RF Systems I

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Definitions & basic concepts

dB *t*-domain vs. ω-domain phasors

Decibel (dB)

Convenient logarithmic measure of a power ratio.
A "Bel" (= 10 dB) is defined as a power ratio of 10¹. Consequently, 1 dB is a power ratio of 10^{0.1}≈1.259
If *rdb* denotes the measure in dB, we have:

 $rdb = 10 \text{ dB} \log\left(\frac{P_2}{P_1}\right) = 10 \text{ dB} \log\left(\frac{A_2^2}{A_1^2}\right) = 20 \text{ dB} \log\left(\frac{A_2}{A_1}\right)$

 $\frac{P_2}{P_1} = \frac{A_2^2}{A_1^2} = 10^{rdb/(10 \text{ dB})}$

 $\frac{A_2}{A_2} = 10^{rdb/(20 \, \text{dB})}$

| rdb | -30 dB | -20 dB | -10 dB | -6 dB | -3 dB | o dB | 3 dB | 6 dB | 10 dB | 20 dB | 30 dB |
|---------------|--------|--------|--------|-------|-------|------|------|------|-------|-------|-------|
| P_{2}/P_{1} | 0.001 | 0.01 | 0.1 | 0.25 | .50 | 1 | 2 | 3.98 | 10 | 100 | 1000 |
| A_2/A_1 | 0.0316 | 0.1 | 0.316 | 0.50 | .71 | 1 | 1.41 | 2 | 3.16 | 10 | 31.6 |

• Related: dBm (relative to 1 mW), dBc (relative to carrier)

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Time domain – frequency domain (1)

- An arbitrary signal g(t) can be expressed in ω -domain using the *Fourier transform* (FT). $g(t) \rightarrow G(\omega) = \frac{1}{\sqrt{2\pi}} \int_{0}^{\infty} g(t)e^{j\omega t} dt$
- The inverse transform (IFT) is also referred to as **Fourier Integral** G(ω) ••• $g(t) = \frac{1}{\sqrt{2\pi}} \int_{-\infty}^{\infty} G(\omega) e^{-j\omega t} d\omega$
- The advantage of the ω-domain description is that linear time-invariant (LTI) systems are much easier described.
- The mathematics of the FT requires the extension of the definition of a *function* to allow for infinite values and nonconverging integrals.
 - The FT of the signal can be understood at looking at "what frequency components it's composed of".

Time domain – frequency domain (2)

- For *T*-periodic signals, the FT becomes the Fourier-Series, $d\omega$ becomes $2\pi/T$, \int becomes Σ .
- The cousin of the FT is the *Laplace transform*, which uses a complex variable (often s) instead of *j*ω; it has generally a better convergence behaviour.
- Numerical implementations of the FT require discretisation in *t* (sampling) and in ω. There exist very effective algorithms (FFT).
- In digital signal processing, one often uses the related z-Transform, which uses the variable $z = e^{j\omega\tau}$, where τ is the sampling period. A delay of $k\tau$ becomes z^{-k} .



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Fixed frequency oscillation (steady state, CW) Definition of phasors

• General: $A \cos(\omega t - \varphi) = A \cos \omega t \cos \varphi + A \sin \omega t \sin \varphi$

This can be interpreted as the projection on the real axis of a circular motion in the complex plane. Re {A(cos φ + j sin φ)e^{jωt}}

The complex amplitude \tilde{A} is called "phasor";



 $\tilde{A} = A(\cos\varphi + j\sin\varphi)$

Calculus with phasors

- Why this seeming "complication"?: Because things become easier!
- Using $\frac{d}{dt} \equiv j \omega$, one may now forget about the rotation with ω and the projection on the real axis, and do the complete analysis making use of complex algebra!



Slowly varying amplitudes

• For band-limited signals, one may conveniently use "slowly varying" phasors and a fixed frequency RF oscillation.

• So-called in-phase (I) and quadrature (Q) "baseband envelopes" of a modulated RF carrier are the real and imaginary part of a slowly varying phasor.

