# Tutorial: The Likelihood Interpretation of the Kalman Filter.

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<sup>&</sup>lt;sup>1</sup>This is only an extract from this work and contains references to material from other chapters in the larger document.

#### 1 Introduction

The Kalman filter [1] has long been regarded as the optimal solution to many tracking and data prediction tasks, [2]. Its use in the analysis of visual motion has been documented frequently. The standard Kalman filter derivation is given here as a tutorial exercise in the practical use of some of the statistical techniques outlied in previous sections. The filter is constructed as a mean squared error minimiser, but an alternative derivation of the filter is also provided showing how the filter relates to maximum likelihood statistics. Only by making this association can we understand the fundamental assumtions we are making regarding the data when we apply the filter to analysis tasks.

The purpose of filtering is to extract the required information from a signal, ignoring everything else. How well a filter performs this task can be measured using a cost or loss function. Indeed we may define the goal of the filter to be the minimisation of this loss function.

## 2 Mean squared error

Many signals can be described in the following way;

$$y_k = a_k x_k + n_k \tag{1}$$

where;  $y_k$  is the time dependent observed signal,  $a_k$  is a gain term,  $x_k$  is the information bearing signal and  $n_k$  is the additive noise.

The overall objective is to estimate  $x_k$ . The difference between the estimate of  $\hat{x}_k$  and  $x_k$  itself is termed the error;

$$f(e_k) = f(x_k - \hat{x}_k) \tag{2}$$

The particular shape of  $f(e_k)$  is dependent upon the application, however it is clear that the function should be both positive and increase monotonically [3]. An error function which exhibits these characteristics is the squared error function;

$$f(e_k) = (x_k - \hat{x}_k)^2 \tag{3}$$

Since it is necessary to consider the ability of the filter to predict many data over a period of time a more meaningful metric is the expected value of the error function;

$$loss function = E(f(e_k)) \tag{4}$$

This results in the mean squared error (MSE) function;

$$\epsilon(t) = E(e_k^2) \tag{5}$$

#### 3 Maximum likelihood

The above derivation of mean squared error, although intuitive is somewhat heuristic. A more rigorous derivation can be developed using  $maximum\ likelihood\ statistics$ . This is achieved by redefining the goal of the filter to finding the  $\hat{x}$  which maximises the probability or likelihood of y. That is;

$$max\left[P(y|\hat{x})\right] \tag{6}$$

Assuming that the additive random noise is Gaussian distributed with a standard deviation of  $\sigma_k$  gives;

$$P(y_k|\hat{x}_k) = K_k exp - \left(\frac{(y_k - a_k \hat{x}_k)^2}{2\sigma_k^2}\right)$$
 (7)

where  $K_k$  is a normalisation constant. The maximum likelihood function of this is;

$$P(y|\hat{x}) = \prod_{k} K_k exp - \left(\frac{(y_k - a_k \hat{x}_k)^2}{2\sigma_k^2}\right)$$
 (8)

Which leads to;

$$log P(y|\hat{x}) = -\frac{1}{2} \sum_{k} \left( \frac{(y_k - a_k \hat{x}_k)^2}{\sigma_k^2} \right) + constant$$
 (9)

The driving function of equation 11.9 is the MSE, which may be maximised by the variation of  $\hat{x}_k$ . Therefore the mean squared error function is applicable when the expected variation of  $y_k$  is best modelled as a Gaussian distribution. In such a case the MSE serves to provide the value of  $\hat{x}_k$  which maximises the likelihood of the signal  $y_k$ .

In the following derivation the *optimal* filter is defined as being that filter, from the set of all possible filters which minimises the mean squared error.

#### 4 Kalman Filter Derivation

Before going on to discuss the Kalman filter the work of Norbert Wiener [4], should first be acknowledged. Wiener described an optimal *finite impulse response* (FIR) filter in the mean squared error sense. His solution will not be discussed here even though it has much in common with the Kalman filter. Suffice to say that his solution uses both the auto correlation and the cross correlation of the received signal with the original data, in order to derive an impulse response for the filter.

