Entry test example questions

- 1. $x_a(t) = cos(2\pi f_a t)$ was sampled with sampling period T_s . Find normalized frequency, normalized angular frequency θ or period of the sampled signal. Plot the (sampled) signal spectrum. $(f_a, T_s \text{ or } f_s \text{ given, their proportion rational or irrational...})$
- 2. An analog signal with spectrum extending from $-f_a$ to $+f_a$ has been sampled with a sampling period T_s (or frequency f_s). Plot the spectrum after sampling (different values of f_a w.r.t. f_s)

Matlab notes

For help, use help <subject>, note that UPPERCASE is used to mark keywords in help only, not in real usage in Matlab....

An exception is in the scripts developed for this lab - their names ARE uppercase.

For sampling a real, analog signal with an A/D converter use Matlab command: y=GETDATA(Nsamples_in_block, [Kblocks, [Tsampling, [leave_bias]]])

Tsampling is in seconds, "[]" denote optional arguments.

We usually use Kblocks=1. Note that actual sampling frequency will be taken from a small predefined set available with the used soundcard. Typically you can sample signals within ± 2.5 Volts.

For plotting DT signals, use markers (plot(n,x,'o') or '-o'). For their continuous couterparts, use lines.

Exercises

Italics denote optional tasks. Bold suggests what should be in the report

- 1. NOT using Matlab, **plot** (with a pen or pencil) two periods of sine wave with f = 3200 Hz. Mark the points where the wave will be sampled at $f_s = 32 \ kHz$. Note number of samples per period.
- 2. Using Matlab sin() function, try to repeat the picture on screen plot. Finally extend the plot to 100 samples length (with the same parameters). (Then, show your result to the teacher.) In the report: copy the Matlab commands used to produce the data.
- 3. Applying sign(x+eps) to your signal x obtain a square wave and plot it. (hint: eps is added to avoid exact zero in x being converted to zero square wave is either +1 or -1).
- 4. Use A/D converter to get similar signals (sine and square wave) from a generator. Set amplitude to about 1 V. Describe differences between simulated and real-world plots.
- 5. Label an x-axis of the plot of real-world sine wave (in Matlab) with time units, then repeat with sample indices (hint: plot(xvalues, yvalues, 'marker');). In the report copy the Matlab commands used, and copy approx. two periods from the screen plot.
- 6. Sample a sinusoidal signal with much bigger amplitude (few volts), and with much smaller amplitude. See the effects of clipping and of noise.

- 7. Plot a simulated DT sinusoid with normalized frequency $f_n = f/f_s$ equal to 0.1, 0.3, 0.5, 0.9, 1.1, 2.1 ($\theta = 2\pi f_n$, $x(n) = sin(n\theta)$). Note number of samples in period. For plots @ f_n 0.1 and 1.1 draw a copy in your report, and explain why in some cases different f_n result in identical plots try to draw (by pencil) the underlying CT signal. If you are brave enough, draw the underlying CT signal with Matlab using 9 additional samples between original ones; show it to the teacher and/or copy Matlab expression to the report.
- 8. Sample a physical signal: 1.1 kHz sinusoid sampled at $f_s=32$ kHz, 1024 samples; plot it in Matlab using physical time as horizontal axis (create the time axis variable t). Then decimate signal by 16 (i.e. leave every 16th sample) or by 32.

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Hint: use xdecimated=x(1:16:end); tdecimated=t(1:16:end),
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Plot it: plot(t,x,tdecimated,xdecimated,'-x'). Note the resulting sampling frequency and describe undersampling effects.
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9. Keep the plot (or save in some variable the data to replot it again). Repeat the experiment with 0.1 kHz, 2.1 kHz etc. Compare the results with exercise 8.

Hint: read help on figure() command.

- 10. Create a set of test signals (all with length of 32 samples).
 - (a) A unit impulse signal $\delta(n)$ by using

```
N=32;
n=[-3:(N-4)];
dlt=(n==0);
stem(n,dlt);
```

(b) A unit step signal u(n):

```
ustep=(n>=0);
stem(n,ustep);
```

(c) Shifted unit impulse $\delta(n-K)$; use your table number as KYou can do it by adding some zeros at the left side of the samples vector and trimming the right side to the previous length (example shown for shift by 4)

```
dlt4=[zeros(1,4) dlt]; %add 4 zeros
dlt4=dlt4(1:N); %trim to N samples again
stem(n,dlt,'b');
hold on
stem(n,dlt4,'g');
hold off
```

- (d) A short impulse (e.g with 7 samples length) by subtraction of two unit steps (one shifted in time) or by clever use of ones() and zeros().
- 11. Use the prepared signals (δ(n), u(n), δ(n n₀), finite-time impulse) as x to test a linear system implemented by matlab command y=filter([1, 2, 1], [1 -.9],x); plot the resulting output signals y; try to notice how may the y depend on x.

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