

Design and Application of a Hilbert Transformer in a Digital Receiver

SDR'11 - WinnComm

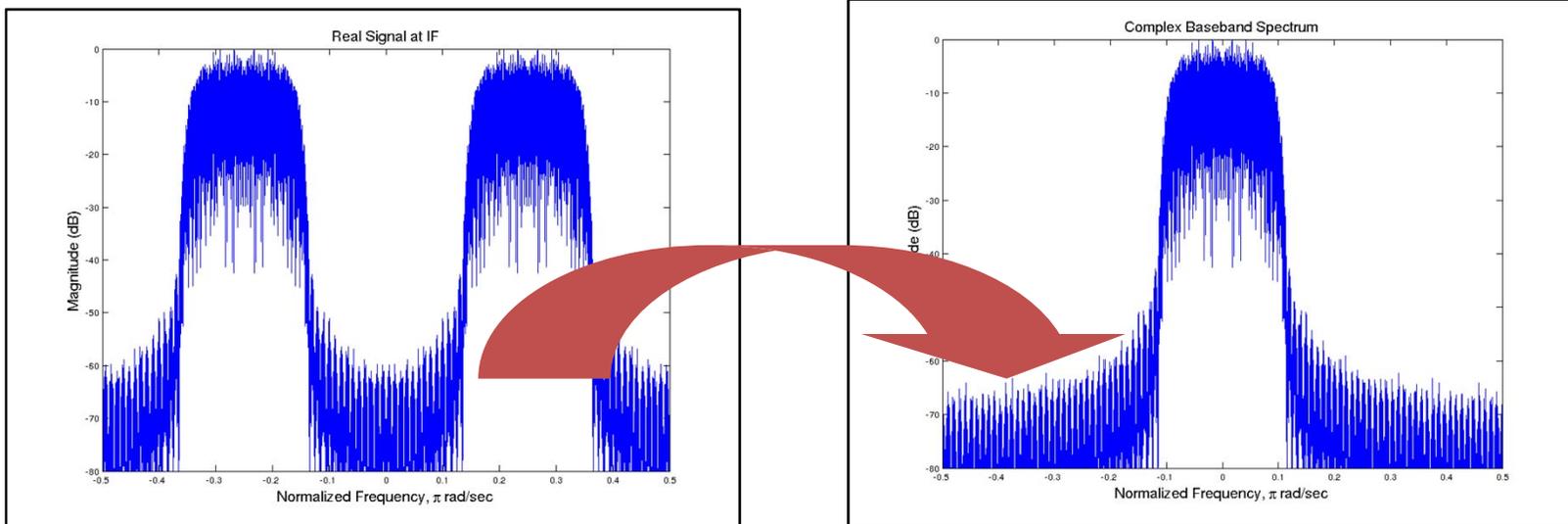
November 29, 2011

Matt Carrick



Motivation

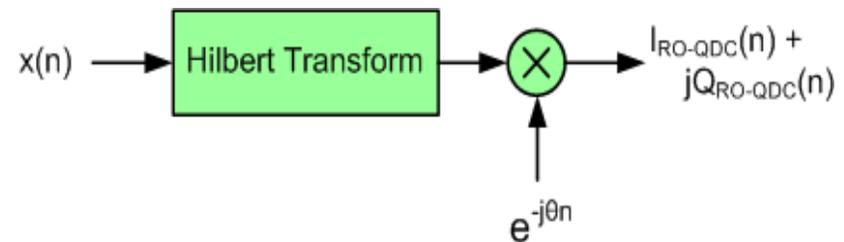
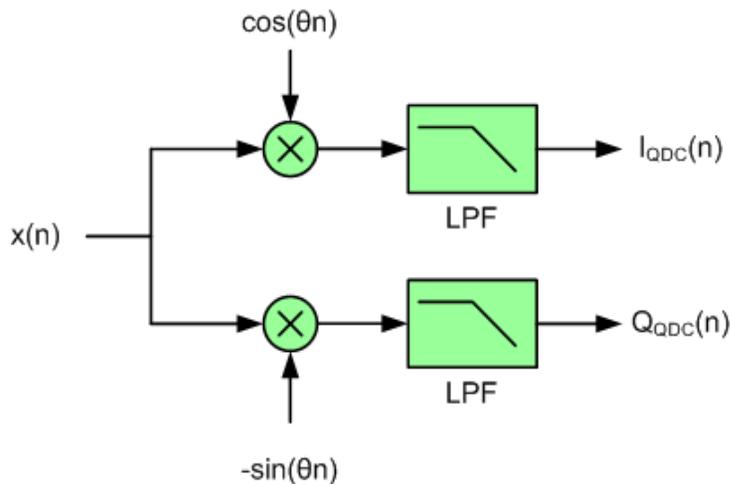
- Given real spectrum at arbitrary IF, how to get to complex baseband?
- Constraints:
 - Real A/D
 - Minimize Processing Power



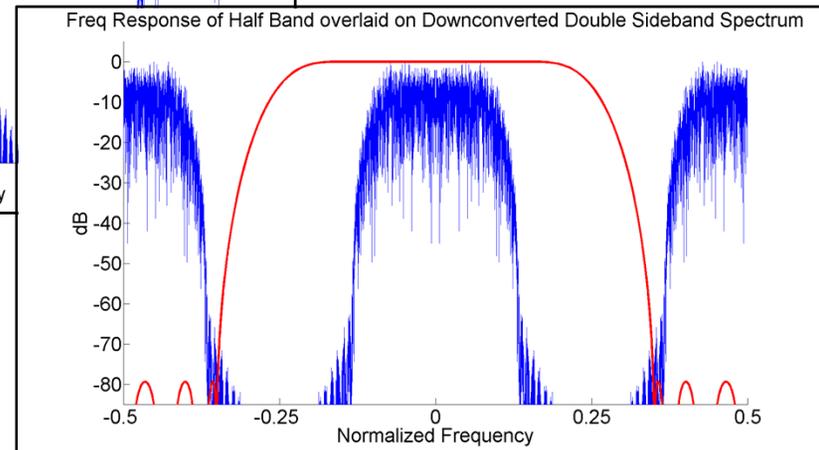
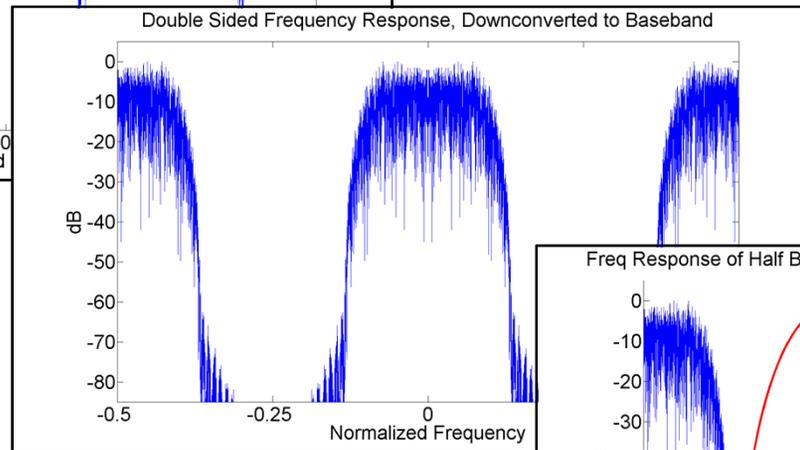
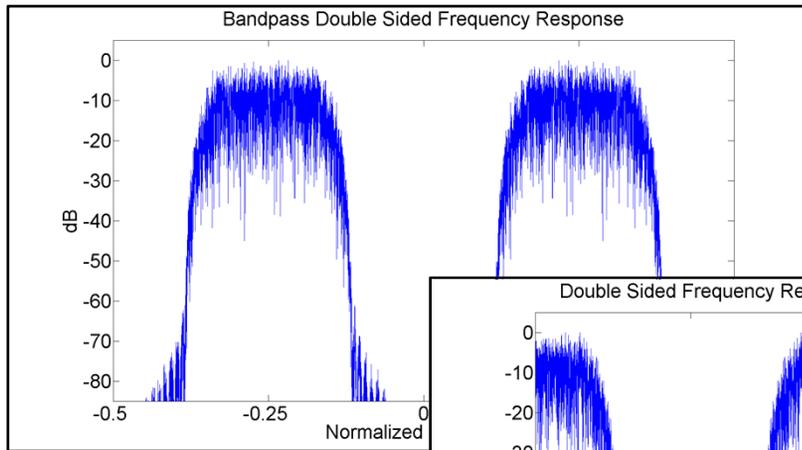
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- ***Comparison of quadrature downconverter, downconversion with Hilbert transformer***
 - Hilbert Transform Review
 - Hilbert Transform Filter Design Through Windowing
 - Hilbert Transform Filter Design in Frequency
 - Designing a Half Band Filter
 - Results
 - Implementation of Hilbert Transform Filter

