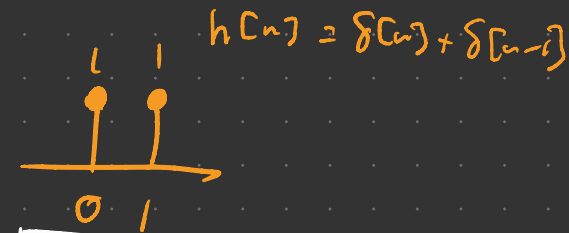
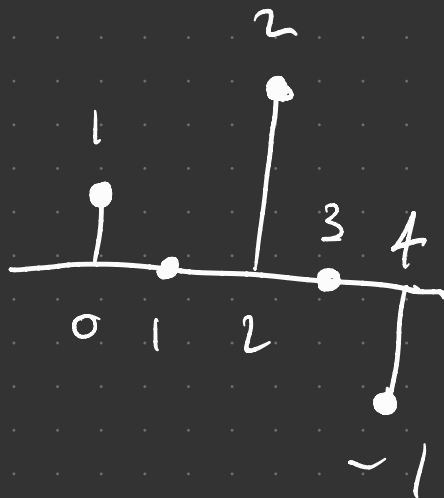
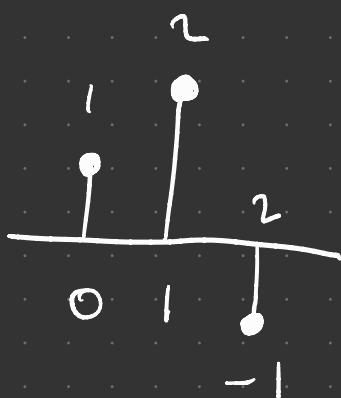


Last Time: Multirate Systems

Ex:

$$u[n] = \begin{cases} v[\frac{n}{2}], & \text{even } n \\ 0, & \text{odd } n \end{cases}$$



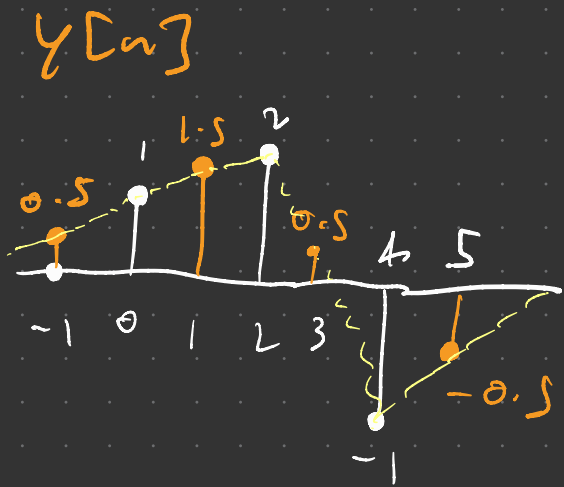
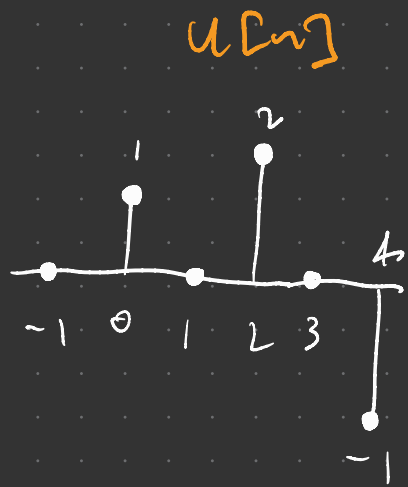
Q: Why is this bad?

A: Blocking artifacts.

(Imagine upsampling by factor 8)

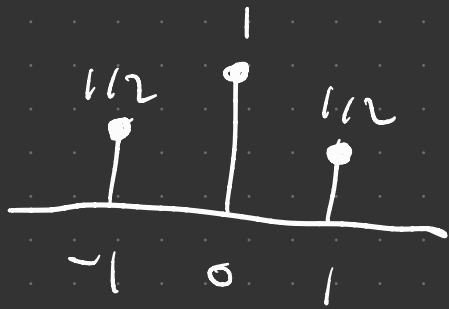
Remark: Interpolation filter design is a kind of art.

Q: What about linear interpolation?



Exercise: what is $h[n]$?

Solⁿ:



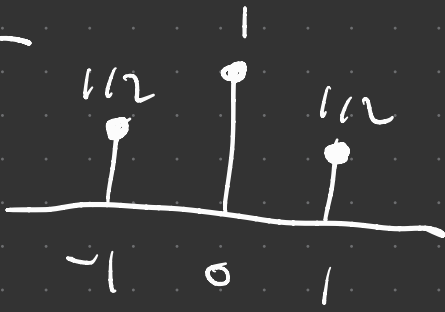
$$y[n] = \frac{u[n+1] + u[n-1]}{2} + u[n]$$

$$h[n] = \frac{1}{2} \delta[n+1] + \delta[n] + \frac{1}{2} \delta[n-1]$$

Q: why is this bad?

A: Looks too sharp. We want something that looks smooth.

