

# Homework 9

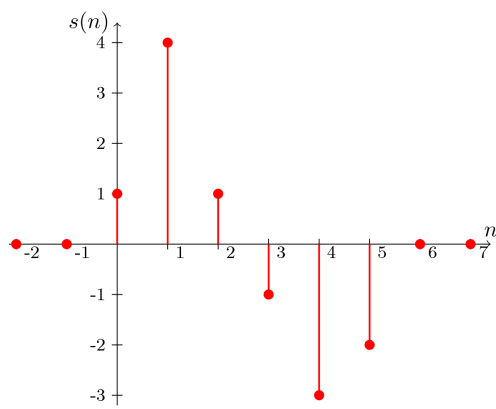
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```
figure('position', [0, 0, 800, 300])
```

### [✓] ADSI Problem 6.1: Matched filters

Consider a deterministic signal that is only non-zero in the vicinity of  $n=0$  as shown below and zero for all other values of  $n$ .



The signal is disturbed by additive noise from an AR(3) process given by

$$v(n) = -0.5v(n-1) - 0.5v(n-2) - 0.25v(n-3) + w(n)$$

where  $w(n) \sim WGN(0, 4)$ .

The goal of this problem is to construct a matched filter that can help us distinguish between the presence or absence of the signal in the noise.

```
clear variables;
```

## 1) Plot a realisation of the signal with the added noise

Create a realization of the noise and add the signal somewhere in the noise. Plot the result and comment on whether the signal is detectable.

```
D = 50; % The place to embed signal into the noise

% Generate the deterministic signal s[n]
s = [1, 4, 1, -1, -3, -2]';

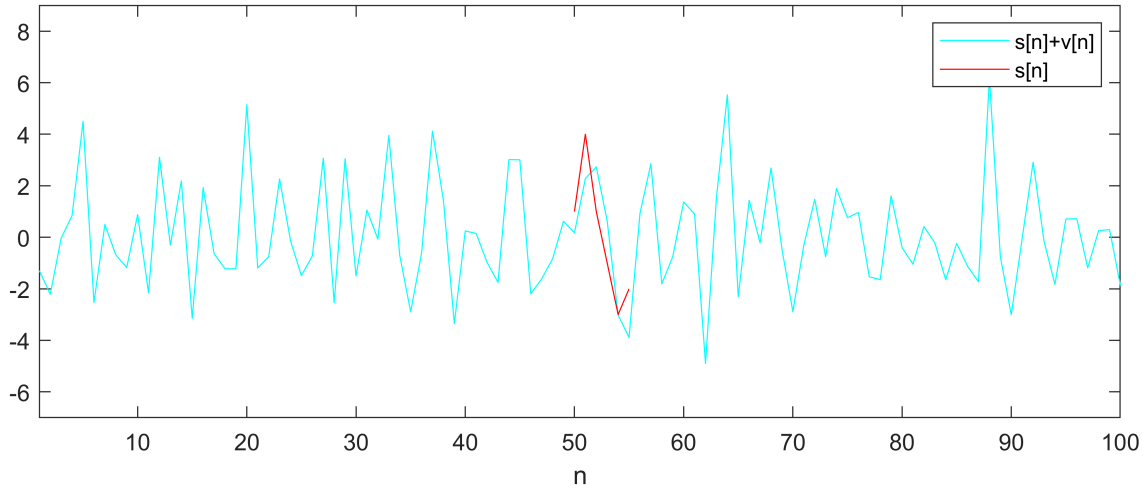
% Generate white noise w[n] with zero mean and variance 4
w_var = 4;
w = sqrt(w_var) * randn(10000, 1);

% Generate noise from an AR(3) process
b = 1;
a = [1, 0.5, 0.5, 0.25];
v = filter(b, a, w);

% Extract 100 samples to remove the transient effect
% at the beginning and at the end of the filtered signal
x = v(500:600);

