

TUNE COMPUTATION VIA MODEL FITTING TO SWEPT MACHINE RESPONSE MEASUREMENT

M.G. Abbott*, G. Rehm, Diamond Light Source, UK

Abstract

At Diamond Light Source we compute the horizontal and vertical tunes by fitting a simple multi-pole resonator model to the measured electron beam frequency response. The transverse (and longitudinal) tune response is measured by sweeping an excitation across the range of possible tune frequencies and synchronously measuring the IQ response.

The multi-pole resonator model is a good fit to the measured behaviour, but the fitting process is surprisingly challenging. Problems include noisy measurements, very complex beam responses in the presence of increasing chromaticity, poor data when the beam is close to instability, and a number of challenges with the stability of the algorithm.

The tune fitting algorithm now in use at Diamond has been developed and refined over many years. It is finally stable enough to work reliably throughout most beam operating conditions. The algorithm involves alternating peak finding and non-linear fitting, with a fairly naive mathematical approach; the main focus is on providing reliable results.

INTRODUCTION

The synchrotron beam has natural frequencies of oscillation in the horizontal, vertical, and longitudinal directions: transverse “betatron tunes” and longitudinal “synchrotron tunes”. The precise betatron tune frequencies are of considerable interest to machine physicists, and need to be measured at various stages during machine operation.

At Diamond Light Source the storage ring configuration has evolved into a state where movement of the tunes (driven mainly by insertion device movements affecting machine optics) can result in loss of beam lifetime and injection efficiency, so it is necessary to actively measure and correct the betatron tunes in both transverse axes. This therefore requires a reliable measurement of the tunes, during both machine startup conditions and normal operation.

A complicating factor when computing the tunes arises from interference between transverse and longitudinal oscillations, depending on the chromaticity settings of the synchrotron. This interaction can result in significant side-lobes at (roughly) multiples of synchrotron frequency offset either side of the main tune measurement. Depending on machine conditions it can become difficult to identify the central tune frequency. This is addressed by the process described here, but there remain some operating conditions that can be surprisingly difficult to interpret.

We are able to take advantage of phase and magnitude measurement to fit a reasonably sophisticated model.

* michael.abbott@diamond.ac.uk

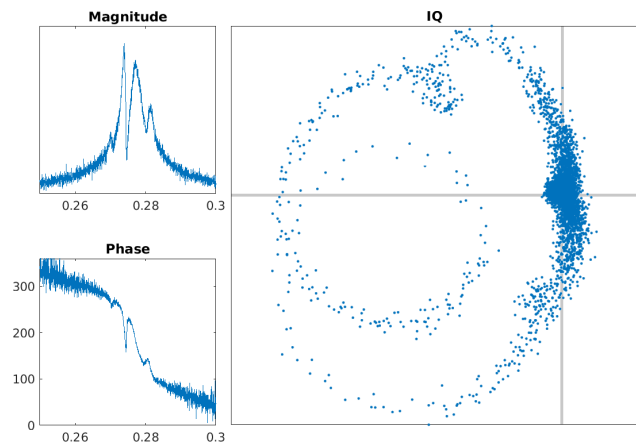


Figure 1: Illustration of swept response, showing magnitude (in arbitrary units) and phase (in degrees) against fractional tune, and the corresponding complex IQ measurements. This sweep shows a typically complex response with multiple lobes, illustrating the problem of identifying the true tune in the presence of large synchrotron side-lobes.

RESPONSE MEASUREMENT

Tune measurement at Diamond is integrated into the operation of the Multi-Bunch Feedback (MBF) system [1], and is done by exciting the beam with a swept sinusoidal oscillation and synchronously measuring the response. The result is a complex number $z(\omega)$ at each sampled frequency ω representing the phase and magnitude of the machine response, computed thus:

$$z(\omega) = \sum_{t \in \text{dwell}(\omega)} e^{-iK\omega t} x(t)$$

where $\text{dwell}(\omega)$ is typically 100 turns per frequency step, and K is a frequency scaling factor.

Because measurement and stimulus are both confined to a single location in the machine, we are only able to see the fractional part of the machine tune, but in practice this is the only part that needs to be measured. For convenience, all frequencies ω are scaled to fractions of machine revolution frequency.

When using the MBF system for tune sweeping we have a number of options, including which bunches to excite and measure, which overall phase advance between bunches to apply (this is referred to as the “mode”), strength of excitation, and the dwell time at each frequency. At present we excite and measure all bunches at a mode of 80, and typically sweep 4096 points of a frequency range of 0.1 around the nominal tune point over a period of about 780 ms. From this the tune measurement is updated at just over 1 Hz.

