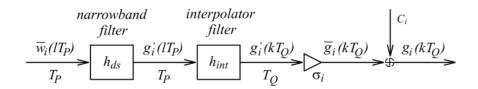
## Digital Communications and Laboratory Second Homework

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## Problem 1

## Problem 2

A flat fading channel with only one top  $h_0(nT_c)$  was studied, assuming a *Rice factor* of k=2 dB and normalized  $M_{h_0}$ . Moreover, a classical *Doppler Spectrum* with  $f_dT_c = 40 \cdot 10^{-5}$  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
$\overline{\{b_n\}}$ ,	$n=0,\ldots,21$ :		
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  $H_{ds}(f) = \sqrt{D(f)}$  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  $D(f) = 1 \cdot |H_{ds}|^2$  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.

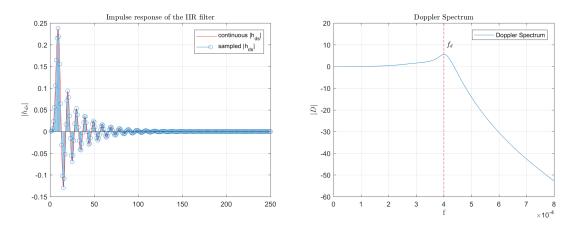


Figure 2. Impulse response of the IIR filter and Doppler Spectrum

## Bibliography

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