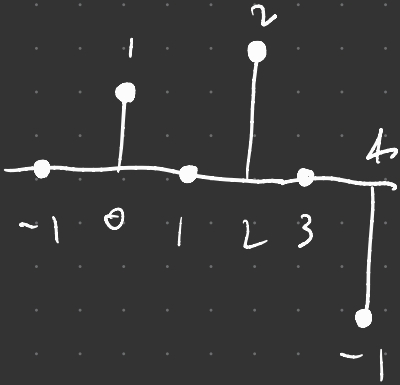
Obs:



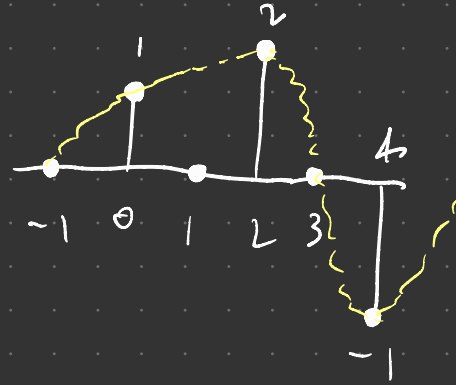
looks kinda like
a sinc \Rightarrow low-pass

Goal:

$u[n]$

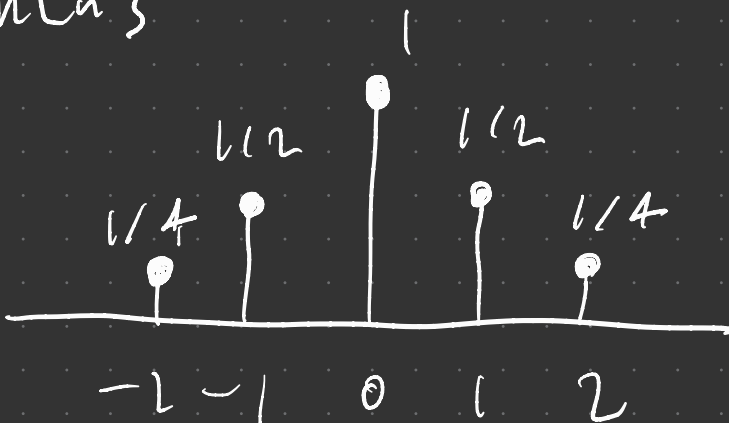


$y[n]$



Exercise: What about the filter

$h[n]$



?

Obs: It's bad because it changes the original samples!

check what happens @ $y[0]$

$$y[0] = 1 + 0 + \frac{1}{4}(2) = \frac{3}{2} \neq u[0]$$

we destroyed the original samples!

Goal: Interpolate and preserve samples.

Obs: The issue is at even coeff. of $h[n]$.

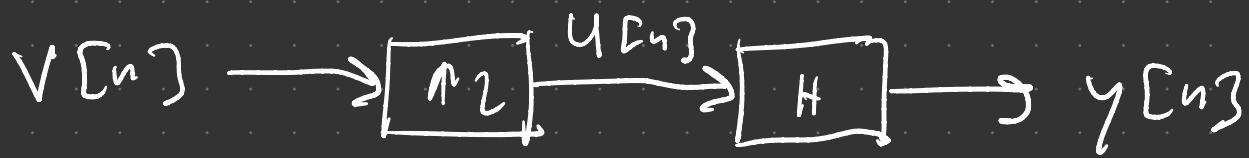
Property: An interpolation filter must satisfy