A typical measurement is shown in Fig. 1.

Content from this work may be used under the terms of the CC BY 3.0 licence (© 2019). Any distribution of this work must maintain attribution to the author(s), title of the work, publisher, and DOI

MODELLING MEASURED RESPONSE

The swept response is modelled by fitting a simple multi-pole resonator model. Starting with the motivating example of a single narrow band resonator, this can be described by the following differential equation (with driving term y , resonator bandwidth ν , nominal centre frequency ω_0):

$$y = \ddot{x} + 2\nu\dot{x} + \omega_0^2 x .$$

The Laplace transform of this gives us the following equation (where X, Y are the Laplace transforms of x, y , and defining $\omega_c^2 \equiv \omega_0^2 - \nu^2$):

$$Y = (s^2 + 2\nu s + \omega_0^2)X = ((s + \nu)^2 + \omega_c^2)X .$$

Noting now that the quadratic term in s has zeros at $s = -\nu \pm i\omega_c$, the corresponding system response X/Y can be written as (writing $a_0 \equiv 1/2i\omega_c$, $b_0 \equiv -\nu + i\omega_c$):

$$\frac{X}{Y} = \frac{1}{(s - b_0)(s - b_0^*)} = \frac{a_0}{s - b_0} - \frac{a_0}{s - b_0^*} .$$

At this point we can make some simplifying assumptions: the measured swept response covers a narrow range around ω_c , and $\nu \ll \omega_c$ (i.e., our Q factor is large). In this case we can ignore the $a_0/(s - b_0^*)$ term and write our model M as:

$$M(\omega) = \frac{X}{Y}(i\omega) \approx \frac{a_0}{i\omega - b_0} = \frac{a}{\omega - b}$$

where $a \equiv -ia_0$ and $b \equiv -ib_0 = \omega_c + i\nu$. We work with this latter form, which we now regard as a “single pole resonator”, and we allow a to be a free parameter to represent other sources of scaling and phase variation which we will want to capture.

Ignoring the small offset between damped frequency ω_c and nominal frequency ω_0 , the following parameters can be computed from a single pole (a, b) :

- centre frequency = $\text{re}(b)$
- peak width = $\text{im}(b)$
- peak “power” $\int_{\mathbb{R}} |M(\omega)|^2 d\omega = |a|^2 / \text{im}(b)$
- phase = $\angle ia$

Note that the peak area $\int_{\mathbb{R}} |M(\omega)| d\omega$ diverges, and the “power” (integral of magnitude squared) above is a purely formal quantity, in particular peak powers cannot meaningfully be added together.

Finally we model the full multi-pole resonator model as a sum of N single pole resonators $(a_n, b_n)_{n \in 1..N}$, plus a correction factor c to capture any background offset:

$$M(\omega) = \sum_{n=1}^N \frac{a_n}{\omega - b_n} + c .$$

This is a rational function in ω , which is a natural class of functions in control theory, and this simple model turns out to be a remarkably good fit to measured tune sweeps.

PEAK FITTING ALGORITHM

The input to the tune fitting algorithm is a pair of equal sized waveforms $(\vec{\omega}, \vec{z})$, where $\vec{\omega}$ is the swept frequency scale and \vec{z} is the measured frequency response for each corresponding frequency as a complex number.

One peak at a time is fitted to the measured data until a fitting step fails, or until a configured number of peaks has been reached. The peak fitting algorithm consists of the following steps:

1. Set initial model $M := M_0$ with $N = 0, c = 0$.
2. Compute residue $\vec{r} = \vec{z} - M(\vec{\omega})$.
3. Find largest peak in $|\vec{r}|^2$ after smoothing, find region around this peak.
4. Compute an initial (a, b) fit to \vec{r} around discovered peak and extend model with $M' = M + (a, b)$.
5. Compute refined model M'' from initial model M' using Levenberg-Marquardt optimisation
6. Assess M'' :
 - If M'' fails assessment then exit with current model M .
 - If $N + 1 = N_{\max}$ then exit with new model M'' .
 - Otherwise assign $M := M''$, increment N , go back to step 2.

At this point the peaks from the successful model are sorted into ascending order of centre frequency and are handed over to the final tune computation stage.

