

TTT4120 Digital Signal Processing Solutions for Problem Set 8

Problem 1:

(a).

The filter H(z) is a first order all-pole filter. The output of the filter is therefore an AR[1] process.

(b).

The first order predictor is defined as

$$\widehat{x}[n] = -a_1 x[n-1]$$

The prediction coefficient $-a_1$ can be found by minimizing the prediction error power $\sigma_f^2 = E\{f^2[n]\}$, where $f[n] = x[n] - \hat{x}[n]$ is the prediction error. We have that

$$\sigma_f^2 = E\{(x[n] - \widehat{x}[n])^2\} = E\{(x[n] + a_1x[n-1])^2\}$$

The optimal value for a_1 can be found by minimizing the prediction error power, i.e $\frac{\partial \sigma_f^2}{\partial a_1} = 0$. Thus, we get

$$E\{2(x[n] + a_1x[n-1])x[n-1]\} = 0,$$

$$\gamma_{xx}(1) + a_1 \gamma_{xx}(0) = 0 \Rightarrow a_1 = -\frac{\gamma_{xx}(1)}{\gamma_{xx}(0)}.$$

In the previous problem set the autocorrelation function of the signal x[n] was found as

$$\gamma_{xx}(m) = \left(-\frac{1}{2}\right)^{|m|}.$$

Thus, we have $a_1=\frac{1}{2}$. By repeating the procedure for the second order predictor, we get $a_1=\frac{1}{2}$ and $a_2=0$. This means that we can not obtain further reduction of the prediction error by using a higher order peredictor.

The above results could be expected, since x[n] is an AR[1] process. The optimal predictor is thus the first order predictor with prediction coefficient equal to the filter coefficient.

Problem 2:

(a).

This is a MA(1)-process, as only the current and the former value of the input signal are used in forming the output signal.

(b).

We have

$$x[n]x[n-1] = \left(w[n] - 0.5w[n-1]\right)\left(w[n-l] - 0.5w[n-l-1]\right)$$

$$= w[n]w[n-l] - 0.5w[n]w[n-l-1] - 0.5w[n-1]w[n-l] + 0.25w[n-1]w[n-l-1].$$

Taking the expectation of this leads to

$$\begin{split} \gamma_{xx}(l) &= \gamma_{ww}(l) - 0.5\gamma_{ww}(l+1) - 0.5\gamma_{ww}(l-1) + 0.25\gamma_{ww}(l) \\ &= 1.25\gamma_{ww}(l) - 0.5\left(\gamma_{ww}(l+1) + \gamma_{ww}(l-1)\right) \\ &= 1.25\sigma_{w}^{2}\delta(l) - 0.5\sigma_{w}^{2}\left(\delta(l+1) + \delta(l-1)\right) \\ &= \begin{cases} 1.25 & l = 0, \\ -0.5 & l = \pm 1, \\ 0 & \text{otherwise.} \end{cases} \end{split}$$

The expression for the power density spectrum $\Gamma_{xx}(f)$ can be found as follows

$$\Gamma_{xx}(f) = \sum_{l=-1}^{1} \gamma_{xx}(l)e^{-j\omega l} = -\frac{1}{2}e^{j\omega} + 1.25 - \frac{1}{2}e^{-j\omega} = 1.25 - \cos(2\pi f).$$

(c).

The optimal predictor of order p is given by:

$$\widehat{x}[n] = -\sum_{k=1}^{p} a_k x[n-k].$$

To obtain simple matrix equations that we can solve using Matlab, we use the version of the Yule-Walker that does not contain σ_f^2 , i.e.