$$h[2n] = \delta[n] \quad \rightarrow \quad (\downarrow 2)h[n]$$

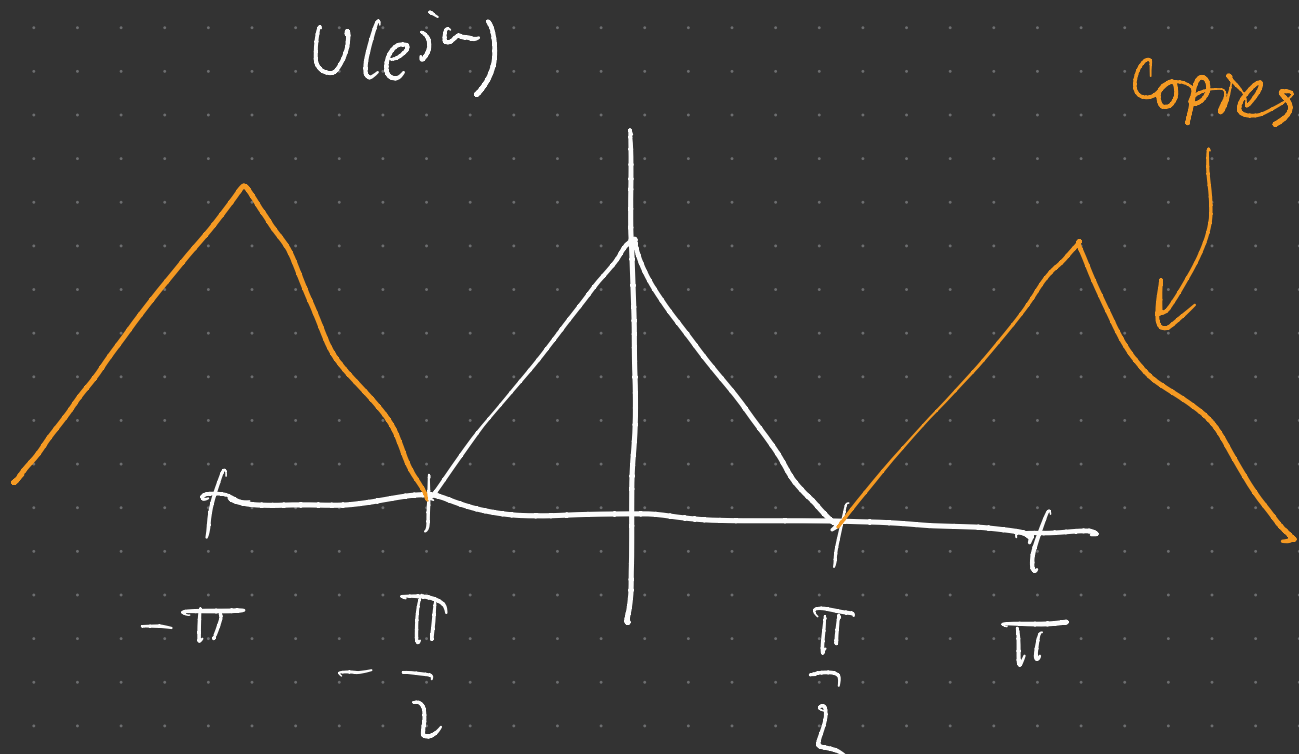
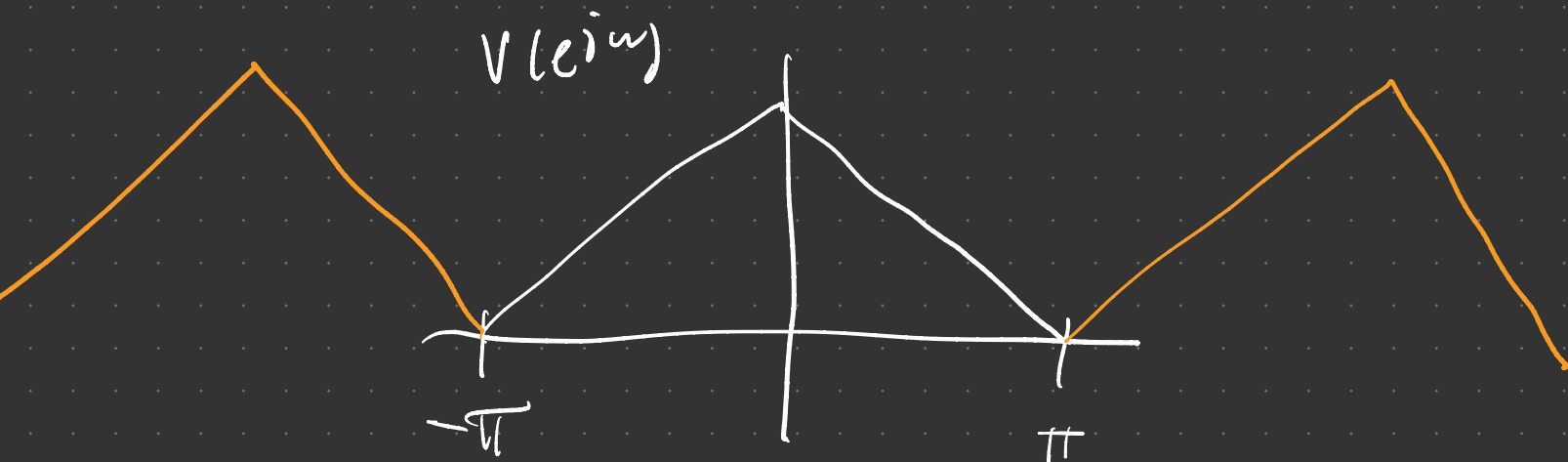
half-band condition

Obs: Interpolation filter design is only about the odd coeff.

Frequency - Domain Analysis



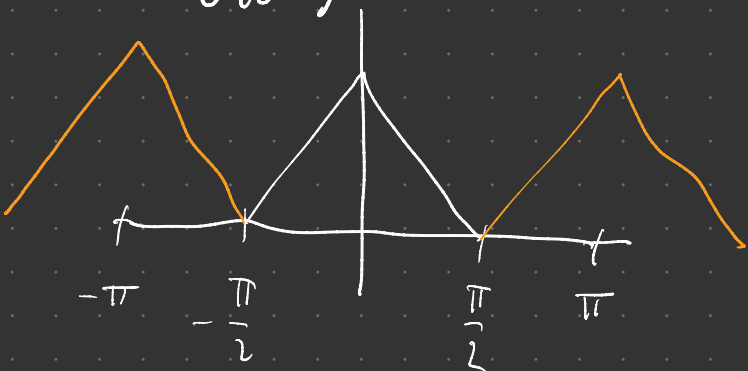
$$U(z) = V(z^2) \Rightarrow U(e^{j\omega}) = V(e^{j2\omega})$$