Peak Discovery and Initial Fit

Figure 2 shows the initial step of peak discovery. The residue \vec{r} is smoothed and decimated (we simply average in bins of 32 samples), the second derivative is computed,

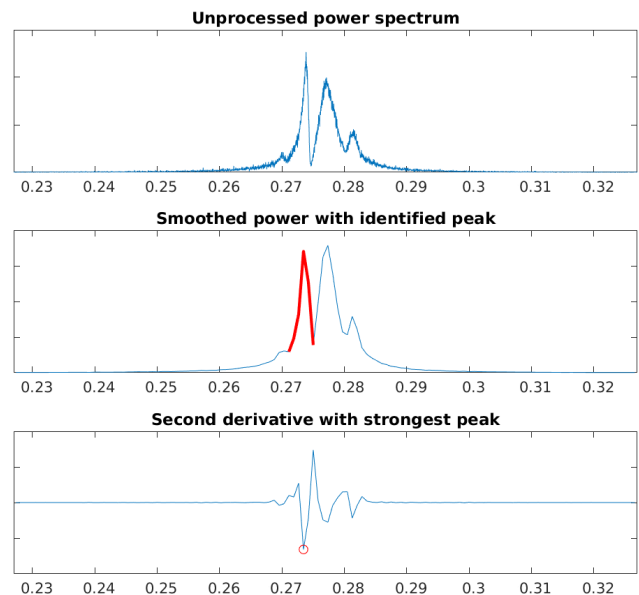


Figure 2: Initial peak discovery: the power spectrum is smoothed and decimated, and the strongest peak is identified by finding the frequency with the strongest curvature (shown circled in red at bottom). This is then used to find the range for the initial fit (highlighted in red in the centre graph).

Content from this work may be used under the terms of the CC BY 3.0 licence (© 2019). Any distribution of this work must maintain attribution to the author(s), title of the work, publisher, and DOI

and the most negative point (point of strongest curvature) is taken as the centre of the next peak to be fitted. This starting point is then used to discover an interval I spanning the discovered peak.

Next it is important to create an initial fit to this peak to properly prime the refinement step. We would like to minimise the error term $\sum_{i \in I} |r_i - a/(\omega_i - b)|^2$ over the interval I , but this is a non-linear problem. Multiplying through by $\vec{\omega} - b$ and a weighting term \vec{w} we get the minimisation problem

$$\text{minimise } \sum_{i \in I} |w_i(a + br_i - r\omega)|^2$$

which is linear in a and b and can therefore be solved directly. We set $w_i \equiv |r_i|^2$ to partially cancel out the change in weighting and to improve focusing of the fit on the centre of the peak.

This newly fitted peak is added to the model, the c term is recomputed, and the model is refined with the next step.

Levenberg-Marquardt Optimisation

The Levenberg-Marquardt (LM) algorithm [2–4] is an iterative non-linear least squares minimisation process. This algorithm takes an initial starting model and refines it iteratively to reduce the fitting error. We run a few rounds of this algorithm after adding each new peak to improve the fit.

The algorithm acts on the measured machine response $(\vec{\omega}, \vec{z})$ and parameters $\beta = ((a_n, b_n)_{n \in 1..N}, c)$ defining the model $M(\omega) = M(\beta; \omega)$ to be refined. Each step of the algorithm computes a new $\beta' = \text{LM}(\beta; \vec{\omega}, \vec{z})$ to reduce the error term

$$\chi^2(\beta) = \sum_i \frac{|z_i - M(\beta; \omega_i)|^2}{\sigma_i^2}$$

Ideally σ_i should be the measurement error for \vec{z}_i and should be used to guide termination of the LM process, but at present we ignore this term. An estimate for σ can be probably be recovered from the tails of the swept response, but this has not yet been investigated.

Computing β' requires the calculation of the partial derivative terms $\partial M(\beta; \omega)/\partial \beta$, but fortunately these are easy to compute:

$$\frac{\partial M}{\partial a_n} = \frac{1}{\omega - b_n}, \quad \frac{\partial M}{\partial b_n} = \frac{a_n}{(\omega - b_n)^2}, \quad \frac{\partial M}{\partial c} = 1.$$

Behaviour of LM Refinement This algorithm is very sensitive to initial conditions, and can produce some unhelpful results: the biggest problem is a tendency for poles to wander and interfere with one another. This problem becomes more pronounced as more poles are added, and in general it is difficult to fit all three peaks reliably.

Figure 3 shows the process of successful incremental peak fitting and refinement. Figure 4 shows the more complex case of failing when adding a third peak: in this case we see the importance of the initial conditions.

