

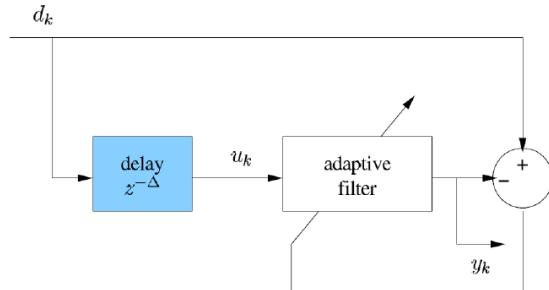
Examples of Quiz 6 questions

* In what kind of situations adaptive filtering is needed?

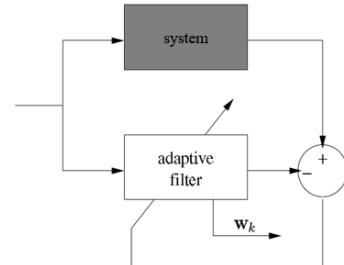
* What causality requirement must hold when adaptive filtering is applied?

* How can the LMS filter be applied to achieve active noise cancellation in headphones?

* For what purpose is the filtering configuration shown on the right used for?

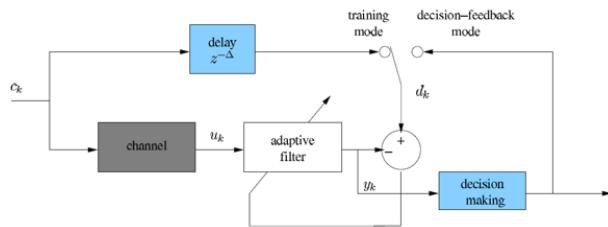


* What is the purpose of the adaptive filtering configuration shown on the right?



Output w_k = filter coefficients

* Explain how channel equalization configuration shown below works.



* Sketch the structure of adaptive LMS filter and describe how it works.

* In an adaptive filter, the FIR filter has 3 coefficients. At time $k = 0$, all coefficients are equal to zero, that is, $\mathbf{w}(0) = [0, 0, 0]$. Also, the buffer for the reference input contains initially zeros. The first values at the reference input are $u(0) = 1, u(1) = -1, u(2) = 3, u(3) = -1$, and $u(4) = 2$. The corresponding values at the primary input are $d(0) = -1, d(1) = 2, d(2) = 1, d(3) = 2$, and $d(4) = -1$. Compute the FIR filter coefficient vectors $\mathbf{w}(k)$ ($k=1, \dots, 5$) to the table below using the LMS algorithm. As a reminder, the coefficient changes in LMS are computed using $2\mu\mathbf{u}(k)\mathbf{e}(k)$. Use the step-size $\mu = 0.1$.

k	$\mathbf{w}(k)$
0	$[0, 0, 0]$
1	
2	
3	
4	
5	

* How is the steepest descent solution to Wiener filtering related to LMS adaptive algorithm?

* What is least squares regression and how it is related to RLS algorithm?