Downconversion Options

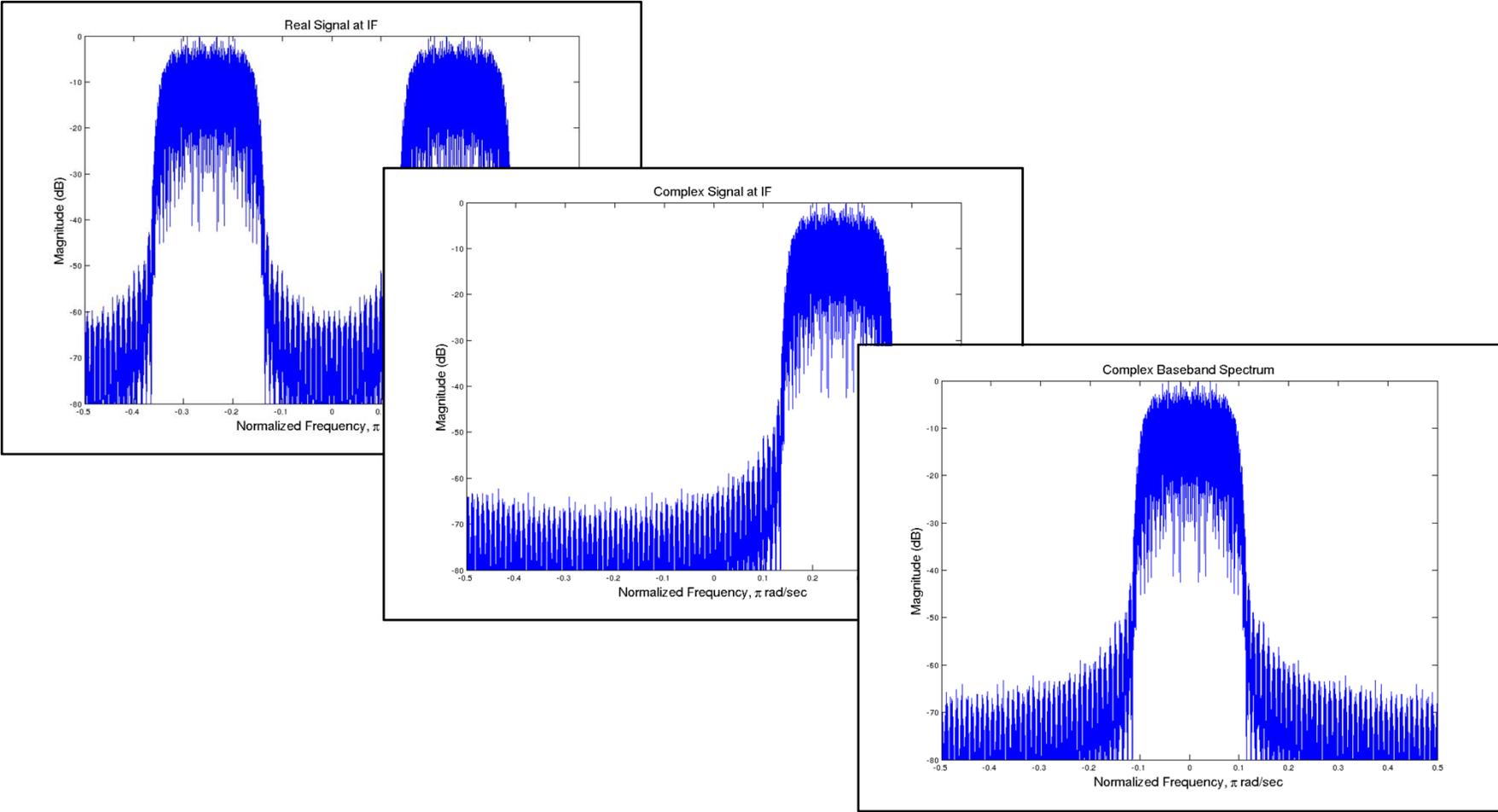
- Quadrature Downconverter
- Hilbert Transform + Heterodyne
- Other Options
 - Alias to baseband, Polyphase filter bank + FFT



Downconversion With Quadrature Downconverter

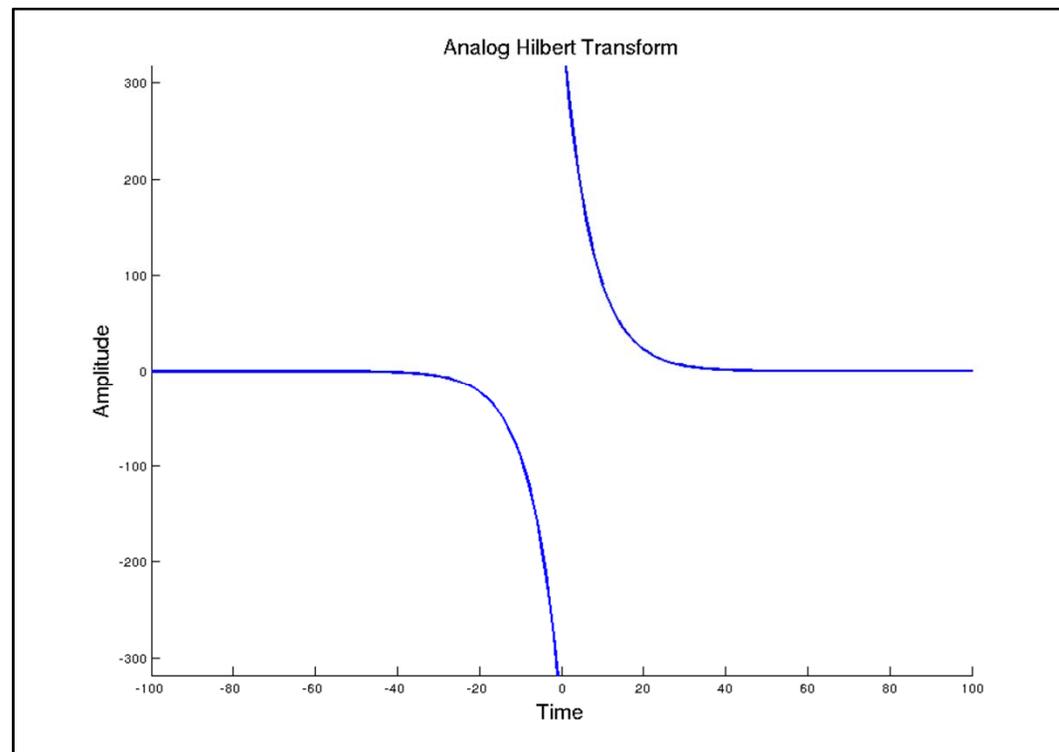


Downconversion With Hilbert Transform



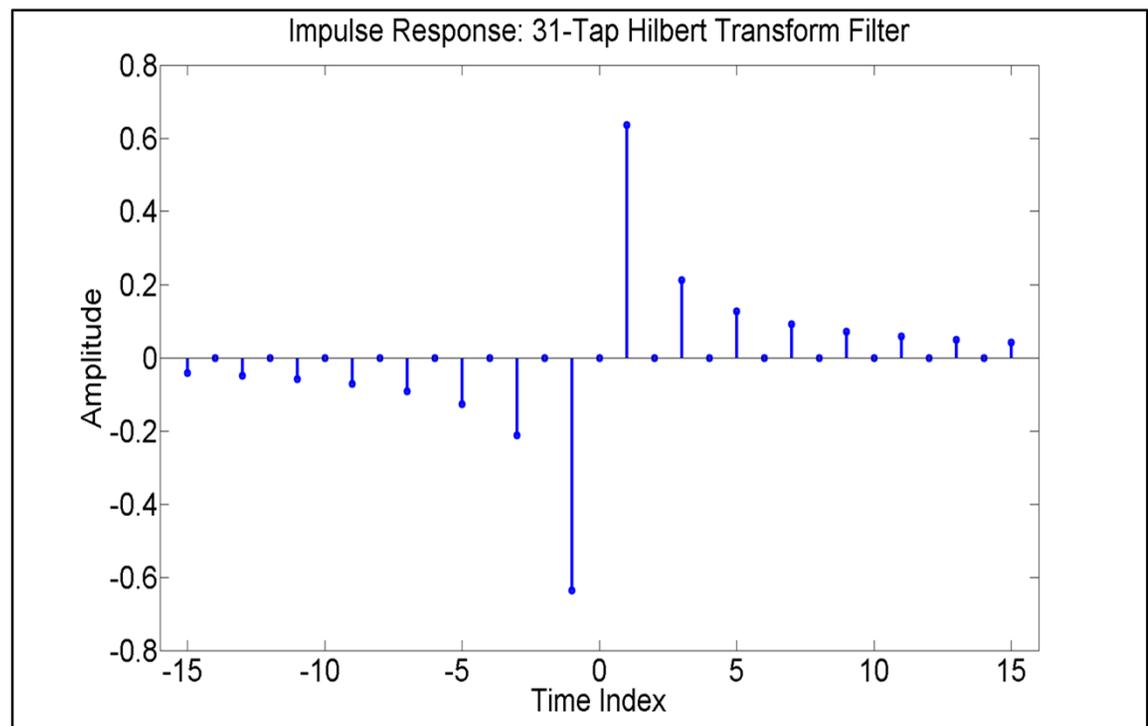
Hilbert Transform Review

- Convolutional Operator, Analog Representation
 - $x'(t) = x(t) * h(t)$
 - $h(t) = 1/\pi t$



Hilbert Transform Review (Con't)