Kalman also presented a prescription of the optimal MSE filter. However Kalman's prescription has some advantages over Weiner's; it sidesteps the need to determine the impulse response of the filter, something which is poorly suited to numerical computation. Kalman described his filter using state space techniques, which unlike Wiener's perscription, enables the filter to be used as either a smoother, a filter or a predictor. The latter of these three, the ability of the Kalman filter to be used to predict data has proven to be a very useful function. It has lead to the Kalman filter being applied to a wide range of tracking and navigation problems. Defining the filter in terms of state space methods also simplifies the implementation of the filter in the discrete domain, another reason for its widespread appeal. Such descriptions are important for explaining how to construct algorithms, but in order to understand when estimates using this technique are expected to be quantitatively valid we need to understand the filering process in terms of conventional statistics. This will also explain to us why it is that under the right circumstances the various matrices computed for the algorithm can be used as estimates of errors on computed parameters. This interpretation is crucial for any scientific application of these techniques.

# 5 State space derivation

Assume that we want to know the value of a variable within a process of the form;

$$x_{k+1} = \Phi x_k + w_k \tag{10}$$

where;  $x_k$  is the state vector of the process at time k, (nx1);  $\Phi$  is the state transition matrix of the process from the state at k to the state at k+1, and is assumed stationary over time, (nxm);  $w_k$  is the associated white noise process with known covariance, (nx1).

Observations on this variable can be modelled in the form;

$$z_k = Hx_k + v_k \tag{11}$$

where;  $z_k$  is the actual measurement of x at time k, (mx1); H is the noiseless connection between the state vector and the measurement vector, and is assumed stationary over time (mxn);  $v_k$  is the associated measurement error. This is again assumed to be a white noise process with known covariance and has zero cross-correlation with the process noise, (mx1).

As was shown in section ?? for the minimisation of the MSE to yield the optimal filter it must be possible to correctly model the system errors using Gaussian distributions. The covariances of the two noise models are assumed stationary over time and are given by;

$$Q = E\left[w_k w_k^T\right] \tag{12}$$

$$R = E\left[v_k v_k^T\right] \tag{13}$$

The mean squared error is given by 11.5. This is equivalent to;

$$E\left[e_{k}e_{k}^{T}\right] = P_{k} \tag{14}$$

where;  $P_k$  is the error covariance matrix at time k, (nxn).

Equation 11.14 may be expanded to give;

$$P_k = E\left[e_k e_k^T\right] = E\left[\left(x_k - \hat{x}_k\right)\left(x_k - \hat{x}_k\right)^T\right]$$
(15)

Assuming the prior estimate of  $\hat{x}_k$  is called  $\hat{x}_k'$ , and was gained by knowledge of the system. It possible to write an update equation for the new estimate, combing the old estimate with measurement data thus;

$$\hat{x}_k = \hat{x}'_k + K_k (z_k - H\hat{x}'_k) \tag{16}$$

where;  $K_k$  is the Kalman gain, which will be derived shortly. The term  $z_k - H\hat{x}_k'$  in eqn. 11.16 is known as the innovation or measurement residual;

$$i_k = z_k - H\hat{x}_k \tag{17}$$

Substitution of 11.11 into 11.16 gives;

$$\hat{x}_k = \hat{x}'_k + K_k (Hx_k + v_k - H\hat{x}'_k) \tag{18}$$

Substituting 11.18 into 11.15 gives;

$$P_{k} = E \left[ \left[ (I - K_{k}H) (x_{k} - \hat{x}'_{k}) - K_{k}v_{k} \right] \right]$$

$$\left[ (I - K_{k}H) (x_{k} - \hat{x}'_{k}) - K_{k}v_{k} \right]^{T}$$
(19)

At this point it is noted that  $x_k - \hat{x}'_k$  is the error of the prior estimate. It is clear that this is uncorrelated with the measurement noise and therefore the expectation may be re-written as;

$$P_{k} = (I - K_{k}H) E \left[ (x_{k} - \hat{x}'_{k}) (x_{k} - \hat{x}'_{k})^{T} \right] (I - K_{k}H)$$

$$+ K_{k}E \left[ v_{k}v_{k}^{T} \right] K_{k}^{T}$$
(20)

Substituting equations 11.13 and 11.15 into 11.19 gives;

$$P_{k} = (I - K_{k}H) P'_{k} (I - K_{k}H)^{T} + K_{k}RK_{k}^{T}$$
(21)

where  $P'_k$  is the prior estimate of  $P_k$ .