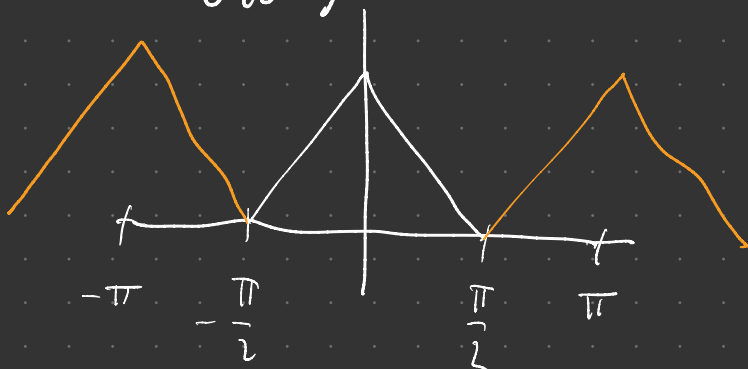
Remark: We don't want the spectrum to look like this!

Two Options:

$U(e^{i\omega})$

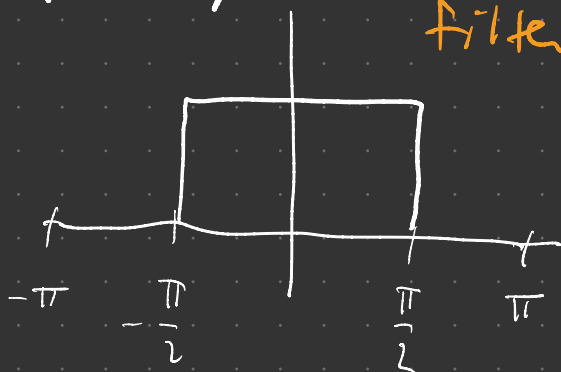


$U(e^{i\omega})$



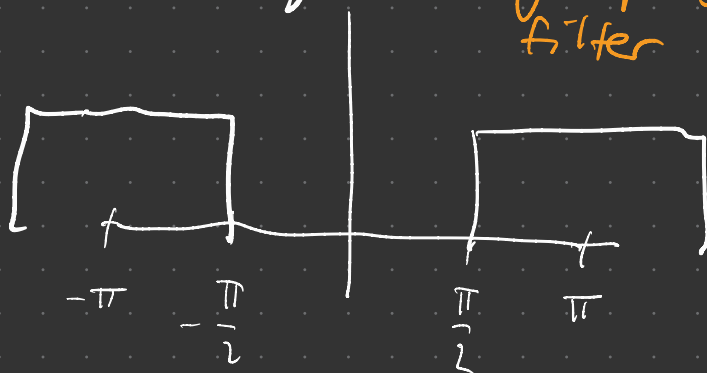
$H(e^{i\omega})$

low-pass filter



$G(e^{i\omega})$

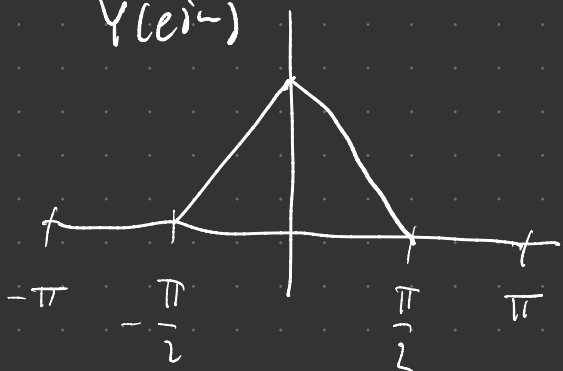
high-pass filter



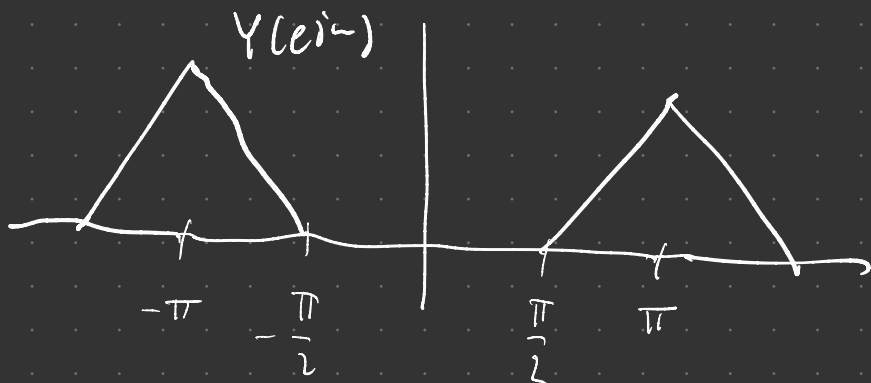
||

||

$Y(e^{i\omega})$



$Y(e^{i\omega})$



Low-pass Interp.

High-pass Interp.

We have already seen low-pass interpolation.

Q: What is high-pass interpolation?

Q: Given a low-pass filter, is there a corresponding high-pass filter?

A: Yes.

$$G(e^{j\omega}) = H(e^{j(\omega-\pi)})$$

Shift in freq.

$$g[n] = e^{j\pi n} h[n]$$

modulation in time

$$= (-1)^n h[n]$$

Obs: If $h[n]$ satisfies half band condition, then so does $g[n]$. (Even index stays the same).

