

#### **Signal Processing in MATLAB**

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#### **Outline**

- Built-in MATLAB support for Signal Processing
- Overview of the Signal Processing Toolbox
- Some new MATLAB 5 Signal Processing features

#### **How to Start MATLAB**

- Login to UNIX prompt
- type matlab



#### **Signal Representation**

Weighted Sum of Sinusoids

```
x[n] = a1*sin(w1*n+phi1) + a2*sin(w2*n+phi2) + a3*sin(w3*n+phi3), 0 <= n <= N-1
```

Implementation 1 - non-vectorized, ala C or FORTRAN

```
N = 100;
a = [1 1/sqrt(2) 0.5];
w = [1 2 3]*.051*2*pi;
phi = [0 0 0];

x = zeros(N,1);
for n = 0:N-1
    for k = 1:3
        x(n+1) = x(n+1) + a(k)*sin(w(k)*n + phi(k));
    end
end
```

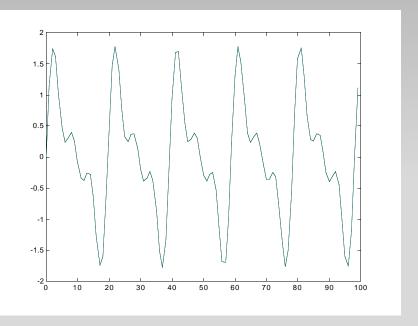


#### **Vectorization**

Implementation 2 - vectorized

```
n = 0:N-1;
x1 = sin(n'*w + phi(ones(1,N),:))*a';
```

- Plotting and Comparison
  - plot(0:N-1,[x x1])
  - norm(x-x1)





#### How did we do that?

```
sin(n'*w + phi(ones(1,N),:))*a'
```

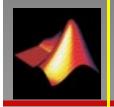
- dash ' is the (conjugate) transpose of a matrix
- •Outer-product n'\*w is N-by-3 matrix think of it as a weighted replication of row w with elements of n as weights
- •Replicated row phi into matrix [phi; phi; ... phi] using : indexing notation
- •Sum and sine are *element-wise* operations
- •Matrix Multiplication is a linear combination of column vectors



#### Functions - sos.m

• Create a file entitled sos.m containing the following text:

```
function x = sos(N,a,w,phi)
% SOS Weighted sum of sinusoids
% Inputs:
% N - length of sequence
% a - vector of amplitudes
% w - vector of frequencies (in radians)
% phi - vector of phases
% Uses Implementation 1 (non vectorized)
for n = 0:N-1
   x(n+1) = 0;
    for k = 1:3
       x(n+1) = x(n+1) + a(k)*cos(w(k)*n + phi(k));
    end
end
```



#### **About Functions**

- Help is automatic
  - The help for the function is everything after the 'function' line that starts with % (the comment character), up to the first line that is not a comment.
  - To get help on our function, (or ANY function in MATLAB), type 'help sos.m'
- Input and output arguments

```
function x = sos(N,a,w,phi)
```

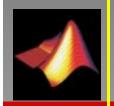
- This line defines N, a, w and phi as input arguments, and x as output argument
- These arguments are local to the function sos
  - We can have a variable of the same name in the calling workspace
  - Don't exist before or after function execution



#### **Another Function - sos1.m**

```
function x1 = sos1(N,a,w,phi)
% SOS1 Weighted sum of sinusoids
% Inputs:
% N - length of sequence
% a - vector of amplitudes
% w - vector of frequencies (in radians)
% phi - vector of phases
% Uses Implementation 2 (vectorized)

n = 0:N-1;
x1 = cos(n'*w + phi(ones(1,N),:))*a';
```



#### **Timing Comparison**

FACTOR OF 20 SPEED INCREASE BY USING VECTORIZATION



#### **Noise Signals**

- There are two functions, rand and randn, which generate matrices of random numbers
  - x=rand(m,n) creates m-by-n matrix of independent, uniformly distributed real numbers on interval [0,1]
  - x=randn(m,n) creates m-by-n matrix of independent, Normally distributed real numbers with mean 0 and variance 1
- rand and randn have internal states that determine the numbers produced. This state is initialized when you start MATLAB, and you can reset it at will. Example:



#### **More on Noise**

 Note that the random number generators have been improved in MATLAB 5. The old random number generators are still present and are activated by the command rand('seed',0). It is recommended that you remove any references to 'seed' from all your old MATLAB code so that you can use the improved generators.

```
Example:

High frequency noise
h = remez(40,[0 .4 .6 1],[0 0 1 1]);
noise = 5*randn(2*N,1);
noise = filter(h,1,noise);
noise = noise(end-N+1:end); % make it length N
```

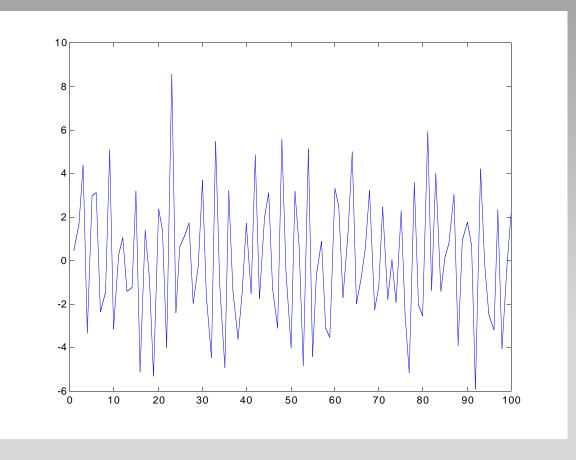
Note use of MATLAB 5 feature "end"

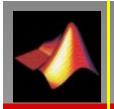


#### **Additive Noise Example**

```
x = x + noise;
plot(0:N-1,x)
```

The signal is completely buried!





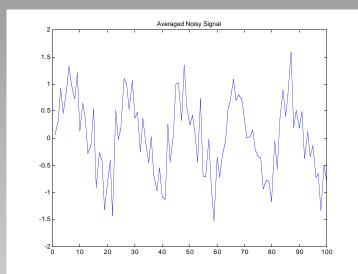
#### Attempt to remove noise

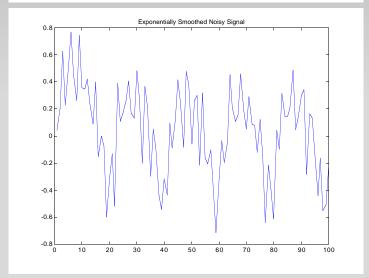
Smoothing with running average of 7 points

$$y[n] = 1/7 * (x[n] + x[n-1] + ... + x[n-6])$$

 "Exponential Smoothing" where present sample is forced to be similar to previous sample

$$y[n] = 0.9*y[n-1] + 0.1*x[n]$$







#### Digital Filtering with filter

 Both these smoothing operations and more can be implemented with filter:

```
y = filter([1 1 1 1 1 1 1]/7,1,x); % moving average FIR y = filter(.1,[1 -.9],x); % exponential smoothing IIR
```

FILTER One-dimensional digital filter.

Y = FILTER(B,A,X) filters the data in vector X with the filter described by vectors A and B to create the filtered data Y. The filter is a "Direct Form II Transposed" implementation of the standard difference equation:

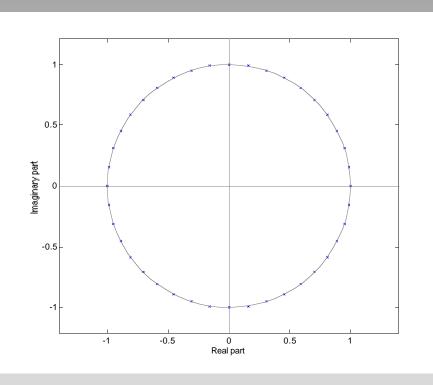
```
a(1)*y(n) = b(1)*x(n) + b(2)*x(n-1) + ... + b(nb+1)*x(n-nb)
- a(2)*y(n-1) - ... - a(na+1)*y(n-na)
```