Equation 11.21 is the error covariance update equation. The diagonal of the covariance matrix contains the mean squared errors as shown;

$$P_{kk} = \begin{bmatrix} E \left[ e_{k-1} e_{k-1}^T \right] & E \left[ e_{k} e_{k-1}^T \right] & E \left[ e_{k+1} e_{k-1}^T \right] \\ E \left[ e_{k-1} e_{k}^T \right] & E \left[ e_{k} e_{k}^T \right] & E \left[ e_{k+1} e_{k}^T \right] \\ E \left[ e_{k-1} e_{k+1}^T \right] & E \left[ e_{k} e_{k+1}^T \right] & E \left[ e_{k+1} e_{k+1}^T \right] \end{bmatrix}$$

$$(22)$$

The sum of the diagonal elements of a matrix is the *trace* of a matrix. In the case of the error covariance matrix the trace is the sum of the mean squared errors. Therefore the mean squared error may be minimised by minimising the trace of  $P_k$  which in turn will minimise the trace of  $P_{kk}$ .

The trace of  $P_k$  is first differentiated with respect to  $K_k$  and the result set to zero in order to find the conditions of this minimum.

Expansion of 11.21 gives;

$$P_k = P_k' - K_k H P_k' - P_k' H^T K_k^T + K_k (H P_k' H^T + R) K_k^T$$
(23)

Note that the trace of a matrix is equal to the trace of its transpose, therefore it may written as;

$$T[P_k] = T[P'_k] - 2T[K_k H P'_k] + T[K_k (H P'_k H^T + R) K_k^T]$$
(24)

where;  $T[P_k]$  is the trace of the matrix  $P_k$ .

Differentiating with respect to  $K_k$  gives;

$$\frac{dT[P_k]}{dK_k} = -2(HP'_k)^T + 2K_k (HP'_kH^T + R)$$
 (25)

Setting to zero and re-arranging gives;

$$(HP_k')^T = K_k (HP_k'H^T + R) \tag{26}$$

Now solving for  $K_k$  gives;

$$K_k = P_k' H^T (H P_k' H^T + R)^{-1} (27)$$

Equation 11.27 is the Kalman gain equation. The innovation,  $i_k$  defined in eqn. 11.17 has an associated measurement prediction covariance. This is defined as;

$$S_k = HP_k'H^T + R (28)$$

Finally, substitution of equation 11.27 into 11.23 gives;

$$P_{k} = P'_{k} - P'_{k}H^{T} (HP'_{k}H^{T} + R)^{-1} HP'_{k}$$

$$= P'_{k} - K_{k}HP'_{k}$$

$$= (I - K_{k}H) P'_{k}$$
(29)

Equation 11.29 is the update equation for the error covariance matrix with optimal gain. The three equations 11.16, 11.27, and 11.29 develop an estimate of the variable  $x_k$ . State projection is achieved using;

$$\hat{x}'_{k+1} = \Phi \hat{x}_k \tag{30}$$

To complete the recursion it is necessary to find an equation which projects the error covariance matrix into the next time interval, k + 1. This is achieved by first forming an expressions for the prior error;

$$e'_{k+1} = x_{k+1} - \hat{x}'_{k+1} = (\Phi x_k + w_k) - \Phi \hat{x}_k = \Phi e_k + w_k$$
 (31)

Extending equation 11.15 to time k + 1;

$$P'_{k+1} = E\left[e'_{k+1}e^{T'}_{k+1}\right] = E\left[\left(\Phi e_k + w_k\right)\left(\Phi e_k + w_k\right)^T\right]$$
(32)

Note that  $e_k$  and  $w_k$  have zero cross-correlation because the noise  $w_k$  actually accumulates between k and k+1 whereas the error  $e_k$  is the error up until time k. Therefore;

$$P'_{k+1} = E\left[e'_{k+1}e^{T'}_{k+1}\right]$$

$$= E\left[\Phi e_k \left(\Phi e_k\right)^T\right] + E\left[w_k w_k^T\right]$$

$$= \Phi P_k \Phi^T + Q$$
(33)

This completes the recursive filter. The algorithmic loop is summarised in the diagram of figure 11.5.