Trick: Flip every other coeff of $h[n]$ to get $g[n]$.

Exercise: Given the linear interp-filter

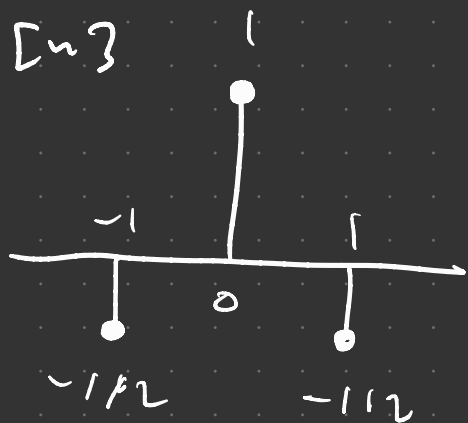
$h[n]$



What is the corresponding high-pass filter $g[n]$?

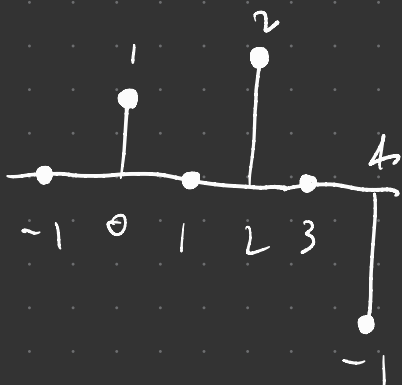
Soln:

$g[n]$

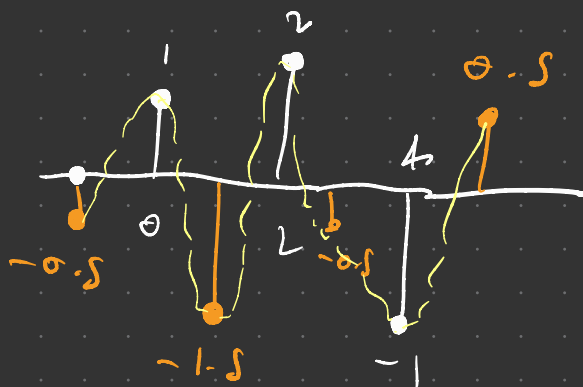


really wiggly
 \Rightarrow high-pass

$u[n]$



$y[n]$



Obs: Original samples are preserved because $g[n]$ satisfies the half-band condition.

Q: Is upsampling linear?

A: Yes

$\alpha, \beta \in \mathbb{R}$

Proof
(12) $(\alpha V_1[n] + \beta V_2[n])$

$$= \begin{cases} \alpha V_1[\frac{n}{2}] + \beta V_2[\frac{n}{2}], & n \text{ even} \\ 0, & n \text{ odd} \end{cases}$$

$$= \alpha \begin{cases} V_1[\frac{n}{2}], & n \text{ even} \\ 0, & n \text{ odd} \end{cases} + \beta \begin{cases} V_2[\frac{n}{2}], & n \text{ even} \\ 0, & n \text{ odd} \end{cases}$$

$$= \alpha (12) V_1[n] + \beta (12) V_2[n]$$

□

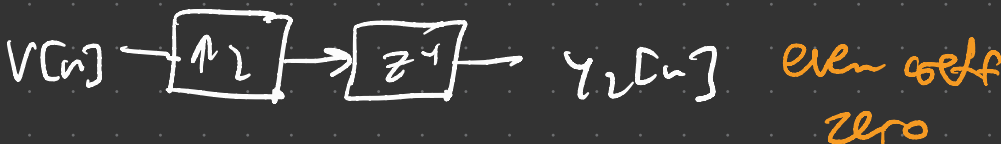
Q: Is upsampling time-invariant?

A: No.

Proof: After upsampling, the signal always has zero for odd coeff.