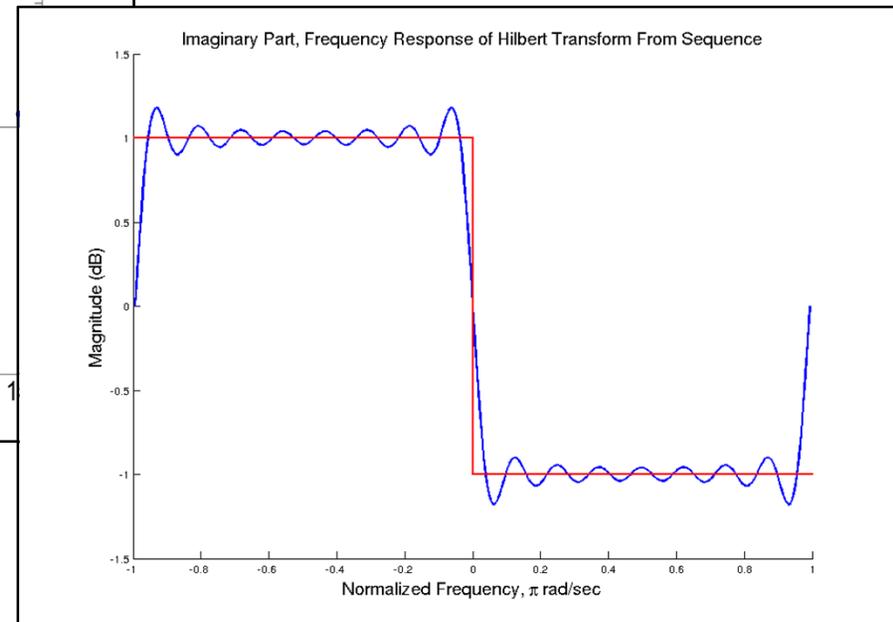
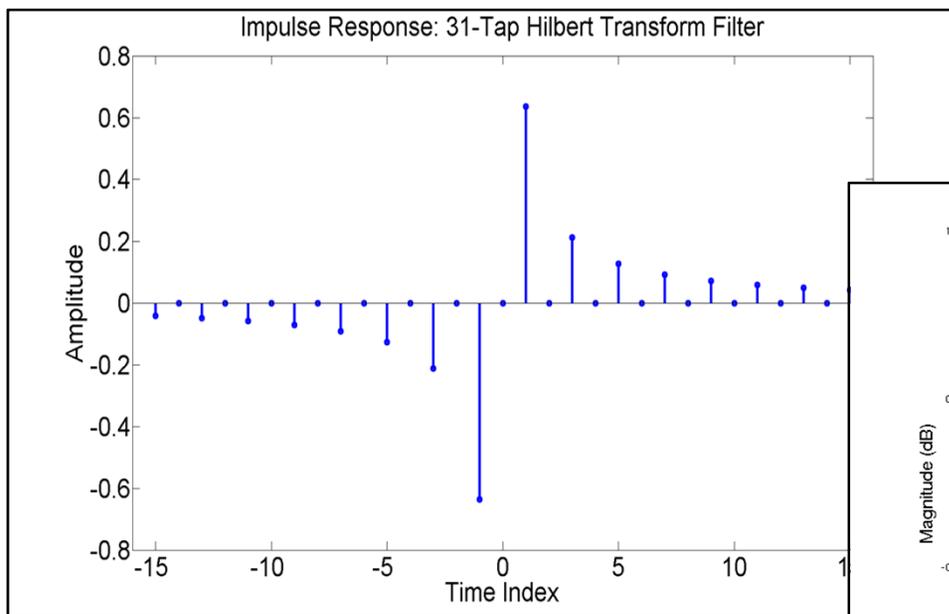
- Digital Representation
 - $x'[n] = x[n] * h[n]$
 - $h[n] = 2/(\pi n)$ for n odd
 - $h[n] = 0$ for n even
- Hilbert Transform
- Hilbert Transformer



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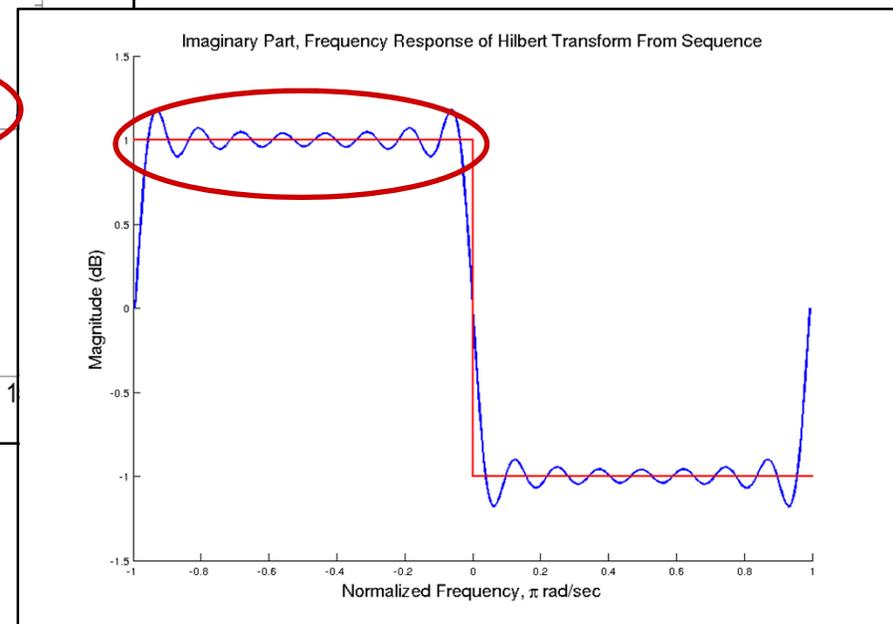
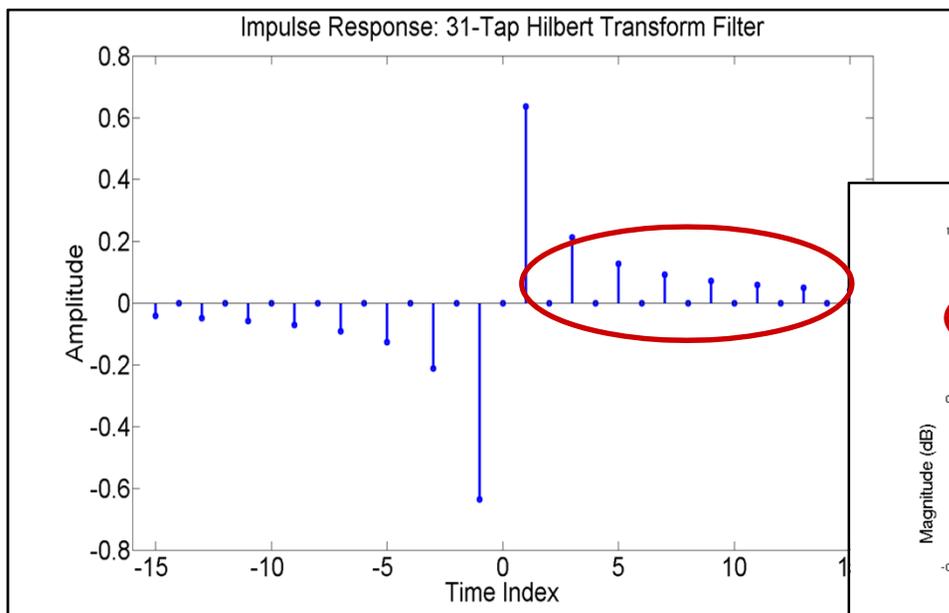
Building Filter from Discrete Sequence