% Embed the signal s[n] in v[n] from D
s_in_x_start = D;
s_in_x_end = D + length(s) - 1;
x(s_in_x_start:s_in_x_end) = x(s_in_x_start:s_in_x_end) + s;

plot(1:length(x), x, 'c', ...
     s_in_x_start:s_in_x_end, s, 'r') % the signal is completely invisible
legend('s[n]+v[n]', 's[n]')
xlabel('n')
xlim([1,100])
ylim([-7, 9])
```



The signal is cannot be distinguished from the noise.

## 2) Plot the power density spectrum of the noise and spectrum of the signal

Plot the power density spectrum of the noise and spectrum of the signal, see Eq. (14.91).

The noise is generated by an AR(3) process plus some white Gaussian noise:

$$v(n) = -0.5v(n-1) - 0.5v(n-2) - 0.25v(n-3) + w(n)$$

where  $w(n) \sim WGN(0, 4)$ .

We can determine the Power Density Spectrum of an ARMA(p,q) process is given by

$$S_{yy}(\omega) = \sigma_x^2 |H(e^{j\omega})|^2 = \sigma_x^2 \left| \frac{\sum_{k=0}^q b_k e^{-j\omega k}}{1 + \sum_{k=1}^p a_k e^{-j\omega k}} \right|^2. \quad (13.133)$$

The power spectrum of an AR(p) process is given by:

$$S_{yy}(\omega) = \sigma_x^2 \left| \frac{1}{1 + \sum_{k=1}^p a_k e^{-j\omega k}} \right|^2$$

For this problem, we have:

$$S_{vv}(\omega) = 4 \left| \frac{1}{1 + 0.5e^{-j\omega} + 0.5e^{-j2\omega} + 0.25e^{-j3\omega}} \right|^2$$

The algorithm is as follows:

1. Use the coefficients  $\{a_1, a_2, \dots, a_p\}$  for the AR(p) model,
2. Compute the transfer function for the AR(p) by computing the sum and finding its reciprocal

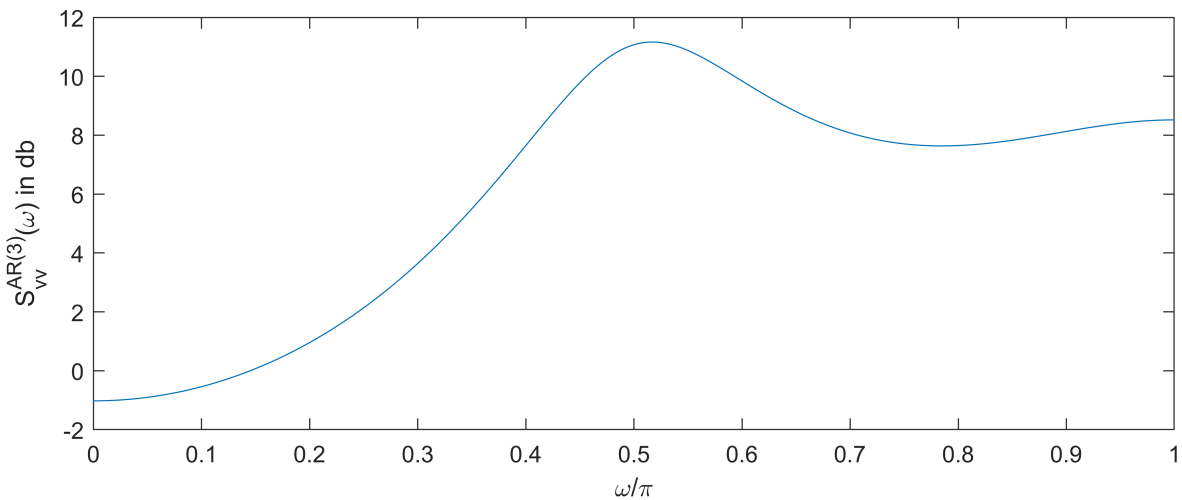
3. Compute the conjugate of the transfer function:  $|H(e^{j\omega})|^2$
4. Multiply it with the variance  $\sigma_x^2$

The algorithm is implemented in the functions `ar2psd()` function:

```
N = 256;