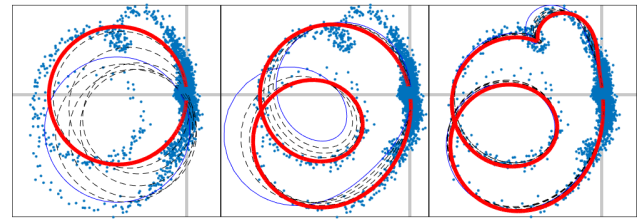


Figure 3: Successful model refinement: fitting three poles one by one. The red curve shows the result after refinement, the thin blue line is the initial fit, and the dashed lines show intermediate stages. It is possible to see from this figure that the first fitted pole moves from one lobe to another.

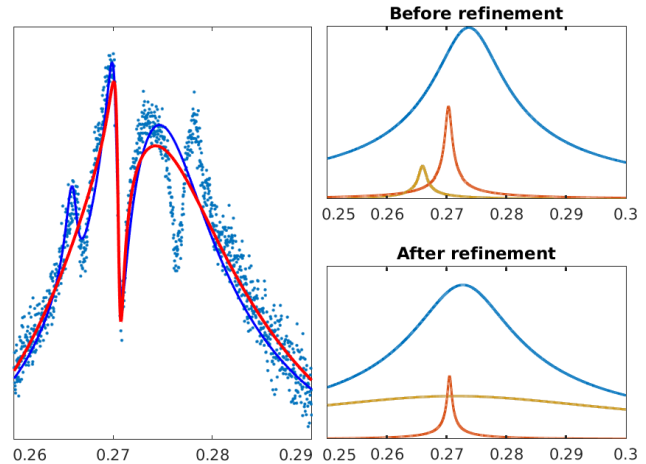


Figure 4: Failed model refinement. On the left the blue curve shows the initial fit, and the top right shows the three separate components of this fit: we can see that the small peak has been chosen. On bottom right this peak has moved and broadened to join the central large peak: this fit is rejected by the model assessment step.

Model Assessment

It is important to evaluate the success of the tune fitting process, so after each LM optimisation step the entire model is assessed. This involves the following checks, if any of them fail the entire fit is discarded:

- Peak width. The peak width is checked against configured minimum and maximum limits. This catches peaks that have run away into the background (very broad peaks), peaks that have got caught on local noise (very narrow peaks), and peaks that have gone negative.
- Peak position. Occasionally the centre of a peak will wander out of the swept data window, and so must be treated as invalid.
- Peak closeness. If two peaks are too close together this is a clear sign of peak merging, and so the model can be discarded.
- Peak merging. Peaks with opposing phase are also a sign of peak merging and cause the model to be rejected.

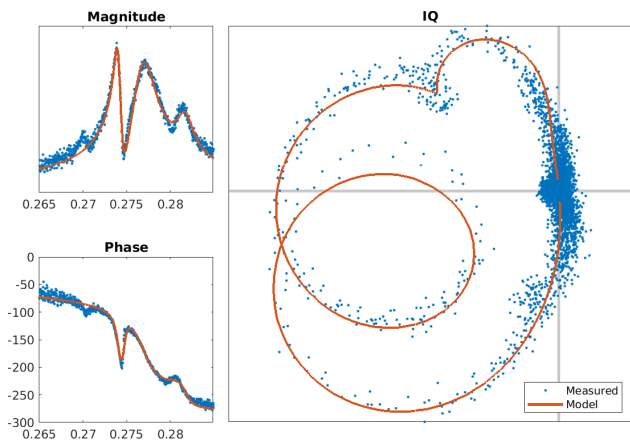


Figure 5: This figure show a successful fit of a three pole model to a typical tune sweep. The two main peaks are fitted very closely, the third peak looks slightly off, while there is a fourth peak which has not been attempted. From this model we can measure a tune of 0.2768 with a phase of 159° (seen through the MBF feedback filter).

Computing the Tune

The original motivation for fitting as many peaks as possible was to find a way to reliably identify the true “tune”. In practice it seems that simply taking the peak with the largest “power” is sufficient. From the model we can now read out the tune, its phase and bandwidth or damping factor, and the synchrotron side bands, if they were successfully fitted.

Figure 5 shows the final result of a successful fit to the data captured in Fig. 1 and refined in Fig. 3.