- $h[n] = 2/(\pi n)$ for n even
- $h[n] = 0$ for n odd



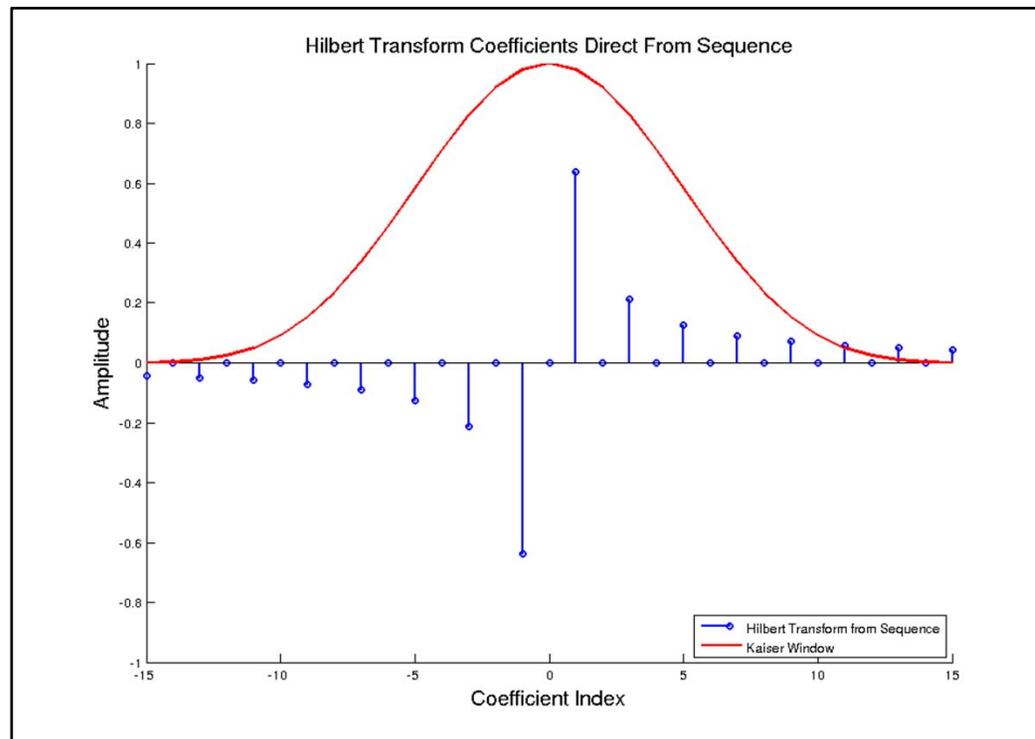
Reducing Ripple

- Ripple due to Gibbs' Phenomenon
- Window coefficients to combat ripple

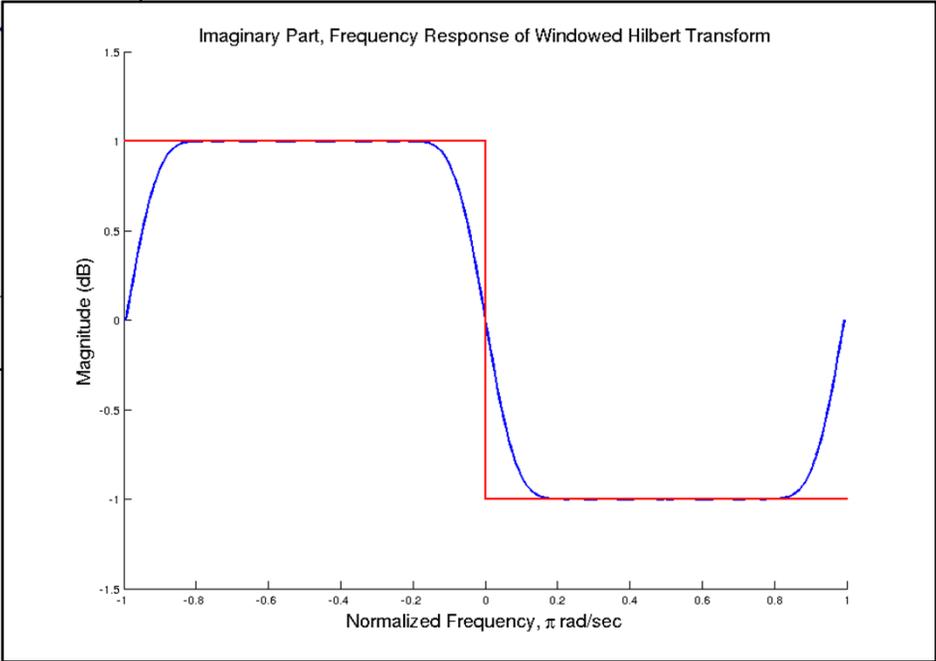
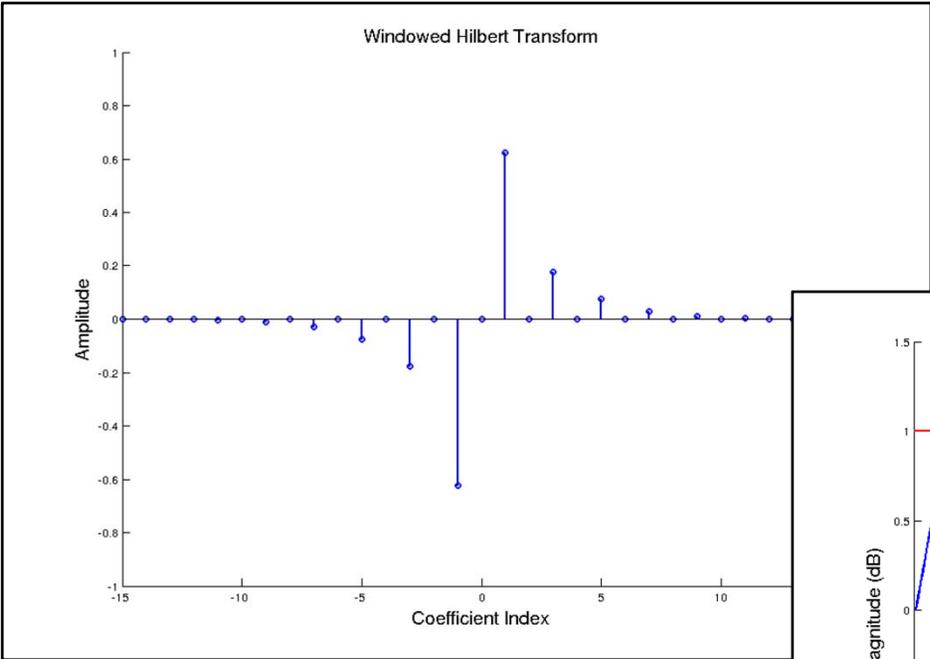


Reducing Ripple (Con't)

- Force tails of filter to zero artificially through windowing

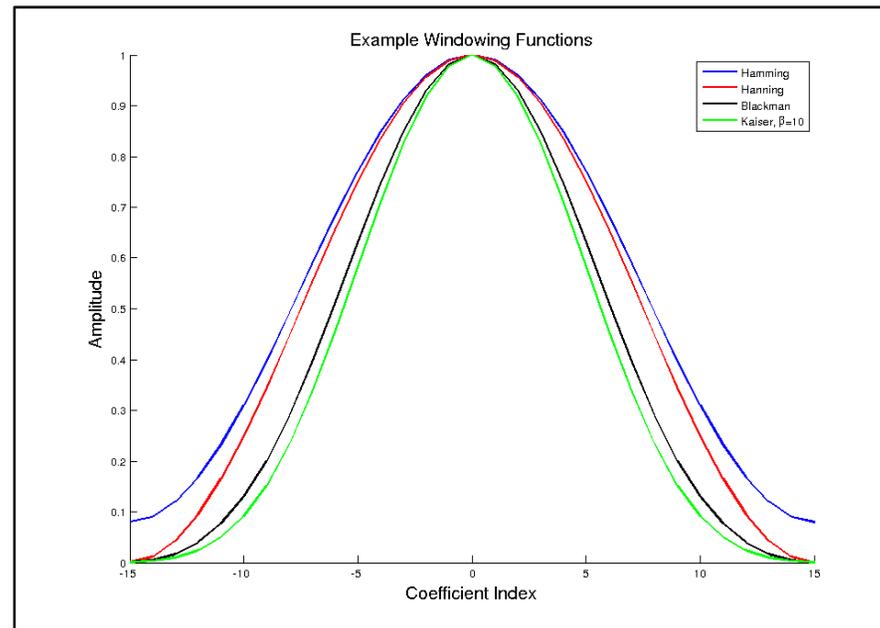


Reducing Ripple (Con't)



Change Design Method

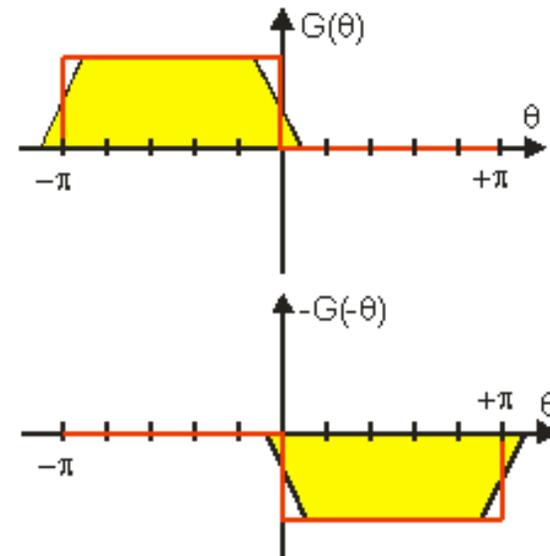
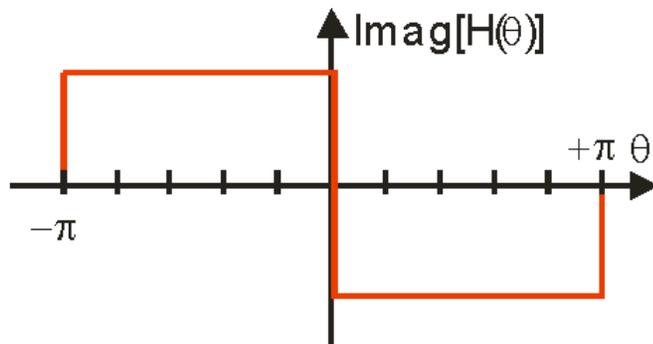
- Choosing 'best' window is difficult
- Instead of designing in time, design in frequency



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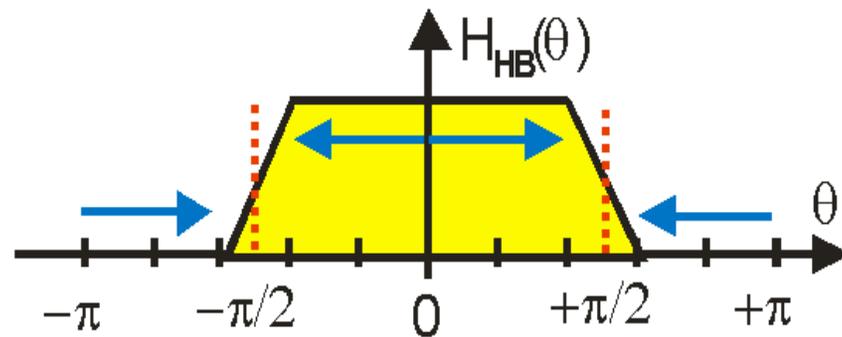
Hilbert Transform Frequency Response

- By definition:
 - $H(\omega) = -j \operatorname{sgn}(\omega)$
 - Approximate with two half band filters
- How to build a half band filter?