#### **Transforms: FFT**

```
A = 1;
W = exp(-j*pi*.05);
M = 40;
z = A* W.^(-(0:M-1));
zplane([],z.')
```

Computes Z-transform at evenly-spaced points around the unit circle

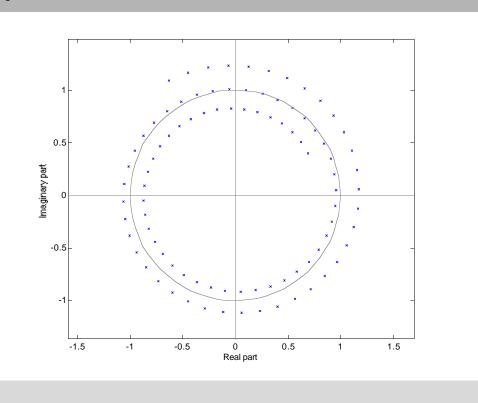




#### **Transforms: CZT**

```
A = .8 *exp( j*pi/6);
W = .995*exp(-j*pi*.05);
M = 91;
z = A* W.^(-(0:M-1));
zplane([],z.')
```

Domain is a spiral or "chirp" in the Z-plane





#### FFT vs. CZT

fft(x,M) Uses prime factor algorithm which is fastest when transform length is a power of 2.

czt(x,M,W,A) Uses next greatest power-of-2 FFT for fast computation.

#### **Timing Example**

```
>> x=randn(2027,1); % a prime length sequence
>> tic, fft(x); toc
elapsed_time =
          0.18934
>> tic, czt(x); toc
elapsed_time =
          0.11305
```

Note: both fft and czt work column-wise on matrices - very useful



# **Zoom Transform Application of Chirp Z-transform**

```
f1 = .4;
f2 = .7;
[b,a] = ellip(5,.1,40,[f1 f2]);
                                        1.002
M = 1000;
A = \exp(j*pi*f1);
W = \exp(-j*pi*(f2-f1)/(M-1));
                                        0.998
                                       0.996
=
0.994
H = czt(b,M,W,A)./czt(a,M,W,A)
f = linspace(f1,f2,M);
                                        0.992
                                         0.99
plot(f,abs(H))
                                        0.988
                                                   0.5 0.55
Frequency
                                               0.45
                                                              0.65
```

Efficiently computes frequency response in passband only

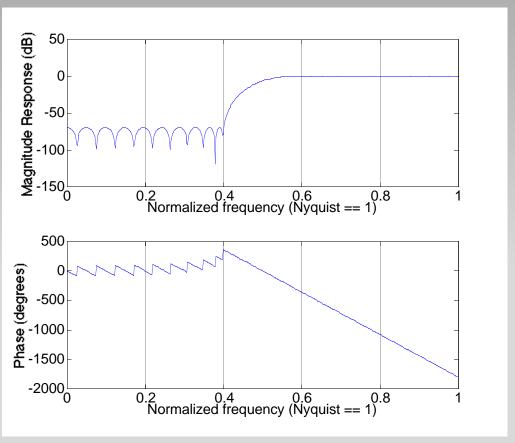


# Signal Processing Toolbox Overview - FIR Filter Design

• FIR example: h = remez(40,[0 .4 .6 1],[0 0 1 1]);

freqz(h)

High-pass Finite Impulse Response equiripple filter



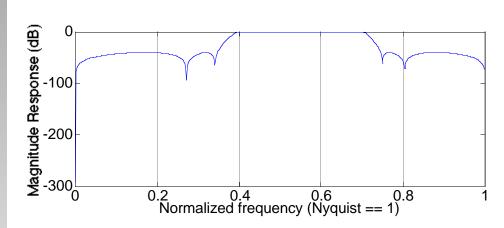


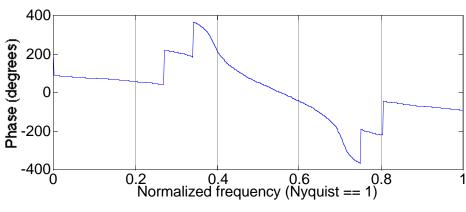
# Signal Processing Toolbox Overview - IIR Filter Design

• IIR example: [b,a] = ellip(5,.1,40,[.4 .7]);

freqz(b,a)