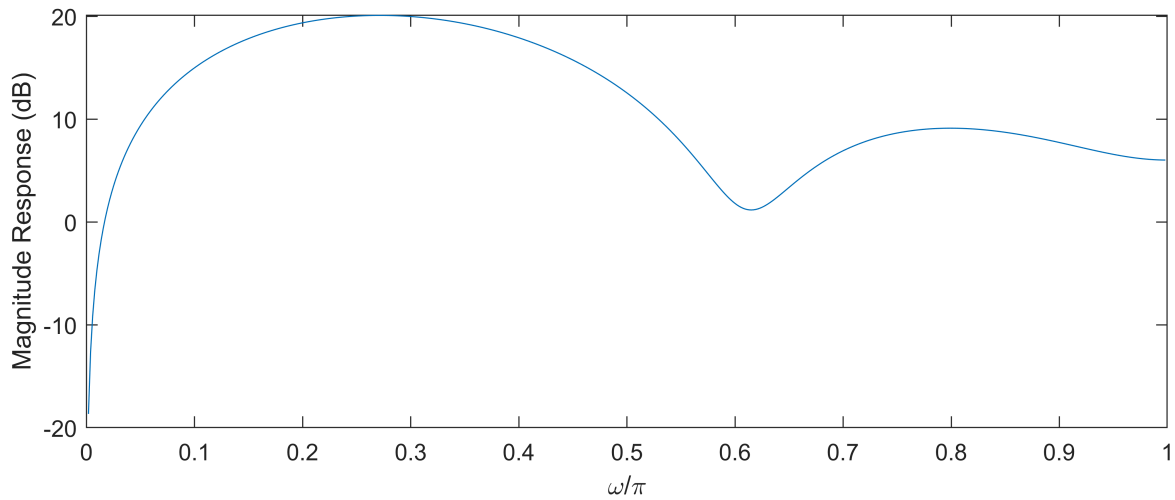
a = [0.5, 0.5, 0.25]; % The coefficients of the AR(3) model
w_var = 4; % The variance of white noise
[S_vv, w] = ar2psd(a, w_var, N); % Compute the PDS of AR(3) model

plot(w/pi, pow2db(S_vv))
xlabel('\omega/\pi')
ylabel('S_{vv}^{AR(3)}(\omega) in db')
```



We can plot the spectrum of the signal  $s(n)$  using the `freqz()` function:

```
[H,w2] = freqz(s, 1);
plot(w2/pi, 20*log10(abs(H)))
xlabel('\omega/\pi')
ylabel('Magnitude Response (dB)')
```



### 3) Calculate the matched filter

Calculate the matched filter, the frequency response of the matched filter and the optimum signal to noise ratio. You can use `xcorr` on the noise realization to find  $\mathbf{R}_v$ .

The impulse response of the optimum (matched) filter is given by:

$$\mathbf{h} = \kappa \mathbf{R}_v^{-1} \mathbf{s}. \quad (14.97)$$

where  $k$  is the normalisation factor. Although the maximum SNR can be obtained by any choice of constant  $\kappa$ , we choose the constant by requiring that:

- (a)  $\mathbf{h}^T \mathbf{s} = 1$ , which yields  $\kappa = 1/\mathbf{s}^T \mathbf{R}_v^{-1} \mathbf{s}$
- (b)  $E(v_0^2[n]) = \mathbf{h}^T \mathbf{R}_v \mathbf{h} = 1$ , which yields  $\kappa = 1/\sqrt{\mathbf{s}^T \mathbf{R}_v^{-1} \mathbf{s}}$ .

```
M = numel(s); % Signal length

% The autocorrelation matrix must be MxM since
% its inverse is multiplied by a M-tap signal s(n)
[r_vv, lags] = xcorr(v, M-1, 'biased');
R_vv = toeplitz(r_vv(M:end));

% Compute normalisation factor
k = 1/sqrt(s'*(R_vv\s)); % Same as 1/sqrt(s'*inv(R_vv)*s)