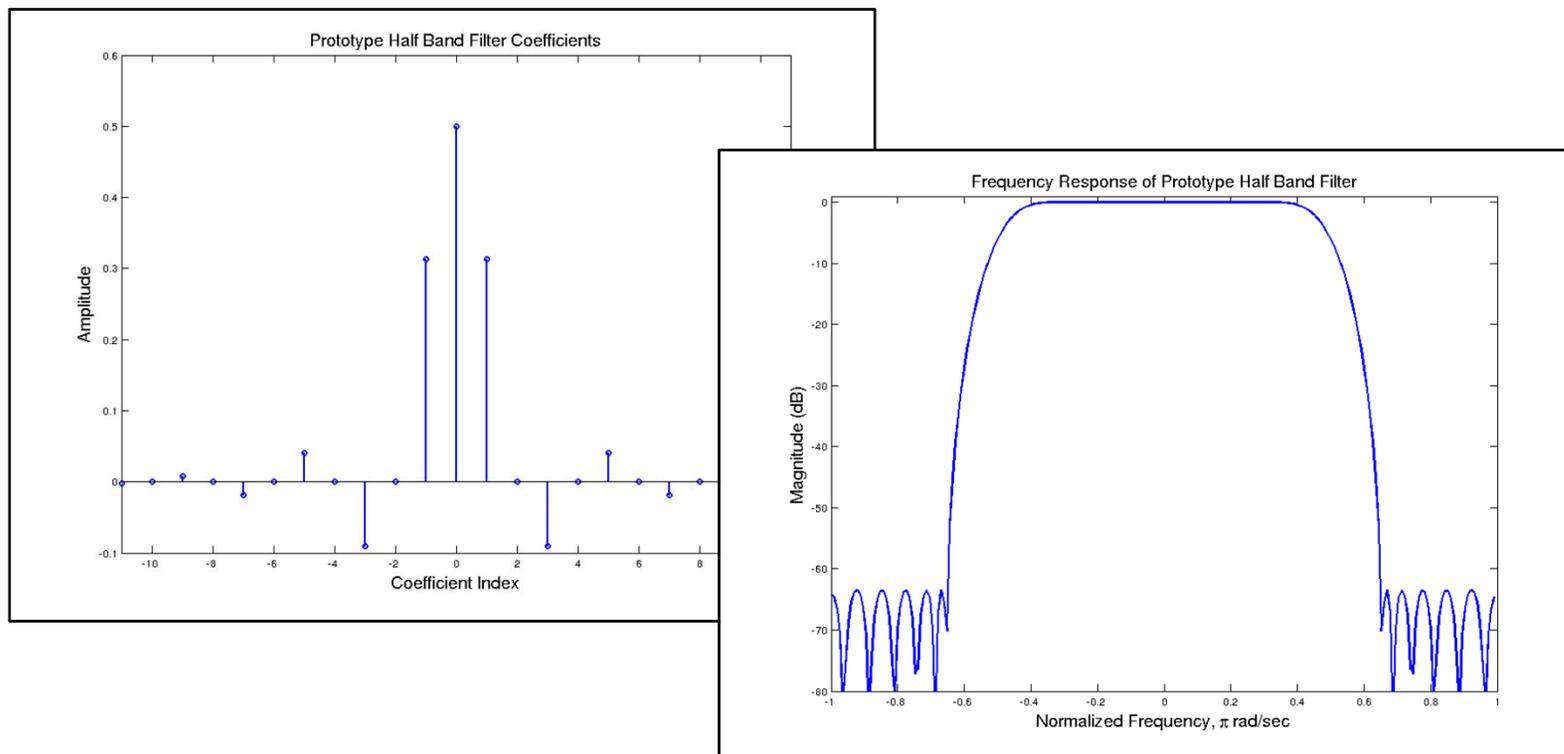
Half Band Filter Design

- A half band filter, filters half the spectrum
- Every other coefficient is zero
- Quick design method (MATLAB code);
 - $f = [0 \text{ } w_c \text{ } 1-w_c \text{ } 1];$
 - $a = [1 \text{ } 1 \text{ } 0 \text{ } 0];$
 - $hb = \text{firpm}(N-1, f, a);$



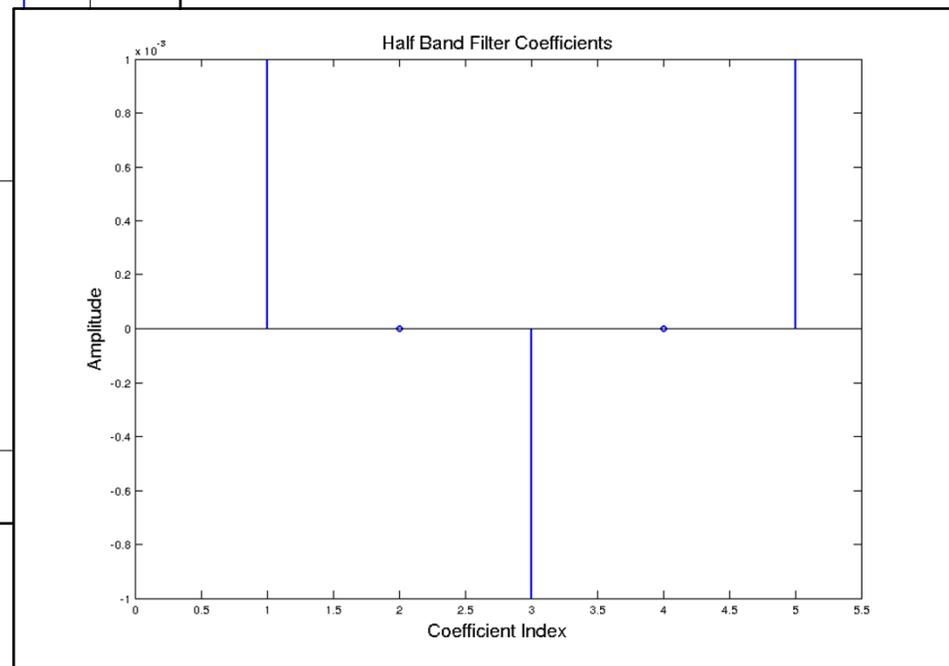
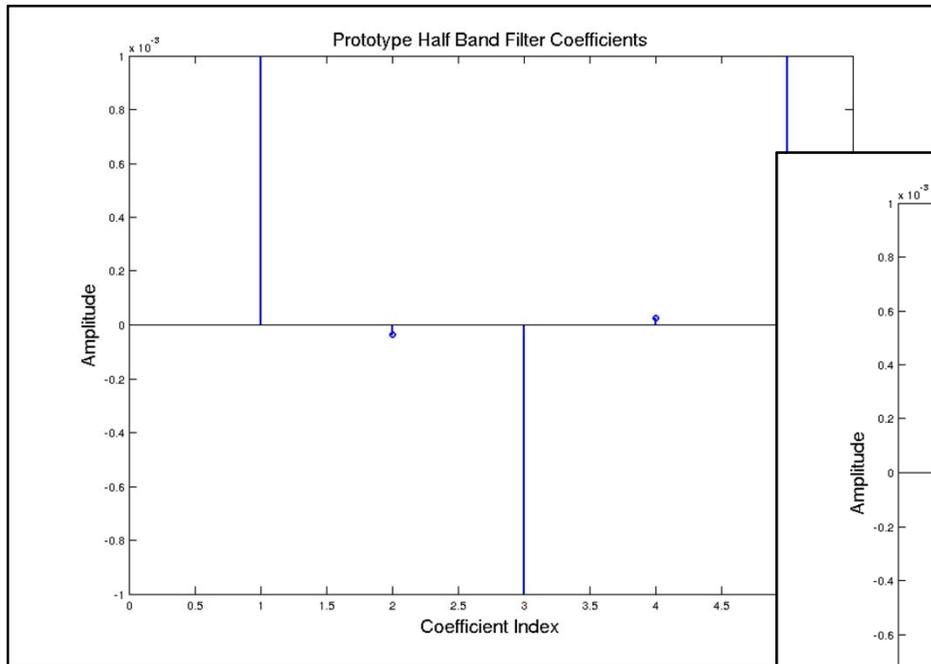
Half Band Filter Design (Con't)

- Coefficients have 'zeros' every other sample
- Frequency response covers appropriate band