$$\begin{split} \gamma_{xx}(0)a_1 &= -\gamma_{xx}(-1) \\ \begin{bmatrix} \gamma_{xx}(0) & \gamma_{xx}(1) \\ \gamma_{xx}(-1) & \gamma_{xx}(0) \end{bmatrix} \begin{bmatrix} a_1 \\ a_2 \end{bmatrix} &= \begin{bmatrix} -\gamma_{xx}(-1) \\ -\gamma_{xx}(-2) \end{bmatrix} \\ \begin{bmatrix} \gamma_{xx}(0) & \gamma_{xx}(1) & \gamma_{xx}(2) \\ \gamma_{xx}(-1) & \gamma_{xx}(0) & \gamma_{xx}(1) \\ \gamma_{xx}(-2) & \gamma_{xx}(-1) & \gamma_{xx}(0) \end{bmatrix} \begin{bmatrix} a_1 \\ a_2 \\ a_3 \end{bmatrix} &= \begin{bmatrix} -\gamma_{xx}(-1) \\ -\gamma_{xx}(-2) \\ -\gamma_{xx}(-3) \end{bmatrix} \end{split}$$

for order one, two, and three respectively. Then we calculate σ_f^2 as

$$\sigma_f^2 = \sum_{k=0}^p a_k \gamma_{xx}(k).$$

The following Matlab code can be used to find the coefficients, and σ_f^2 for each AR order.

```
gamma_xx=[1.25 -0.5 0 0];
R1 = [1.25];
R2 = [1.25 -.5; -.5 1.25];
R3 = [1.25 -.5 0; -.5 1.25 -.5; 0 -.5 1.25];
a_1 = R1^(-1)*-gamma_xx(2)'
sigma_f1 = sum([1 a_1'].*gamma_xx(1:2))
a_2 = R2^(-1)*-gamma_xx(2:3)'
sigma_f2 = sum([1 a_2'].*gamma_xx(1:3))
a_3 = R3^(-1)*-gamma_xx(2:4)'
sigma_f3 = sum([1 a_3'].*gamma_xx(1:4))
```

Approximated values are summarized in the following table.

AR order p	Coefficients	$\overline{\sigma_f^2}$
1	$a_1 = 0.40$	1.05
2	$a_1 = 0.48, a_2 = 0.19$	1.01
3	$a_1 = 0.49, a_2 = 0.24, a_3 = 0.09$	1.00

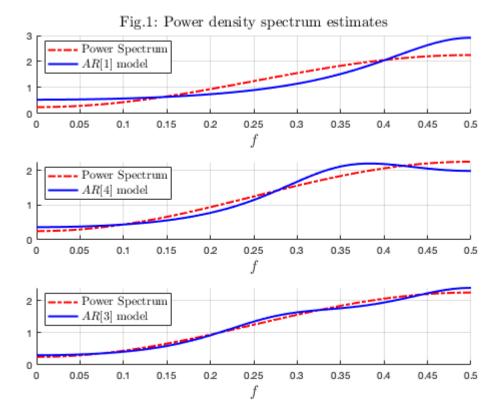
We can see that the mean square error decreases as we increase the model order. This means that the AR model is a better approximation of the MA[1] process.

(d).

The power density spectrum estimate obtained by using an AR[p] model is given by

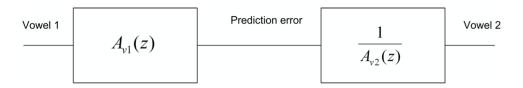
$$\widehat{\Gamma}_{f\!f}(f) = \Gamma_{f\!f}(f) |H(f)|^2 = \sigma_f^2 \left| \frac{1}{A(f)} \right|^2 = \frac{\sigma_f^2}{\left| 1 + \sum_{k=1}^p a_k e^{-j2\pi f k} \right|^2}$$

Fig.1 shows the power density spectrum estimates based on AR models of different order compared to the power density spectrum of the MA[1] process. It can be seen that the estimates become closer to the correct value as the model order increases. Thus, the AR[3] model is the best approximation of these three.



Problem 3:

- First, the vowel sample files are loaded with audioread.
- Then each vowel is modelled as an AR[10] process which the coefficients can be found with 1pc (needs to install the signal processing toolbox).
- To transform a vowel v_i into another vowel v_j we need the prediction error signal from v_i by useing filter and the AR coefficients of that vowel which is shown in the first part of Fig.2.



- Generate the new vowel v_j by filtering the prediction error using the inverse prediction-error filter (with the coefficients that you found for v_j) which represents in the second part of Fig.2.
- The output signal can be played using sound.