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On Modulation

AM PM I-Q

Amplitude modulation

$$(1 + m\cos(\varphi)) \cdot \cos(\omega_c t) = \operatorname{Re}\left\{ \left(1 + \frac{m}{2}e^{j\varphi} + \frac{m}{2}e^{-j\varphi}\right)e^{j\omega_c t} \right\}$$



black: sidebands at $\pm f_m$ blue: sum

m: modulation index or modulation depth example: $\varphi = \omega_m t = 0.05 \omega_c t$ m = 0.5



Phase modulation



$$\operatorname{Re}\left\{e^{j\omega_{c}t+M\sin(\varphi)}\right\} = \operatorname{Re}\left\{\sum_{n=-\infty}^{\infty}J_{n}(M)e^{j(n\varphi+\omega_{c}t)}\right\}$$

M: modulation index (= max. phase deviation)

 $\dot{M} = 1$

 $\varphi = \omega_m t = 0.05 \,\omega_c t$ M = 4



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Spectrum of phase modulation

Plotted: spectral lines for sinusoidal PM at f_m Abscissa: $(f-f_c)/f_m$



Phase modulation with $M=\pi$: red: real phase modulation blue: sum of sidebands $n\leq 3$



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Spectrum of a beam with synchrotron oscillation, M = 1 (=57°)



Vector (I-Q) modulation





I-Q modulation: green: *I* component red: *Q* component blue: vector-sum More generally, a modulation can have both amplitude and phase modulating components. They can be described as the in-phase (I) and quadrature (Q) components in a chosen reference, $cos(\omega_r t)$. In complex notation, the modulated RF is:

 $\operatorname{Re}\left\{\left(I(t) + j Q(t)\right)e^{j \omega_{r} t}\right\} = \\\operatorname{Re}\left\{\left(I(t) + j Q(t)\right)(\cos(\omega_{r} t) + j \sin(\omega_{r} t))\right\} = \\I(t)\cos(\omega_{r} t) - Q(t)\sin(\omega_{r} t)$

So *I* and *Q* are the Cartesian coordinates in the complex "Phasor" plane, where amplitude and phase are the corresponding polar coordinates. $I(t) = A(t) \cos(\varphi)$ $Q(t) = A(t) \sin(\varphi)$

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Vector modulator/demodulator



Digital Signal Processing

Just some basics

Sampling and quantization

- Digital Signal Processing is very powerful note recent progress in digital audio, video and communication!
- Concepts and modules developed for a huge market; highly sophisticated modules available "off the shelf".
- The "slowly varying" phasors are ideal to be sampled and quantized as needed for digital signal processing.
- Sampling (at $1/\tau_s$) and quantization (*n* bit data words here 4 bit):



Digital filters (1)

- Once in the digital realm, signal processing becomes "computing"!
- In a "finite impulse response" (FIR) filter, you directly program the coefficients of the impulse response.





 $Z = e^{j\omega \tau_s}$

Digital filters (2)

• An "infinite impulse response" (IIR) filter has built-in recursion, e.g. like



Transfer function: $\frac{b_0 + b_1 z^{-1} + b_2 z^{-2}}{1 + a_1 z^{-1} + a_2 z^{-2}}$



... is a comb filter

Digital LLRF building blocks – examples

 General D-LLRF board:
 modular!
 FPGA: Field-programmable gate array DSP: Digital Signal Processor



 DDC (Digital Down Converter)
 Digital version of the I-Q demodulator
 CIC: cascaded integrator-comb (a special low-pass filter)



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RF system & control loops

e.g.: ... for a synchrotron: Cavity control loops Beam control loops

Minimal RF system (of a synchrotron)

Low-level RF

High-Power RF



- The frequency has to be controlled to follow the magnetic field such that the beam remains in the centre of the vacuum chamber.
- The voltage has to be controlled to allow for capture at injection, a correct bucket area during acceleration, matching before ejection; phase may have to be controlled for transition crossing and for synchronisation before ejection.

Fast RF Feed-back loop



- Compares actual RF voltage and phase with desired and corrects.
- Rapidity limited by total group delay (path lengths) (some 100 ns).
- Unstable if loop gain =1 with total phase shift 180 ° design requires to stay away from this point (stability margin)
- The group delay limits the gain bandwidth product.
- Works also to keep voltage at zero for strong beam loading, i.e. it reduces the beam impedance.