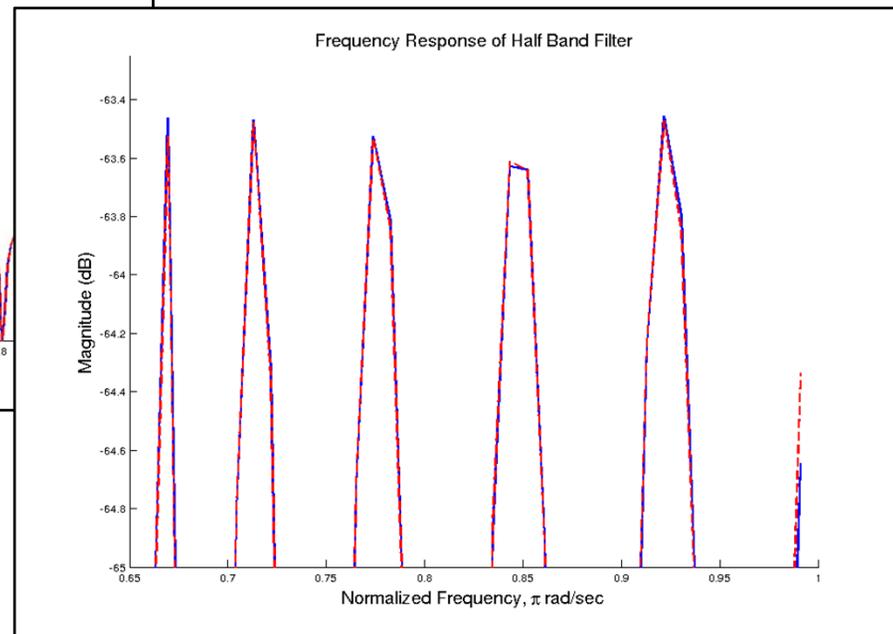
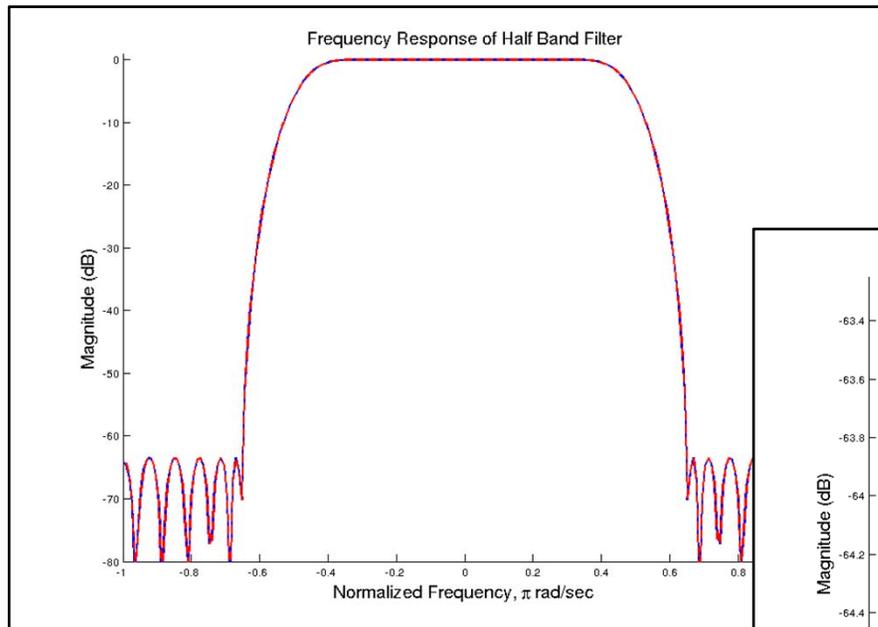
Half Band Filter Design (Con't)

- Parks-McClellan doesn't set zero coefficients to exactly zero
- Force coefficients to zero



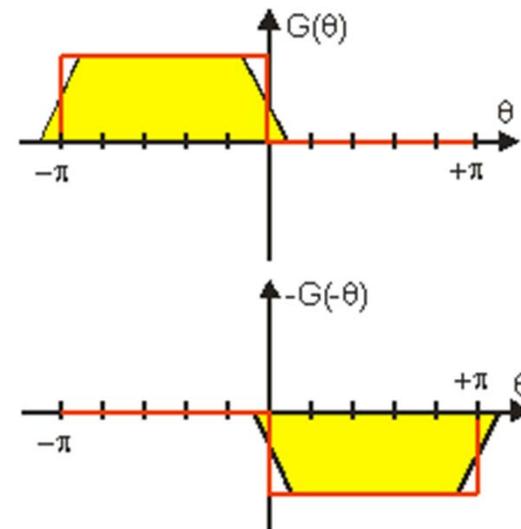
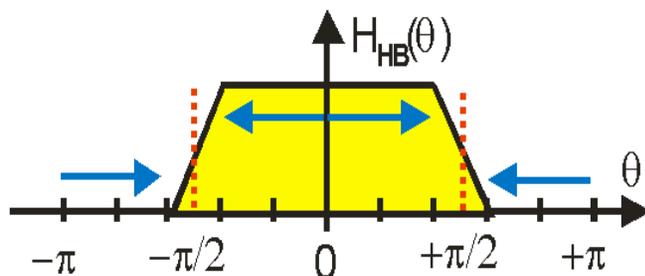
Half Band Filter Design (Con't)

- Change in frequency response is negligible



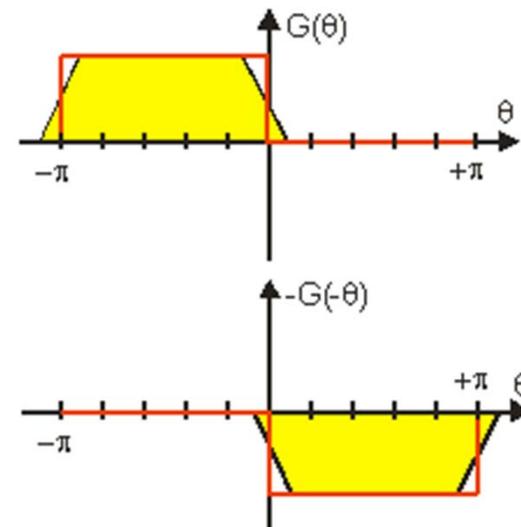
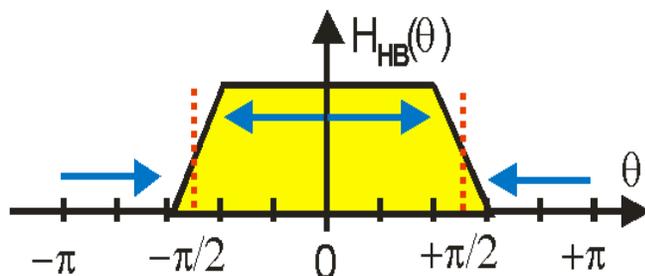
Sum Half Band Filters

- $G(\theta) = H_B(\theta - \pi/2)$
- $-G(-\theta) = -H_B(\theta + \pi/2)$
- $H_{HT}(\theta) = j (G(\theta) + G(-\theta))$
- $H_{HT}(\theta) = j (H_{HB}(\theta - \pi/2) - H_{HB}(\theta + \pi/2))$



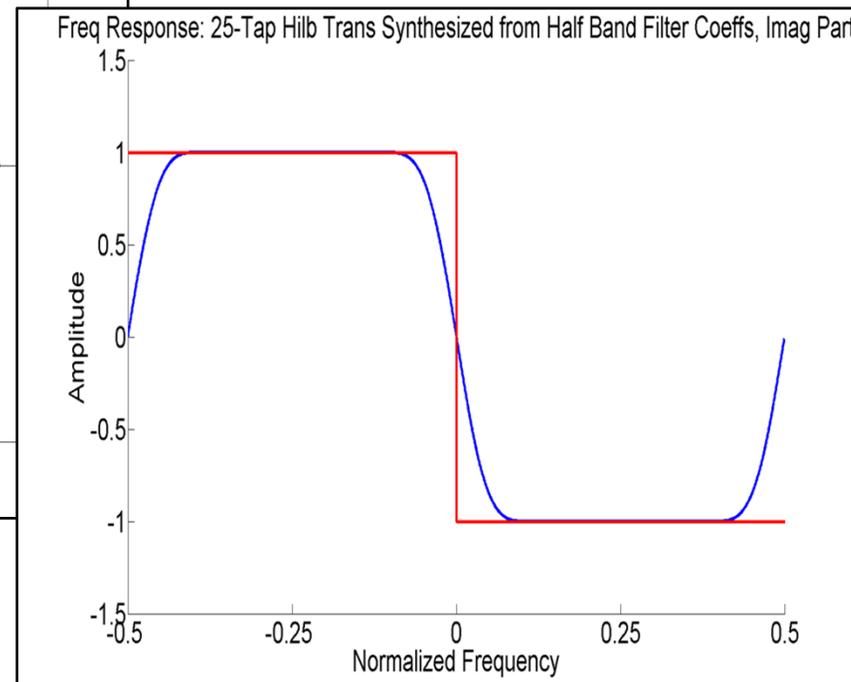
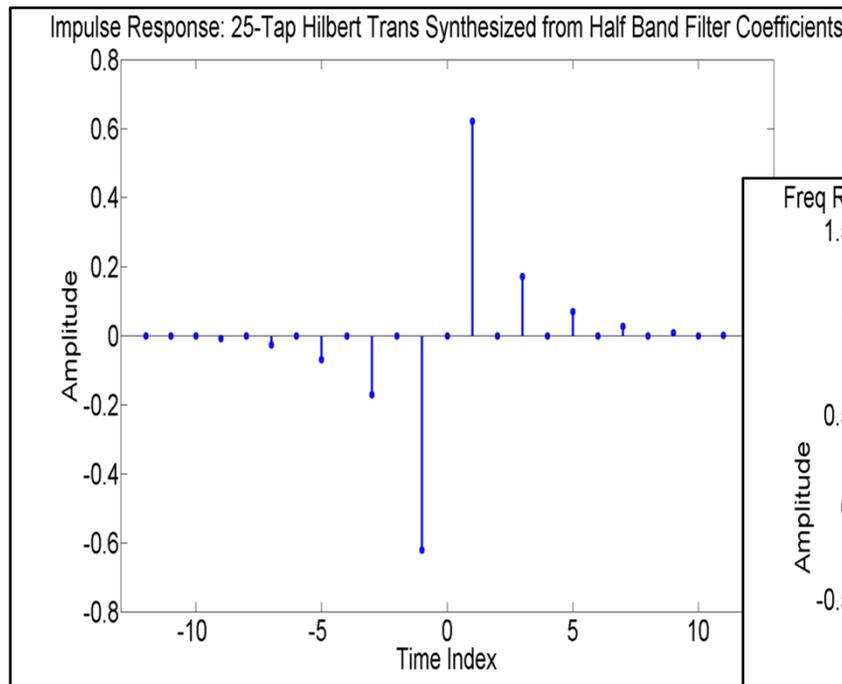
Sum Half Band Filters (Con't)

