



پردازش سیگنال دیجیتال

درس ۲۷

مروری بر فیلترهای وفقی

Overview of Adaptive Filters

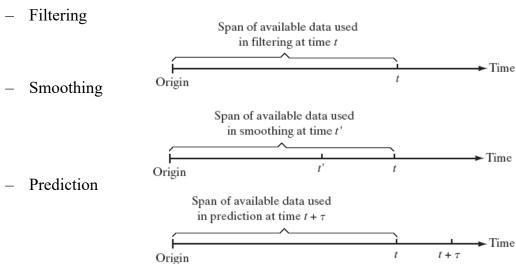
کاظم فولادی قلعه دانشکده مهندسی، پردیس فارابی دانشگاه تهران

http://courses.fouladi.ir/dsp

Overview of Adaptive Filters

The Filtering Problem

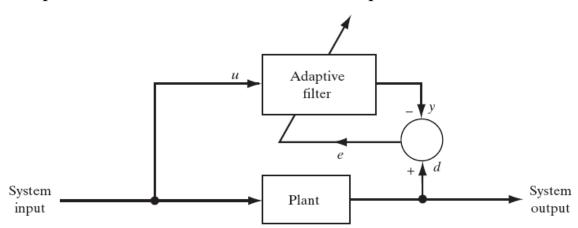
• Filters may be used for three information-processing tasks



- Given an optimality criteria we often can design **optimal filters**
 - Requires a priori information about the environment
 - Example:
 Under certain conditions the so called Wiener filter is optimal in the mean-squared sense
- Adaptive filters are self-designing using a recursive algorithm
 - Useful if complete knowledge of environment is not available a priori

Applications of Adaptive Filters: Identification

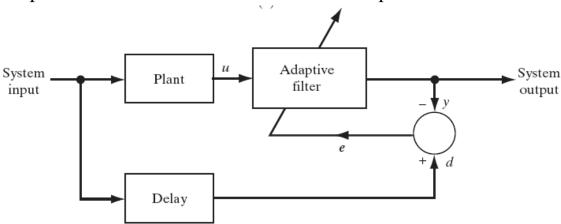
• Used to provide a linear model of an unknown plant



- Parameters
 - u = input of adaptive filter = input to plant
 - -y =output of adaptive filter
 - d = desired response = output of plant
 - e = d y =estimation error
- Applications:
 - System identification

Applications of Adaptive Filters: Inverse Modeling

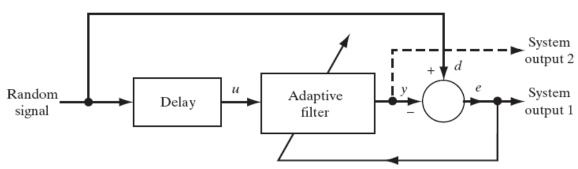
• Used to provide an inverse model of an unknown plant



- Parameters
 - u = input of adaptive filter = output to plant
 - -y =output of adaptive filter
 - d = desired response = delayed system input
 - -e = d y =estimation error
- Applications:
 - Equalization

Applications of Adaptive Filters: Prediction

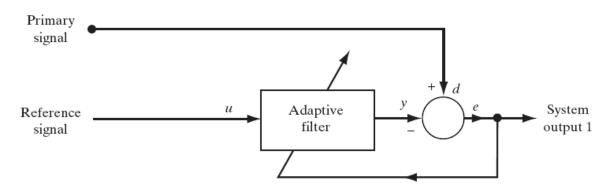
• Used to provide a prediction of the present value of a random signal



- Parameters
 - -u = input of adaptive filter = delayed version of random signal
 - -y =output of adaptive filter
 - d = desired response = random signal
 - -e = d y =estimation error = system output
- Applications:
 - Linear predictive coding

Applications of Adaptive Filters: Interference Cancellation

• Used to cancel unknown interference from a primary signal



Parameters

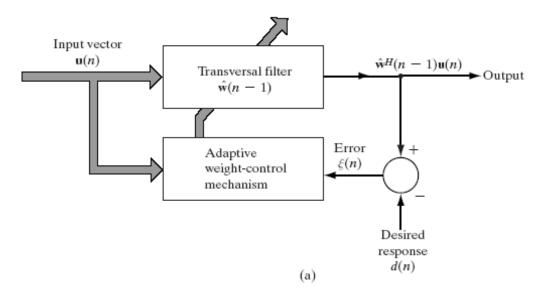
- u = input of adaptive filter = reference signal
- -y =output of adaptive filter
- d = desired response = primary signal
- e = d y = estimation error = system output
- Applications:
 - Echo cancellation

Stochastic Gradient Approach

- Most commonly used type of Adaptive Filters
- Define cost function as mean-squared error
 - Difference between filter output and desired response
- Based on the method of steepest descent
 - Move towards the minimum on the error surface to get to minimum
 - Requires the gradient of the error surface to be known
- Most popular adaptation algorithm is LMS
 - Derived from steepest descent
 - Doesn't require gradient to be know: it is estimated at every iteration
- Least-Mean-Square (LMS) Algorithm

Least-Mean-Square (LMS) Algorithm

- The LMS Algorithm consists of two basic processes
 - Filtering process
 - Calculate the output of FIR filter by convolving input and taps
 - Calculate estimation error by comparing the output to desired signal
 - Adaptation process
 - Adjust tap weights based on the estimation error



LMS Algorithm Steps

• Filter output

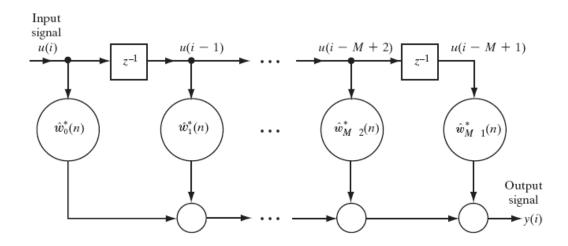
$$y[n] = \sum_{k=0}^{M-1} u[n-k] w_k^*[n]$$

• Estimation error

$$e[n] = d[n] - y[n]$$

• Tap-weight adaptation

$$W_k[n+1] = W_k[n] + \mu u[n-k]e^*[n]$$



Stability of LMS

• The LMS algorithm is convergent in the mean square if and only if the step-size parameter satisfy

$$0 < \mu < \frac{2}{\lambda_{\text{max}}}$$

- Here λ_{max} is the largest eigenvalue of the correlation matrix of the input data
- More practical test for stability is

$$0 < \mu < \frac{2}{\text{input signal power}}$$

- Larger values for step size
 - Increases adaptation rate (faster adaptation)
 - Increases residual mean-squared error