$$Y_1[n] \neq Y_2[n]$$

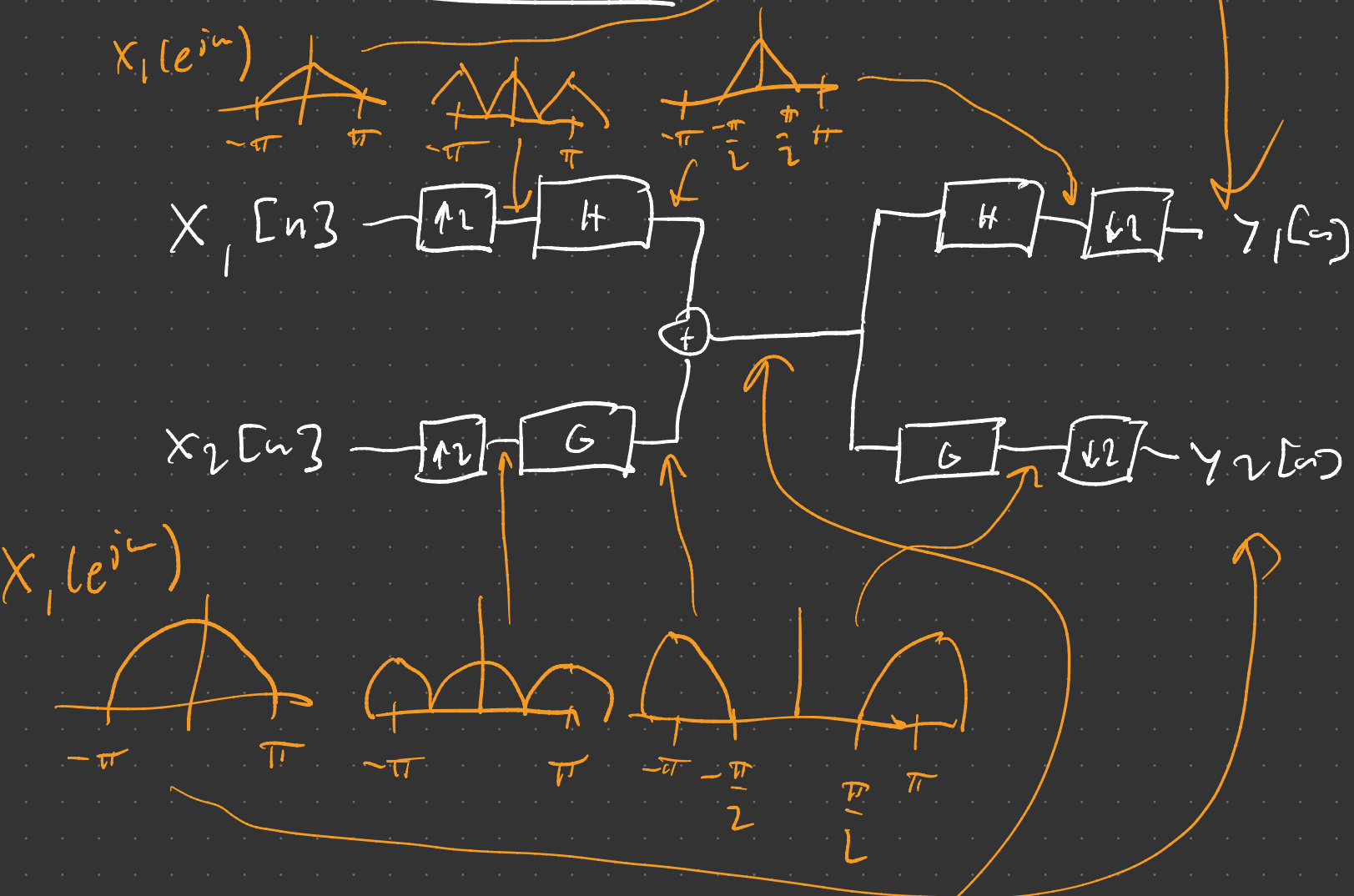


\Rightarrow not time-invariant

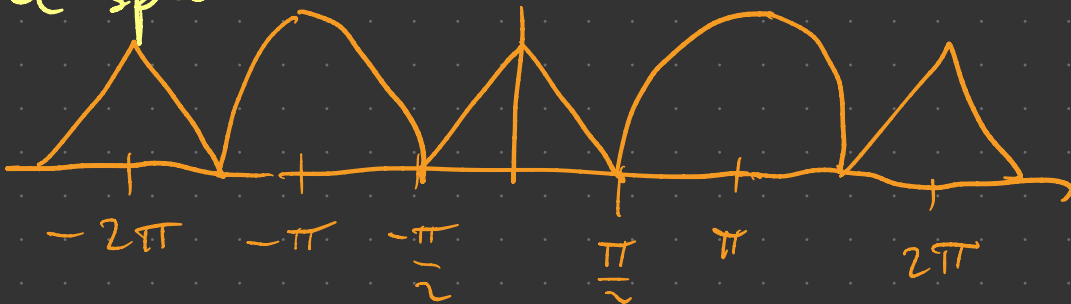
□

Remark: This is why multirate DSP is hard: Operations don't commute.

Multirate Filter bank



Multiplexed
the spectrum



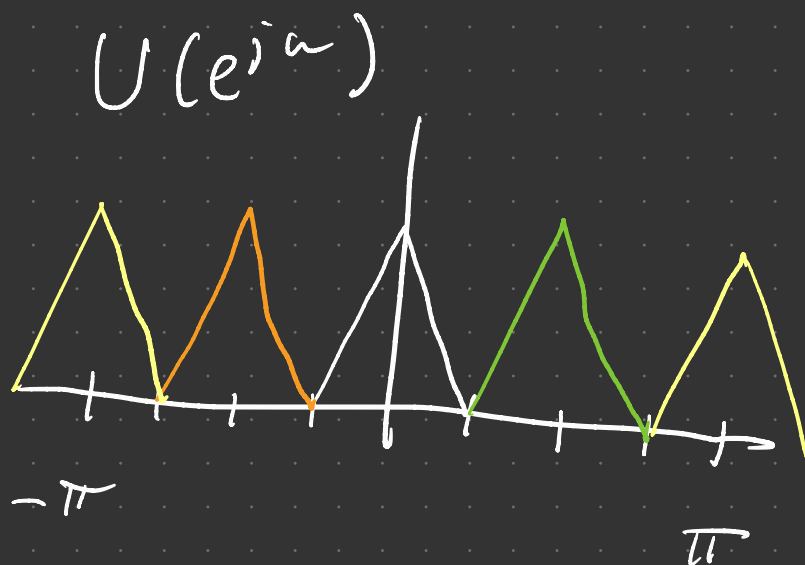
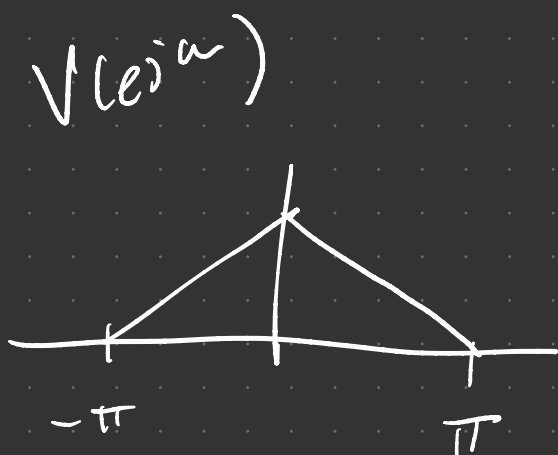
General Case of Upsampling $\frac{n}{L} \in \mathbb{Z}$

