

# Digital Communications and Laboratory

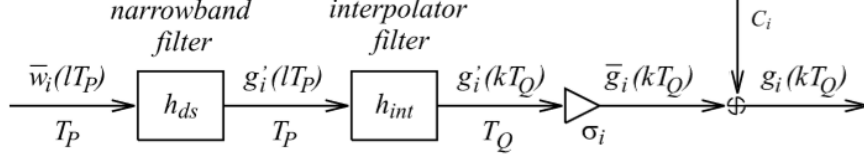
## Second Homework

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## Problem 1

## Problem 2

A flat fading channel with only one tap  $h_0(nT_c)$  was studied, assuming a *Rice factor* of  $k=2$  dB and normalized  $M_{h_0}$ . Moreover, a classical *Doppler Spectrum* with  $f_d T_c = 40 \cdot 10^{-5}$  was considered. The schematic model to generate the coefficient  $h_0$  of the channel is given in Figure 1.



**Figure 1.** Model to generate the coefficient  $h_0$  of the time-varying channel.

The Doppler Spectrum can be generated using a filter  $h_{ds}$  such that  $|\mathcal{H}_{ds}(f)|^2 = D(f)$ . In Table 1 are shown the coefficients used for such filter [1]:

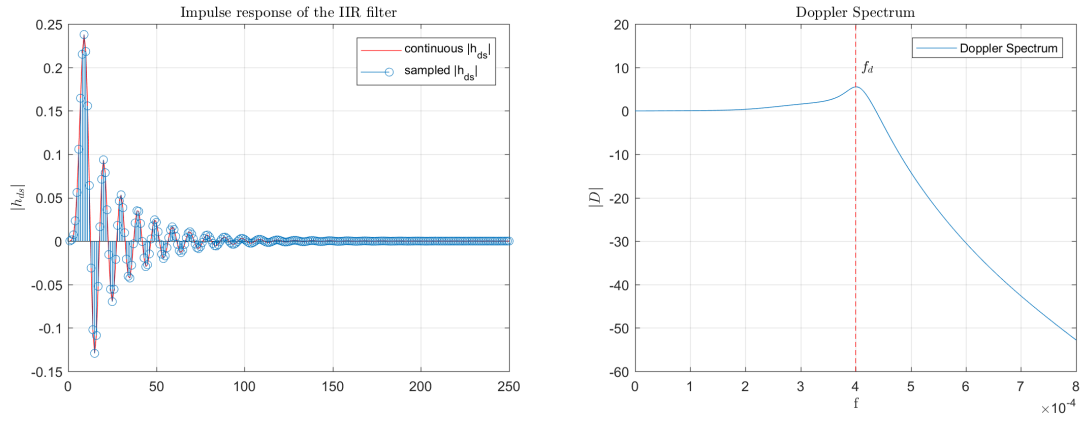
$H_{ds}(z) = B(z)/A(z)$		$f_d T_p = 0.1$	
$\{a_n\}$ ,	$n = 0, \dots, 11:$		
1	-4.4153	8.6283	-9.4592
6.1051	-1.3542	-3.3622	7.2390
-7.9361	5.1221	-1.8401	2.8706e-1
$\{b_n\}$ ,	$n = 0, \dots, 21:$		
1.3651e-4	8.1905e-4	2.0476e-3	2.7302e-3
2.0476e-3	9.0939e-4	6.7852e-4	1.3550e-3
1.8076e-3	1.3550e-3	5.3726e-4	6.1818e-5
-7.1294e-5	-9.5058e-5	-7.1294e-5	-2.5505e-5
1.3321e-5	4.5186e-5	6.0248e-5	4.5186e-5
1.8074e-5	3.0124e-6		

**Table 1.** Coefficients for the IIR filter

The graphical representation of the impulse response of the IIR filter and the Doppler Spectrum is shown in Fig. 2. To obtain  $h_0$ , following the scheme of Fig. 1, the noise component  $w \sim \mathcal{CN}(0, 1)$  was filtered with the IIR filter previously described. Note that the frequency response of this filter is  $\mathcal{H}_{ds}(f) = \sqrt{\mathcal{D}(f)}$  while the PSD of the noise is constant and equal to 1. For this reason, the equivalent impulse response of this part is equal to  $\mathcal{D}(f) = 1 \cdot |\mathcal{H}_{ds}|^2$  which is actually the Doppler spectrum.

The output of the filter was affected by a transient, which we have avoided by considering only values after  $5N_{eq}T_p$ , where  $N_{eq} = \left\lceil -\frac{1}{\ln(|p|)} \right\rceil$  is the equivalent time constant, and  $p$  is the pole with the highest magnitude. Then, after having scaled the coefficient such that  $M_{h_0}/\sqrt{E_{h_{ds}}} = 1$ , the signal was filtered with an interpolation filter of factor  $1/T_Q = T_p/T_c = 250$ . The interpolator

output signal was then multiplied by a constant  $\sigma_0 = \sqrt{M_d}$  to impose the desired power delay profile, and finally added up with another constant,  $C$ , which included the deterministic component according to [1], Page NUMERO PAGINA.



**Figure 2.** Impulse response of the IIR filter and Doppler Spectrum

# Bibliography

- [1] Nevio Benvenuto, Giovanni Cherubini, *Algorithms for Communication Systems and their Applications*. Wiley, 2002.