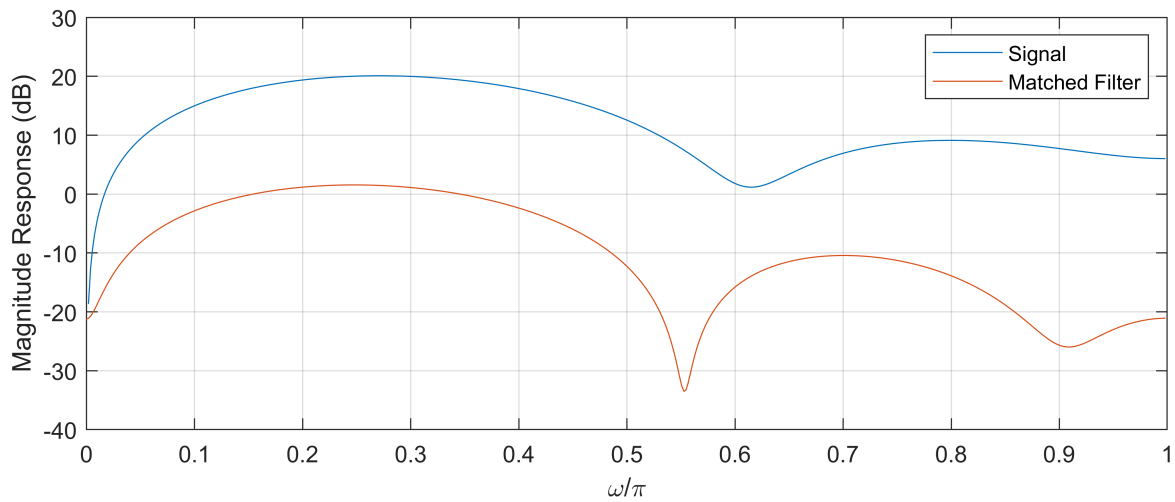
% Compute the filter
h = k*(R_vv\s); % Same as k*inv(R_vv)*s

[H_s, w_s] = freqz(s, 1);
[H_h, w_h] = freqz(h, 1);
plot(w_s/pi, 20*log10(abs(H_s)), w_h/pi, 20*log10(abs(H_h)))
```

```

legend('Signal', 'Matched Filter')
xlabel('\omega/\pi')
ylabel('Magnitude Response (dB)')
grid on;

```



The maximum possible value of the output SNR is given by:

$$\text{SNR}_0 = a^2 \tilde{s}^T \tilde{s} = a^2 s^T R_v^{-1} s. \quad (14.98)$$

Since the attenuation factor  $a$  is not given in this problem, we assume that it is 1:

```

a = 1;
SNR = a^2 * s' * (R_vv\s)

```

```

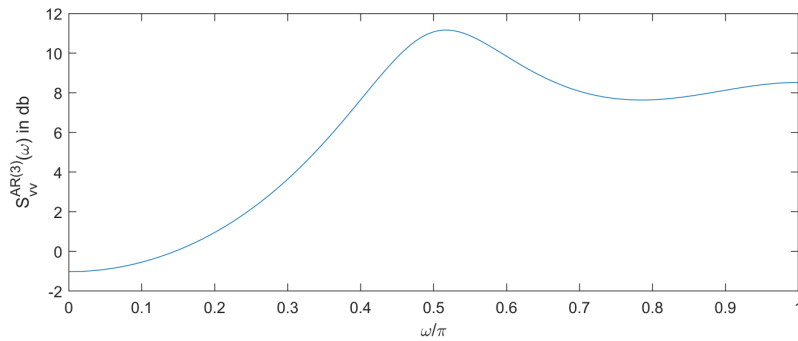
SNR = 11.5754

```

#### 4) Does the frequency response of the matched filter make sense?

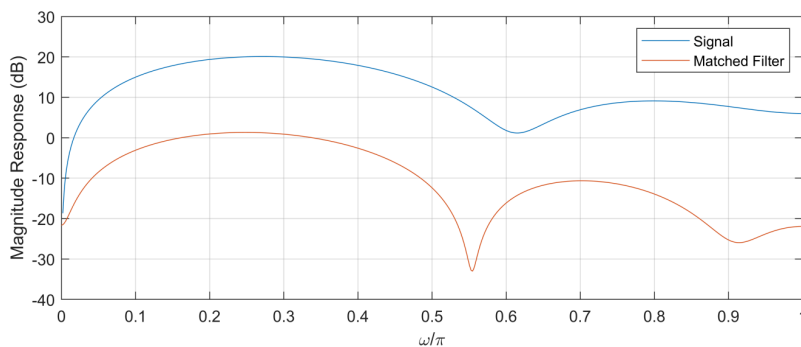
*Does the frequency response of the matched filter make sense seen in relation with the spectra from question 2?*

In question 2) we found that the AR(3) noise model has peak power at 0.5.



Looking at the frequency response of the signal, we observe that it peaks around 0.25 and dips around 20 dB at around 0.6.

The frequency response of the matched filter shows that it is minimising the noise in the region at around 0.55. The noise power peaks at around 0.5



## 5) How efficient is the matched filter?

*Plot the signal before and after the matched filter as well as the square of the output of the matched filter on the same graph and comment on the efficiency of the matched filter in our quest to detect the presence of the signal.*

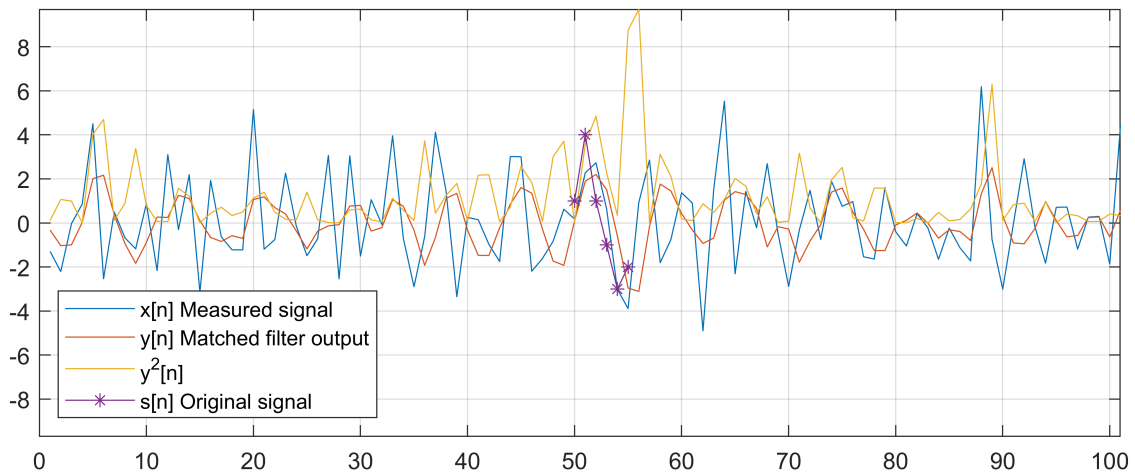
```
% x(n) is the 100 samples extracted from the noise signal v(n)
% y(n) is the result of feeding x(n) to the matched filter
y = filter(h, 1, x);
y_squared = y.^2;