$v[n]$ \rightarrow $\boxed{\uparrow L}$ \rightarrow $u[n] = \begin{cases} v[\frac{n}{L}], & n \text{ is mult. of } L \\ 0, & \text{else} \end{cases}$

insert $(L-1)$
zeros between
samples

$$U(z) = V(z^L) \Rightarrow U(e^{j\omega}) = V(e^{jL\omega})$$

Ex: $L=4$



- 3 copies of the spectrum
- 4 total spectrums

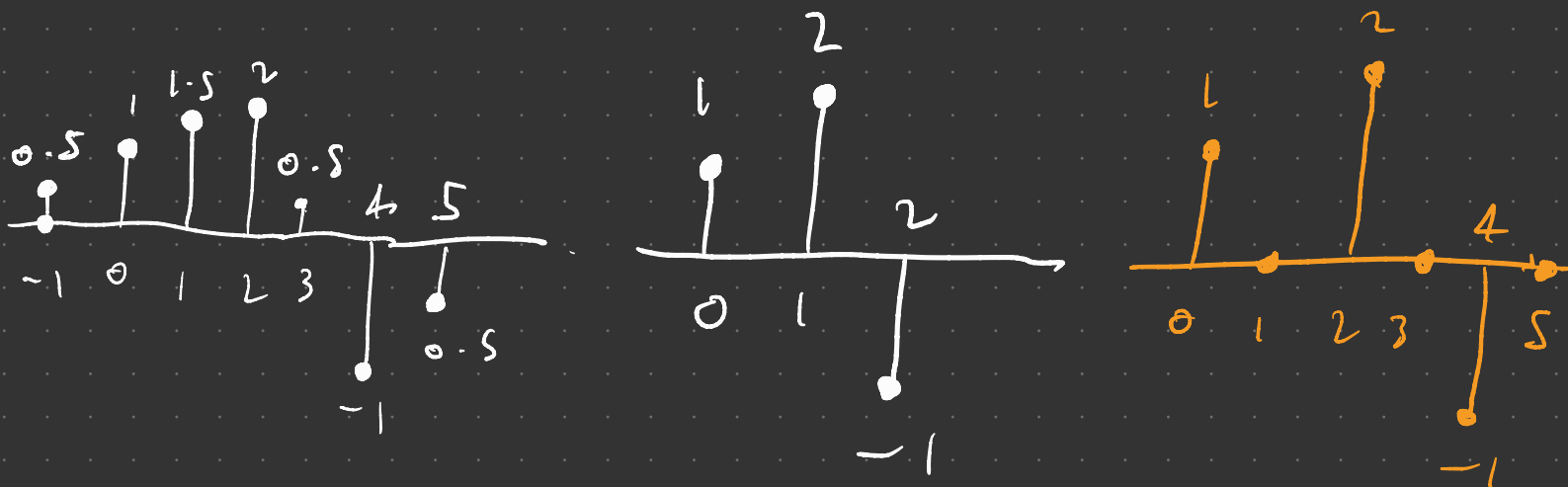
Obs: 4 options for the interp. filter

General: L interp. filters for the L spectrums

Down sampling by Factor 2

Q: What is down sampling?

A: Deleting samples



Q: Is down sampling linear?

A: Yes

Q: Is it time-invariant?

A: No

Q: Is it invertible?

A: No.

Q: How are the spectrums $X(e^{j\omega})$ and $V(e^{j\omega})$ related?