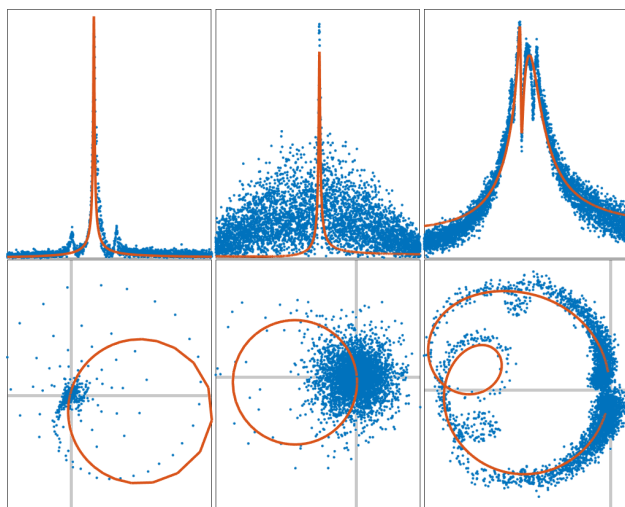


Figure 6: Three challenging fits: magnitude and IQ shown. The first sweep shows strong ringing from sweeping a narrow resonance too quickly, but the fit is successful. The second sweep shows the limit of tune detection, but is again successful. The third sweep shows failure to fit the third important lobe: in this case feedback is flattening the tune response.

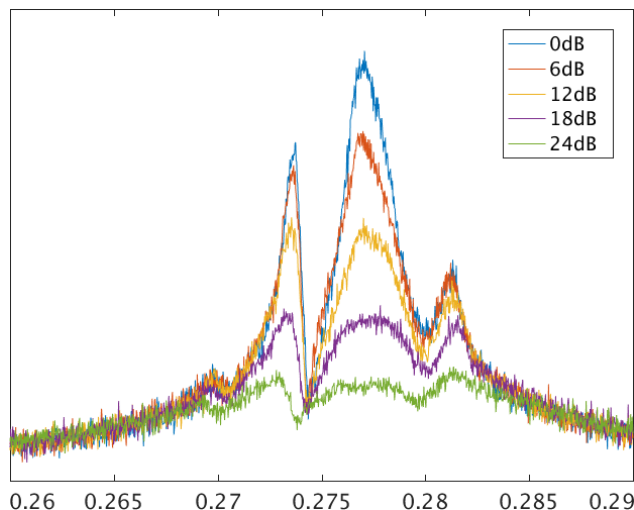


Figure 7: Tune sweeping measurement is normally combined with bunch-by-bunch feedback. This figure shows the impact of the feedback intensity on the tune sweep: increasing feedback flattens the response and makes the fit less reliable.

CHALLENGES

The main challenge has been to develop an algorithm that is robust enough to produce acceptable results throughout the entire operating range of the synchrotron. In particular, tune sweeps look completely different at low and high beam currents, and high chromaticity enhances the synchrotron sidebands.

At very low beam currents the tune resonance can become extremely narrow. The main impact of this is “dragging” of the resonance with the sweep, producing extra oscillations on the swept response. This can be avoided with a narrower or slower sweep.

A small selection from some more challenging sweeps is shown in Fig. 6. In all three cases the data can be improved by changing the sweep conditions. Figure 7 shows the impact of correcting for multi-bunch instabilities on response measurement.

CONCLUSIONS

This algorithm has been developed and has evolved over many years. Throughout the main problem has been to distinguish the true “tune” peak from its neighbours. Now with a reasonably reliable algorithm producing a detailed and accurate fit (when successful) we are able to measure the tune for long periods without “hopping” to adjacent peaks.

Problems still remain with challenging machine conditions, particularly when very strong bunch-by-bunch feedback is required, when operating at extremely low beam currents, or at very high chromaticities. However for normal operation this algorithm is very reliable and is able to operate unattended for long periods.

The code for the tune fitting algorithm is written in Python and can be obtained by contacting the author.

REFERENCES

- [1] M. G. Abbott, G. Rehm, and I. S. Uzun, “A New Transverse and Longitudinal Bunch by Bunch Feedback Processor”, in *Proc. 16th Int. Conf. on Accelerator and Large Experimental Control Systems (ICALEPCS'17)*, Barcelona, Spain, Oct. 2017, pp. 1649–1654. doi:10.18429/JACoW-ICALEPCS2017-THPHA115
- [2] K. Levenberg, “A Method for the Solution of Certain Non-Linear Problems in Least Squares”, in *Quart. Appl. Math.*, vol. 2, pp. 164–168, 1944. doi:10.1090/qam/10666
- [3] D.W. Marquardt, “An algorithm for least-squares estimation of nonlinear parameters”, in *J. Soc. Indust. Appl. Math.*, vol. 11, no. 2, pp. 431–441, 1963. doi:10.1137/0111030
- [4] W.H. Press, B.P. Flannery, S.A. Teukolsky, W.T. Vetterling, *Numerical Recipes in C: The Art of Scientific Computing*. New York, NY, USA:Cambridge University Press, 1988, ch. 14, sec. 4, pp. 540–547.