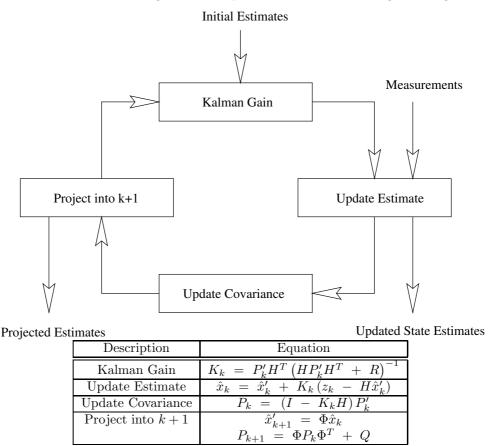


Figure 1: Kalman Filter Recursive Algorithm

## 6 The Kalman filter as a chi-square merit function

The objective of the Kalman filter is to minimise the mean squared error between the actual and estimated data. Thus it provides the best estimate of the data in the mean squared error sense. This being the case it should be possible to show that the Kalman filter has much in common with the *chi-square*. The chi-square merit function is a maximum likelihood function, and was derived earlier, equation 11.9. It is typically used as a criteria to fit a set of model parameters to a model a process known as *least squares* fitting. The Kalman filter is commonly known as a recursive least squares (RLS) fitter. Drawing similarities to the chi-square merit function will give a different perspective on what the Kalman filter is doing.

The chi-square merit function is the negative log likelihood for a set of data with Gaussian measurement errors;

$$\chi^2 = \sum_{i=1}^k \left[ \frac{z_i - h(a_i, x)}{\sigma_i} \right]^2 \tag{34}$$

where;  $z_i$  is the measured value;  $h_i$  is the data model with parameters x, assumed linear in a;  $\sigma_i$  is the variance associated with the measured value.

The optimal set of parameters can then be defined as that which minimises the above function. Expanding out the variance gives;

$$\chi^{2} = \sum_{i=1}^{K} \frac{1}{\sigma_{i} \sigma_{i}} \left[ z_{i} - h(a_{i}, x) \right]^{2}$$
(35)

Representing the chi-square in vector form and using notation from the earlier Kalman derivation;

$$\chi_k^2 = [z_k - h(a, x_k)] R^{-1} [z_k - h(a, x_k)]^T$$
(36)

where;  $R^{-1}$  is the matrix of inverse squared variances, i.e.  $1/\sigma_i\sigma_i$ .

The above merit function is the merit function associated with the latest, kth, measurement and provides a measure of how accurately the model predicted this measurement. In order to find new estimates of the parameters which takes this measurement and all previous data in to account we must find a way of writing down a combined funtion for all data. The inverse model covariance matrix for the parameters  $x_k$  at time k can be used to estimate the expected change in the log likelihood around the current minimum value  $\chi^2_{min}$  for any data up to K as;

$$\chi_{k-1}^{2} = \sum_{k}^{K-1} \left[ z_{k} - h(a, x_{k}) \right] R^{-1} \left[ z_{k} - h(a, x_{k}) \right]^{T} \approx \chi_{min}^{2} + \left( x_{k-1} - \hat{x}_{k-1} \right) P_{k-1}^{\prime - 1} \left( x_{k-1} - \hat{x}_{k-1} \right)^{T}$$
(37)

As the covariance matrix  $P_{k-1}$  is defined as the second derivatives of the log likelihood the approximation here is a Taylor expansion and exact for Gaussian errors on the data and a quadratic log likelihood. To combine the new data with the previous, fitting the model parameters so as to minimise the overall chi-square function, the merit function becomes the summation of the two;

$$\chi^{2} = \chi_{min}^{2} + (x_{k-1} - \hat{x}_{k-1}) P_{k-1}^{\prime - 1} (x_{k-1} - \hat{x}_{k-1})^{T} + [z_{k} - h(a, x_{k})] R^{-1} [z_{k} - h(a, x_{k})]^{T}$$
(38)

We must now determine the minimum of this function with respect to the model paramters x in order to find the new optimal estimate. The first derivative is given by;

$$\frac{d\chi^2}{dx} = 2P_{k-1}^{\prime - 1} (x_{k-1} - \hat{x}_{k-1}) - 2\nabla_x h(a, x_k)^T R^{-1} [z_k - h(a, x_k)]$$
(39)