n = 1:length(x);

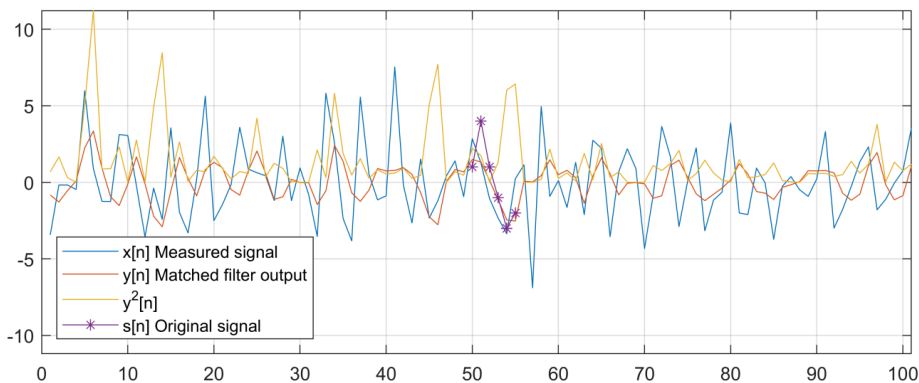
n2 = s_in_x_start:s_in_x_end;
xs = s;

plot(n,x, n,y, n,y_squared, n2,xs,'*-')
xlim([0, numel(n)])
ylim([-max(y_squared), max(y_squared)])
legend({'x[n] Measured signal', 'y[n] Matched filter output', 'y^2[n]', 's[n] Original signal'})

grid on;
```



It's close to impossible to find the original signal  $s[n]$  in the measured signal  $x[n]$ . It is also difficult to detect  $s[n]$  after filtering  $x[n]$  with the matched filter i.e.,  $y[n]$ . It is easier to detect the original signal in the square of the output of the matched filter  $y^2[n]$ . Running the code multiple times i.e., creating different realisations of the noise, we may see many false positives. The figure below shows an example of this. The highest response is measured at  $n = 6$  but we know that the signal is embedded in  $n = 50$ .



## 6) What happens to the optimum SNR when the noise is twice as powerful?

Is the optimum signal to noise ratio halved if the AR(3) is instead driven by twice as powerful white noise, i.e.  $w(n) \sim \text{WGN}(0,8)$ ? Why or why not?

We will compute the SNR again but this time with  $\sigma_w^2 = 8$ :

$$\text{SNR}_0 = a^2 \tilde{s}^T \tilde{s} = a^2 s^T R_v^{-1} s. \quad (14.98)$$