- $H_{HT}(\theta) = j (H_{HB}(\theta - \pi/2) - H_{HB}(\theta + \pi/2))$
- $H_{HB}(\theta - \pi/2) \leftrightarrow j h_{HB}[n] \exp(-j\pi n/2)$
- $H_{HB}(\theta + \pi/2) \leftrightarrow j h_{HB}[n] \exp(j\pi n/2)$
- $h_{HT}[n] = 2 h_{HB}[n] \sin(\pi n/2)$



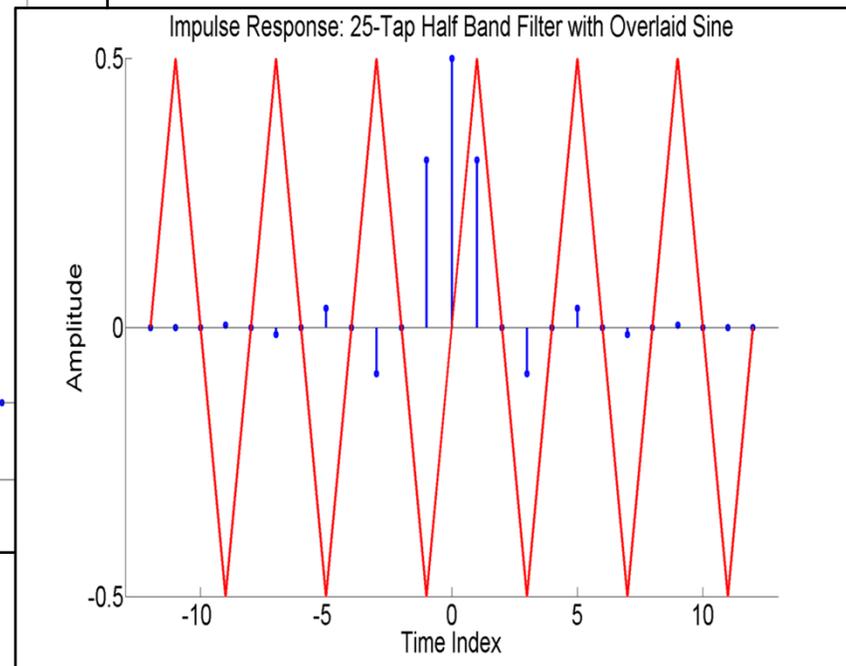
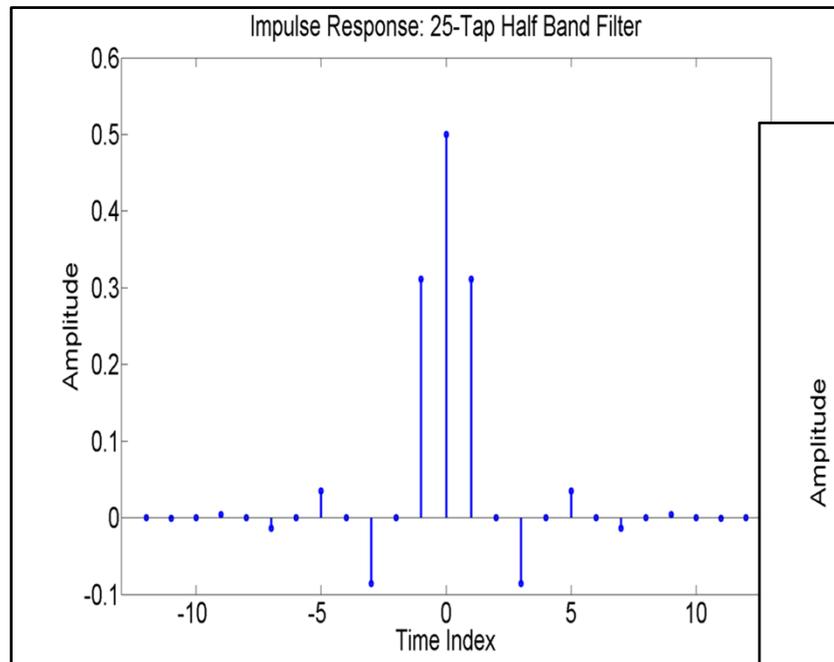
Hilbert Transform Coefs from Half Band Coefs

- $h_{HT}[n] = 2 h_{HB}[n] \sin(\pi n/2)$



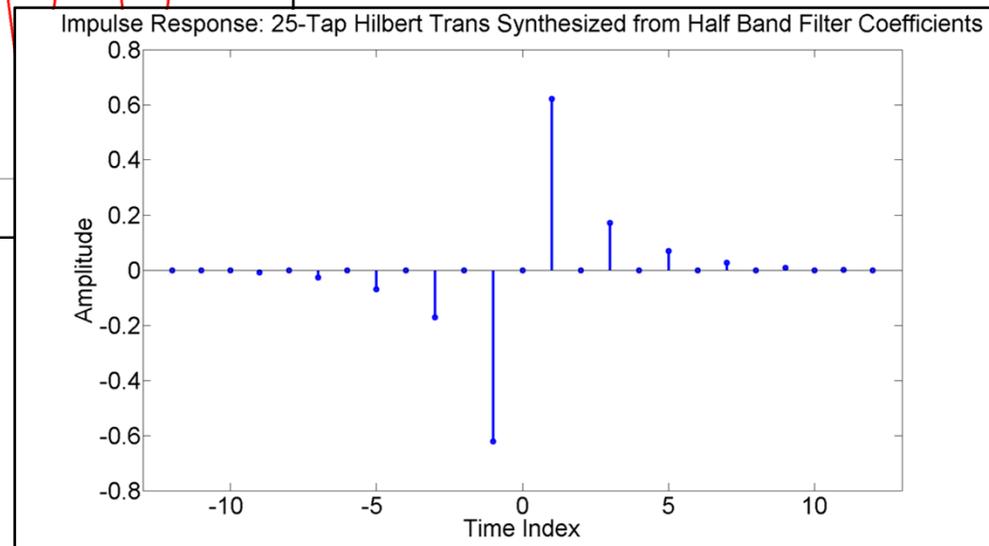
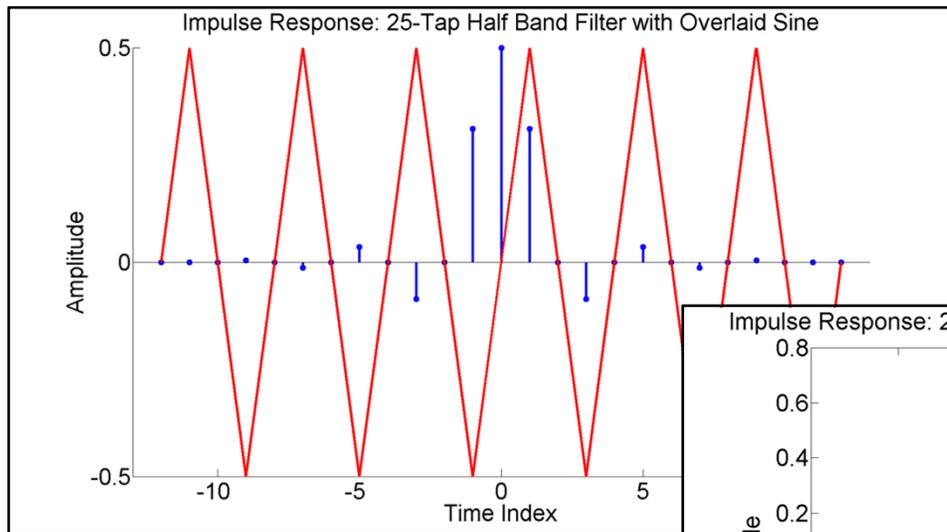
Hilbert Transform Coefs from Half Band (Con't)