A: $U(e^{j\omega}) = V(e^{j2\omega})$

How are $u[n]$ and $x[n]$ related?

$$x[n] : x[0] \quad x[1] \quad x[2] \quad x[3] \quad x[4] \dots$$

$$+ (-1)^n x[n] : x[0] \quad -x[1] \quad x[2] \quad -x[3] \quad x[4] \dots$$

$$2u[n] : 2x[0] \quad 0 \quad 2x[2] \quad 0 \quad 2x[4] \dots$$

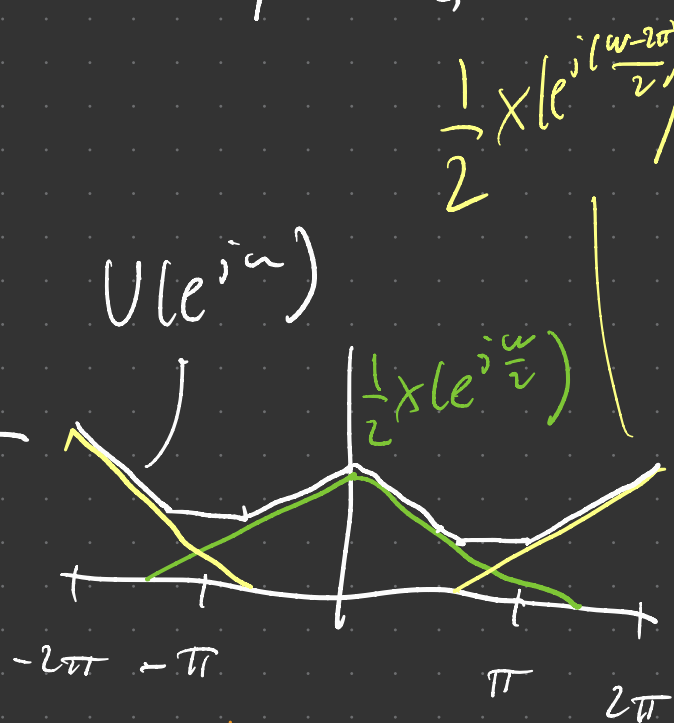
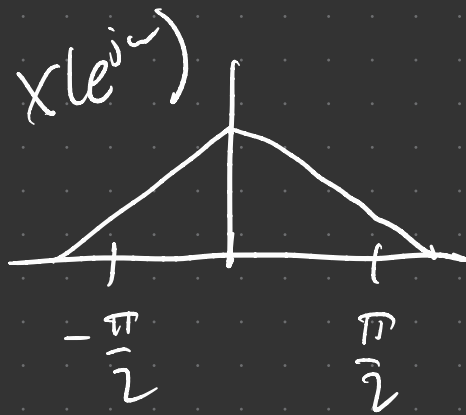
$$\Rightarrow u[n] = \frac{1}{2} (x[n] + (-1)^n x[n])$$

$$\Rightarrow U(e^{j\omega}) = \frac{1}{2} (X(e^{j\omega}) + X(e^{j(\omega-\pi)}))$$

$$\Rightarrow V(e^{j\omega}) = \frac{1}{2} (X(e^{j\frac{\omega}{2}}) + X(e^{j(\frac{\omega-2\pi}{2})}))$$

Obs: Downsampled spectrum is the avg. of original spectrum stretched by factor 2 and that same spectrum shifted by 2π .

Ex:



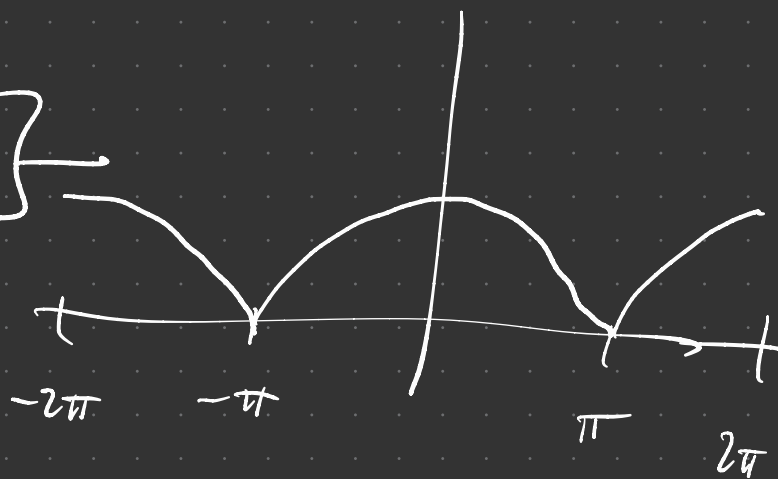
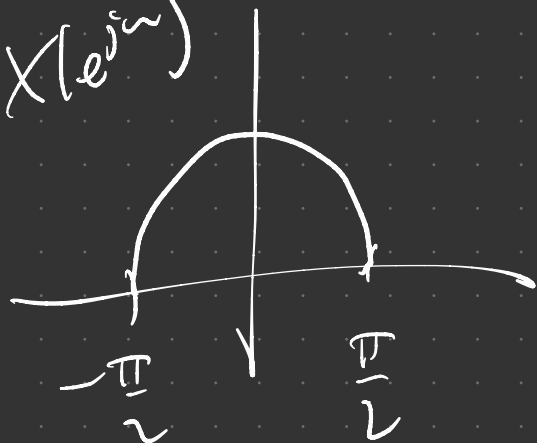
aliasing
destroyed
original
signal

Q: What conditions on $X(e^{j\omega})$ for no aliasing?

A: Band limited $[-\frac{\pi}{2}, \frac{\pi}{2}]$.

Ex:

$X(e^{j\omega})$



Remark: Downsampling is always preceded by filtering

Decimation



anti-aliasing
filter

Obs: Reverse of upsampling.