```
% Generate white noise w[n] with zero mean and variance 8
w2_var = 8;
w2 = sqrt(w2_var) * randn(10000, 1);

% Put the white noise through an AR(3) process
```



```

b = 1;
a = [1, 0.5, 0.5, 0.25];
v2 = filter(b, a, w2);

% Extract 100 samples to remove the transient effect
% at the beginning and at the end of the filtered signal
x2 = v2(500:600);

M = numel(s); % Signal length

% The autocorrelation matrix must be MxM since
% its inverse is multiplied by a M-tap signal s(n)
[r_vv2, lags] = xcorr(x2, M-1, 'biased');
R_vv2 = toeplitz(r_vv2(M:end));

a = 1;
SNR2 = a^2 * s' * (R_vv2\s)

```

```
SNR2 = 6.6825
```

```
SNR2/SNR
```

```
ans = 0.5773
```

The optimal signal-to-noise-ratio is almost halved when the white noise is doubled. The reduction of SNR is not always decreased by a factor of two though. Although the noise power is doubled, the shape of the optimum filter will change because it depends on the measured signal:

$$\mathbf{h} = \kappa \mathbf{R}_v^{-1} \mathbf{s}. \quad (14.97)$$

## 7) Compute SNR using a 6-tap moving average filter

Assume that instead of the matched filter a simple 6-tap moving average filter is used instead. That is

$$\mathbf{h} = \left[ \frac{1}{6} \quad \frac{1}{6} \quad \frac{1}{6} \quad \frac{1}{6} \quad \frac{1}{6} \quad \frac{1}{6} \right]^T$$

Calculate the signal to noise ratio when the moving average filter is used.

We can compute the SNR using the following formula:

$$\text{SNR}_0 = \frac{s_0^2[n_0]}{E(v_0^2[n_0])} = a^2 \frac{(\mathbf{h}^T \mathbf{s})^2}{\mathbf{h}^T \mathbf{R}_v \mathbf{h}}. \quad (14.94)$$