Fast feedback loop at work



- Gap voltage is stabilised!
- Impedance seen by the beam is reduced by the loop gain!
- Plot on the right: $\frac{1+\beta}{R} \left| \frac{Z(\omega)}{1+G \cdot Z(\omega)} \right|$ vs. ω

with the loop gain varying from 0 to 50 dB.

• Without feedback, $V_{acc} = (I_{G0} + I_B) \cdot Z(\omega)$ where $Z(\omega) = \frac{R(1 + \beta)}{1 + jQ\left(\frac{\omega}{\omega_0} - \frac{\omega_0}{\omega}\right)}$ • Detect the gap voltage, feed it back to I_{G0} such that $I_{G0} = I_{drive} - G \cdot V_{acc}$

where *G* is the total loop gain (pick-up, cable, amplifier chain ...) • Result: $V_{acc} = (I_{drive} + I_B) \cdot \frac{Z(\omega)}{1 + G \cdot Z(\omega)}$



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1-turn delay feed-back loop

- The speed of the "fast RF feedback" is limited by the group delay this is typically a significant fraction of the revolution period.
- How to lower the impedance over many harmonics of the revolution frequency?
- Remember: the beam spectrum is limited to relatively narrow bands around the multiples of the revolution frequency!
- Only in these narrow bands the loop gain must be high!
- Install a comb filter! ... and extend the group delay to exactly I turn – in this case the loop will have the desired effect and remain stable!





Field amplitude control loop (AVC)



• Compares the detected cavity voltage to the voltage program. The error signal serves to correct the amplitude

Tuning loop



- Tunes the resonance *f* of the cavity to minimize the mismatch of the PA.
- In the presence of beam loading, this may mean $f_r \neq f$.
- In an ion ring accelerator, the tuning range might be > octave!
- For fixed *f* systems, tuners are needed to compensate for slow drifts.
- Examples for tuners:
 - controlled power supply driving ferrite bias (varying μ),
 - stepping motor driven plunger,
 - motorized variable capacitor, ...

Beam phase loop



Longitudinal motion: •

$$\frac{I^2(\Delta\phi)}{dt^2} + \Omega_s^2(\Delta\phi)^2 = 0$$

Loop amplifier transfer function designed to damp synchrotron oscillation. Modified equation:

$$rac{d^2(\Delta\phi)}{dt^2} + lpha \, rac{d(\Delta\phi)}{dt} + \Omega_s^2(\Delta\phi)^2 = 0$$



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Other loops

• Radial loop: Detect average radial position of the beam, Compare to a programmed radial position, • Error signal controls the frequency. • Synchronisation loop (e.g. before ejection): • 1st step: Synchronize f to an external frequency (will also act on radial position.). • 2nd step: phase loop brings bunches to correct position.

A real implementation: LHC LLRF



Fields in a waveguide

Homogeneous plane wave

 $\vec{E} \propto \vec{u}_{y} \cos\left(\omega t - \vec{k} \cdot \vec{r}\right)$ $\vec{B} \propto \vec{u}_{x} \cos\left(\omega t - \vec{k} \cdot \vec{r}\right)$ $\vec{k} \cdot \vec{r} = \frac{\omega}{c} \left(\cos(\varphi)z + \sin(\varphi)x\right)$

Wave vector k: the direction of \overline{k} is the direction of propagation, the length of \overline{k} is the phase shift per unit length. \overline{k} behaves like a vector.





Z.

Wave length, phase velocity

• The components of \overline{k} are related to the wavelength in the direction of that component as $\lambda_z = \frac{2\pi}{k}$ etc., to the phase velocity as $v_{\varphi,z} = \frac{\omega}{k} = f \lambda_z$.

 $k_{\perp} = \frac{\omega_c}{c}$





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 $\int k = \frac{\omega}{c}$

 $\Rightarrow k_z = \frac{\omega}{\alpha} \left[1 - \left(\frac{\omega_c}{\omega} \right)^2 \right]$

 $k = \omega$

Superposition of 2 homogeneous plane waves





Metallic walls may be inserted where $E_y = 0$ without perturbing the fields. Note the standing wave in *x*-direction!