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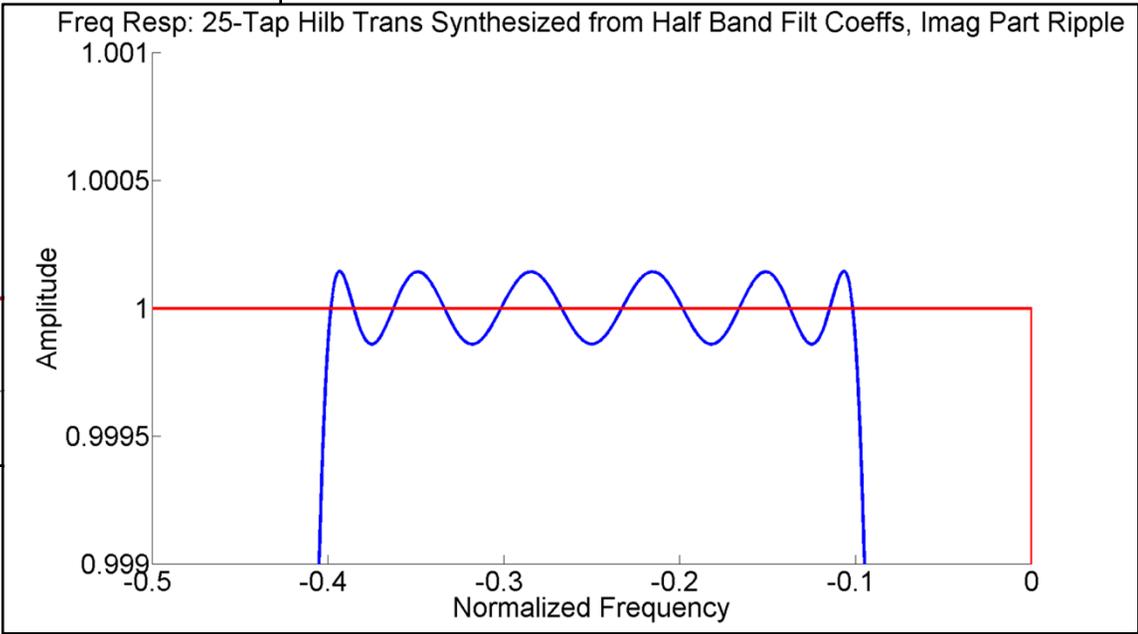
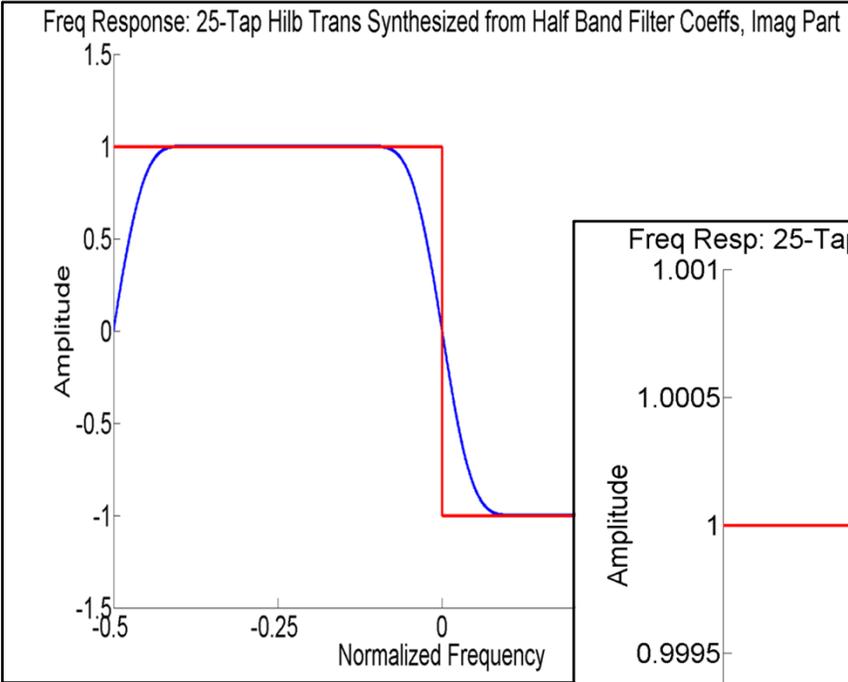
Hilbert Transform Coefs from Half Band (Con't)

- Sine wave applies Hilbert transform filter properties



Hilbert Transform Filter Response

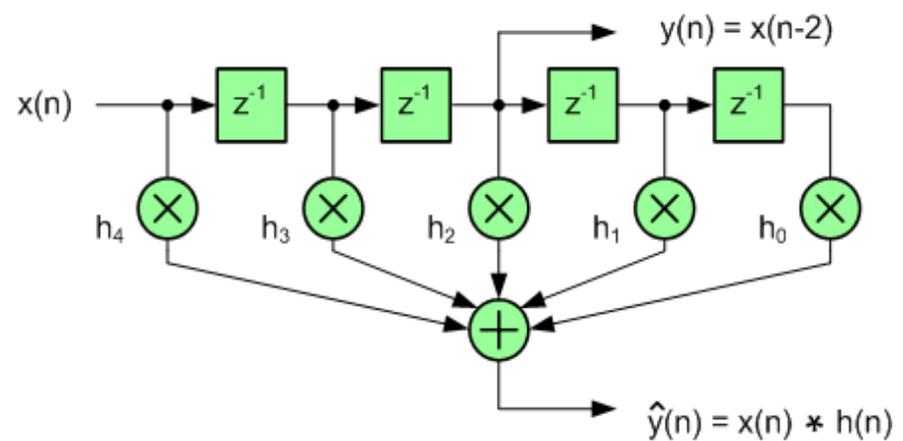
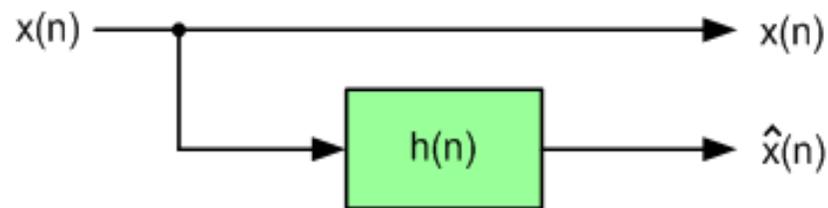
- Greatly improved passband ripple



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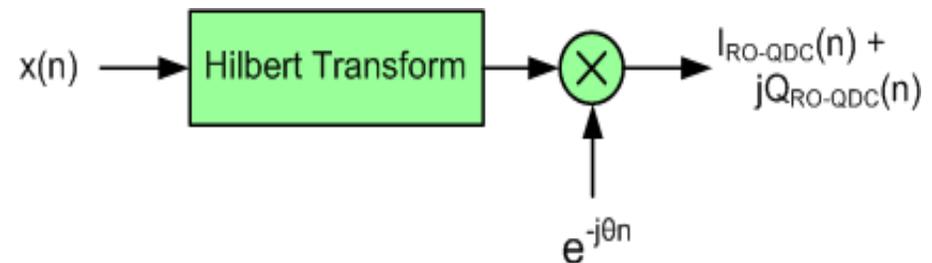
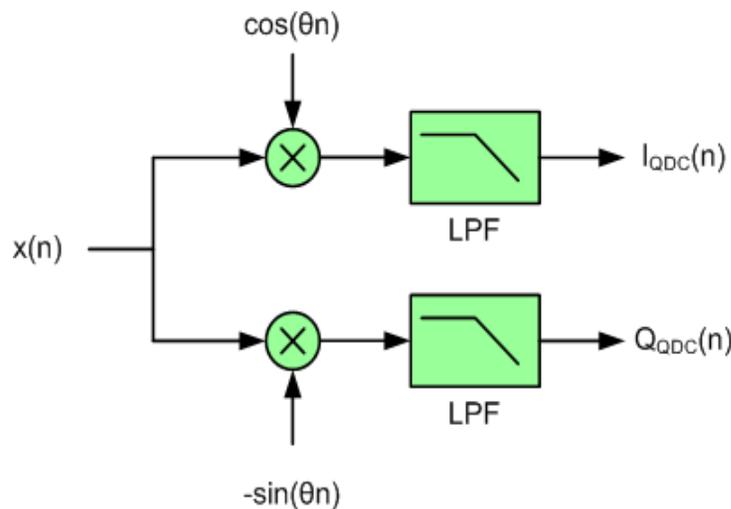
Implementation

- Hilbert Transformer operates on imaginary portion
- Delay real portion accordingly



Total Computations Required

- Quadrature Downconverter
 - Two low pass filters of order N, two multiplies
- Downconversion with Hilbert Transformer
 - One filter of order N, one complex multiply



Conclusion

- Compared quadrature downconverter and downconversion with Hilbert transformer
- Reviewed Hilbert transform
- Discussed windowing Hilbert transform filter
- Designed Hilbert transform filter in frequency
- Covered Design process for half band filter
- Results
- Implementation

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