Band-pass Infinite Impulse Response equiripple (elliptic) filter







# Signal Processing Toolbox Overview - Other Filter Design Techniques

- FIR
  - Parks-McClellan (minimax) remez
  - Least Squares firls
  - Windowed method fir1, fir2
  - Constrained Least Squares fircls, fircls1
  - Complex, nonlinear phase cremez
- IIR
  - Butterworth, Chebyshev Type I and II, Elliptic butter, cheby1, cheby2, ellip
  - Piecewise linear magnitude approx. yulewalk
  - Arbitrary Mag. & Phase invfreqz
  - Generalized Butterworth (lowpass only) maxflat



#### Other M-files in the Signal Proc. Toolbox

- Spectral Analysis estimate Power Spectral Density
  - Welch's method (overlapped modified periodograms) psd
  - Maximum entropy method (AR modeling) pmem
  - MUSIC (eigenanalysis based) pmusic
  - Multitaper (discrete prolate spheroidal sequences) pmtm
- Parametric Modeling find AR or ARMA model for signal
  - AR model via autocorrelation technique lpc
  - ARMA prony
  - iterative ARMA stmcb



# and More M-files in the Signal Proc. Toolbox

- Graphic User Interface (GUI) sptool
- Other
  - Correlation functions xcorr, xcov
  - Hilbert transform hilbert
  - Spectrogram specgram
  - Resampling resample, upfirdn
  - Alternate Filtering schemes filtfilt, fftfilt



# upfirdn - Efficient Multirate FIR Filter bank Implementation



Syntax:

$$y = upfirdn(x,h,p,q)$$

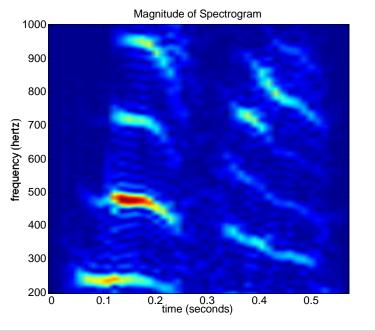
x and h can be arrays to implement BANKS of filters



# upfirdn Example: Compute spectrogram over range of frequencies

```
load mtlb
f = (200:5:1000); % frequencies in Hz
w = hamming(256);
h = exp(-j*(0:255)'*f*2*pi/Fs); % DFT filter bank
s = upfirdn(mtlb,h,1,100); % compute DFT every 100 samples
```

t = (0:100:length(mtlb)+256)/Fs;
imagesc(t,f,abs(s.')), axis xy





#### **New in MATLAB 5**

- Multidimensional Arrays
  - fftn N-dimensional FFT
  - convn N-dimensional convolution (filtering)
- Non-double data containers very useful for image and video work
  - uint8, int8
- Filter works on array inputs filters each column



# Other Signal Processing Relevant Toolboxes

- Image Processing images
- Higher Order Spectral Analysis (HOSA) hosa
- Wavelets wavelet
- Statistics stats
- System Identification ident
- Communications comm
- Optimization optim
- Symbolic Math symbolic
- Control Systems control
- Neural Networks nnet
- Fuzzy Logic fuzzy

For a list of available functions, type help signal, help wavelet, etc.



#### **Some Symbolic Basics in MATLAB 5**

• Define some symbolic variables using 'syms' command

```
» syms x a T w
» int(exp(a*x),0,T)
                    ----- INTEGRATE
ans =
1/a*exp(T*a)-1/a
                    ——— SOLVE EQUATION(S)
\Rightarrow solve(x^3+a)
ans =
                     (-a)^{(1/3)}
[-1/2*(-a)^{(1/3)}-1/2*i*3^{(1/2)}*(-a)^{(1/3)}]
[-1/2*(-a)^{(1/3)+1/2*i*3^{(1/2)*(-a)^{(1/3)}}]
-i/a*exp(1/2*i*T*w)+i/a*exp(-1/2*i*T*w)
y =
2*\sin(1/2*T*w)/w
                   —— TAKE LIMITS
»limit(y,w,0) ◀
ans =
```