This way one gets a hollow rectangular waveguide! CAS Granada - EJ: RF Systems I Rectangular waveguide Fundamental (TE_{10} or H_{10}) mode in a standard rectangular waveguide. E.g. forward wave

electric field

power flow: $\frac{1}{2} \operatorname{Re} \{ \iint \vec{E} \times \vec{H}^* dA \}$









Waveguide dispersion

What happens with different waveguide dimensions (different width a)? The "guided wavelength" λ_g varies from ∞ at f_c to λ at very high frequencies.





Phase velocity $v_{\varphi,z}$

The phase velocity is the speed with which the crest or a zero-crossing travels in *z*-direction. Note in the animations on the right that, at constant *f*, it is $v_{\varphi,z} \propto \lambda_g$. Note that at $f = f_c$, $v_{\varphi,z} = \infty$! With $f \to \infty$, $v_{\varphi,z} \to c$!





Radial waves

- Also radial waves may be interpreted as superpositions of plane waves.
- The superposition of an outward and an inward radial wave can result in the field of a round hollow waveguide.



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Round waveguide modes

a/mm

parameters used in calculation: $f = 1.43, 1.09, 1.13 f_c, a$: radius



GHz

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GHz

<u>a</u>/mm

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a/mm

From waveguide to cavity

Standing wave – resonator

Same as above, but two counter-running waves of identical amplitude.

electric field

No net power flow: $\frac{1}{2} \operatorname{Re} \{ \iint \vec{E} \times \vec{H}^* dA \} = 0$







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magnetic field (90° out of phase)

A piece of round waveguide – pillbox cavity TM₀₁₀-mode (only 1/8 shown)







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Pillbox cavity field (w/o beam tube) The only non-vanishing field components : $E_{z} = \frac{1}{j\omega\varepsilon_{0}} \frac{\chi_{01}}{a} \sqrt{\frac{1}{\pi}} \frac{J_{0}\left(\frac{\chi_{01}\rho}{a}\right)}{aJ_{1}\left(\frac{\chi_{01}}{a}\right)}$ $B_{\varphi} = \mu_0 \sqrt{\frac{1}{\pi}} \frac{J_1\left(\frac{\chi_{01}\rho}{a}\right)}{a J_1\left(\frac{\chi_{01}}{a}\right)}$ $\omega_0\Big|_{pillbox} = \frac{\chi_{01} c}{\alpha}$ $\chi_{01} = 2.40483...$

Accelerating gap

Accelerating Gap



- We want a voltage across the gap!
- It cannot be DC, since we want the beam tube on ground potential.
- Use $\oint \vec{E} \, d\vec{s} = -\iint \frac{dB}{dt} d\vec{A}$
 - The "shield" imposes a upper limit of the voltage pulse duration or — equivalently a lower limit to the usable frequency.
 - The limit can be extended with a material which acts as "open circuit"!
 - Materials typically used:
 - ferrites (depending on *f*-range) magnetic alloys (MA) like Metglas®, Finemet®, Vitrovac®...
 - resonantly driven with RF (ferrite loaded cavities) – or with pulses (induction cell).

Linear induction accelerator

Linear induction accelerator

Acceleration gap

Induction

accelerating cell

$$\oint \vec{E} \cdot d\vec{s} = -\iint \frac{\partial \vec{B}}{\partial t} \cdot d\vec{A}$$

compare: transformer, secondary = beam Acc. voltage during B

ramp.

Beam current

Ferromagnetic cores (high inductive impedance)

Ferrite cavity

PS Booster, '98 0.6 – 1.8 MHz, < 10 kV gap NiZn ferrites

Gap of PS cavity (prototype)



Drift Tube Linac (DTL) – how it works

aka Alvarez*)

For slow particles – protons @ few MeV e.g. the drift tube lengths can easily be adapted.

electric field







*) not Marc, but Luis Walter

Drift tube linac – practical implementations



Transit time factor

The transit time factor is the ratio of the acceleration voltage to the (non-physical) voltage a particle with infinite velocity would see.

$$TT = \frac{|V_{acc}|}{\left|\int E_z dz\right|} = \frac{\left|\int E_z e^{j\,\omega z} dz\right|}{\int E_z dz}$$

The transit time factor of an ideal pillbox cavity (no axial field dependence) of height (gap length) h is:

 $TT = \sin\left(\frac{\chi_{01}h}{2a}\right) / \left(\frac{\chi_{01}h}{2a}\right)$

Field rotates by 360° during particle passage.

 h/λ

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