```

% h_avg = [1, 0, 1, 1, 1, 1]'./5;
h_avg = [1, 1, 1, 1, 1, 1]'./6;
a = 1;
SNR_avg = a^2 * (h_avg'*s)^2 / (h_avg'*R_vv*h_avg)

```

SNR\_avg = 0

The SNR of the moving average filter is zero (or a number very close to zero). The moving average filter gives us the average of a signal. Since the original signal has zero mean, the average is zero. This is not surprising. If we change the filter, then the SNR is no longer zero.

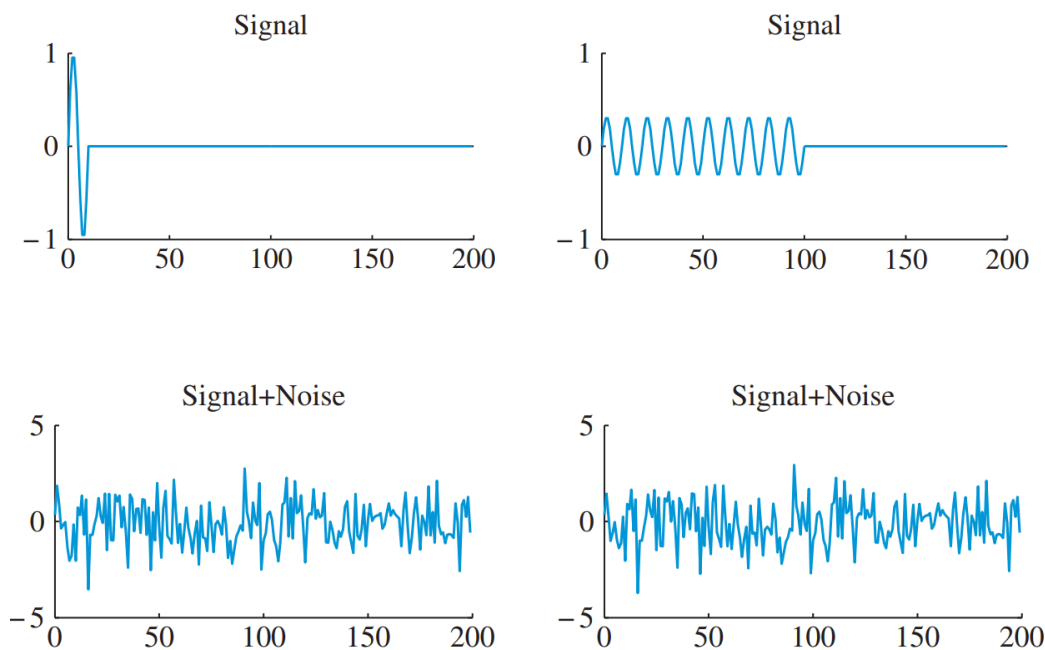
```
h4 = [1, 0, 1, 1, 1, 1]'./5;  
a = 1;  
SNR4 = a^2 * (h4'*s)^2 / (h4'*R_vv*h4)
```

SNR4 = 1.4075

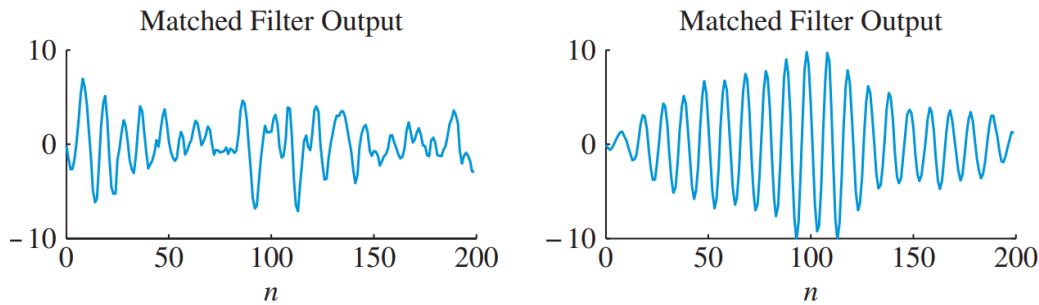
### Problem 14.14: Show identity relations hold for auto- and cross-covariance sequences

In this problem we discuss in detail the matched filtering problem illustrated in Figure 14.18.

**Figure 14.18** The operation of a matched filter in white noise. The two signals have different length ( $p = 10$  and  $p = 100$ , respectively) but the same energy.



We note that the peak response of the matched filter occurs at  $n_0 = p - 1$  because there is no time delay ( $D = 0$ ).



### 1) Determine the impulse response of the matched filter and the output SNR.

Determine the impulse response of the matched filter and the output SNR. Explain why, in the absence of noise, the output is the ACRS of the desired signal.

### 2) Determine the matched filter of a short cosine signal

Suppose that  $p = 10$  and  $s_i[n] = \cos\left(2\pi \frac{n}{10}\right)$ .

Generate  $N = 200$  samples of the noisy signal  $x[n]$  and process it through the matched filter designed for  $s_i[n]$ .

Plot the desired, input, and filtered signals and determine when the matched filter output is output.

### 2) Determine the matched filter of a long cosine signal

### 4) Which signal can be detected more easily by visual inspection?

Which signal can be detected more easily by visual inspection of the matched filter output? Is this justified by comparing the output SNR in each case?

### Problem 14.35:

Consider the two 10-point signals  $x_0[n]$  and  $x_1[n]$  given below:

```
clear variables;
```

### Problem 14.37:

```
clear variables;
```

### Exam 2013, Problem 3

```
clear variables;
```

### Functions

```
function [S, w] = ar2psd(a, v, N)
% AR2PSD Compute the Power Spectral Density from AR(p) coefficients
% [S, w] = ar2psd(a, v, N)
% a: AR(p) coefficients
% v: the variance
% N: number of points in the range [1, pi]
% S: the estimated power spectrum
% w: frequencies
```

```

w = linspace(0, 1, N) * pi;

% Compute the transfer function
% Used Eq. (13.133) in the book
H = ones(N, 1);
for k=1:numel(a)
    H = H + a(k)*exp(-1j * w' * k);
end
H = 1./H;

% Finally compute the PSD
S = v * H.*conj(H);
end

```