The model function  $h(a, x_k)$  with parameters fitted from information to date, may be considered as;

$$h(a, x_k) = h\left(a, (\hat{x}_k + \Delta x_k)\right) \tag{40}$$

where  $\Delta x_k = x_k - \hat{x}_k$ . The Taylor series expansion of the model function to first order is;

$$h(\hat{x}_k + \Delta x) = h(\hat{x}_k) + \Delta x \nabla_x h(\hat{x}_k) \tag{41}$$

Substituting this result into the derivative equation 11.39 gives;

$$\frac{d\chi^{2}}{dx} = 2P_{k}^{\prime-1}(x_{k} - \hat{x}_{k}) 
- 2\nabla_{x}h(a, \hat{x}_{k})^{T}R^{-1}[z_{k} - h(a, \hat{x}_{k}) - (x_{k} - \hat{x}_{k})\nabla_{x}h(a, \hat{x}_{k})]$$
(42)

It is assumed that the estimated model parameters are a close approximation to the actual model parameters. Therefore it may be assumed that the derivatives of the actual model and the estimated model are the same. Further, for a system which is linear in a the model derivative is constant and may be written as;

$$\nabla_x h(a, x_k) = \nabla_x h(a, \hat{x}_k) = H \tag{43}$$

Substituting this into equation 11.39 gives;

$$\frac{d\chi^2}{dx} = 2P_k^{\prime - 1} \Delta x_k + 2H^T R^{-1} H \Delta x_k - 2H^T R^{-1} [z_k - h(a, \hat{x}_k)]$$
(44)

Re-arranging gives;

$$\frac{d\chi^2}{dx} = 2\left[P_k^{\prime - 1} + H^T R^{-1} H\right] \Delta x_k - 2H^T R^{-1} \left[z_k - h\left(a, \hat{x}_k\right)\right] \tag{45}$$

For a minimum the derivative is zero, rearrange in terms of  $\Delta x_k$  gives;

$$\Delta x_k = \left[ P_k^{\prime - 1} + H^T R^{-1} H \right]^{-1} H^T R^{-1} \left[ z_k - h(a, \hat{x}_k) \right]$$
(46)

$$x = \hat{x}_k + \left[ P_k^{\prime - 1} + H^T R^{-1} H \right]^{-1} H^T R^{-1} \left[ z_k - h(a, \hat{x}) \right]$$
(47)

Comparison of equation 11.47 to 11.16 allows the gain,  $K_k$  to be identified as;

$$K_k = \left[ P_k^{\prime - 1} + H^T R^{-1} H \right]^{-1} H^T R^{-1} \tag{48}$$

Giving a parameter update equation of the form;

$$x_k = \hat{x_k} + K_k [z_k - h(a, \hat{x}_k)] \tag{49}$$

Equation 11.49 is identical in form to 11.16 and describes the improvement of the parameter estimate using the error between measured and model projected values. The following sections now prove that the various terms are mathematically equivalent.

## 7 Model covariance update

The model parameter covariance has been considered in its inverted form where it is known as the *information*  $matrix^2$ . It is possible to formulate an alternative update equation for the covariance matrix using standard error propegation for the constraint of the new data z on the estimated parameters x;

$$P^{-1} = \nabla_x z^T R \nabla_x z = H^T R^{-1} H$$

where  $\nabla_x z$  is the vetor of derivatives. This is consistent with the idea of "Fisher Information" and is generally reffered to as the "Cramer-Rao Bound", and tells us how much the data is expected to constrain the parameters. To get the total information including all other measurements so far we need only add the inverse covariances.

$$P_{\nu}^{-1} = P_{\nu}^{\prime - 1} + H^{T} R^{-1} H (50)$$

This result is clearly different to standard covariance update used in the Kalman filter. However, it is possible to show that the covariance updates of equation 11.50 and equation 11.29 are equivalent. The problem is related to the "matrix inversion lemma", and although it is not possible to derive 11.50 from 11.29 we can prove the relationship by showing  $P_k \times P_k^{-1} = I$ .

Starting with

$$[(I - K_k H) P_k'] \times [P_k'^{-1} + H^T R^{-1} H]$$
(51)

Substituting for  $K_k$  gives;

$$\begin{split} & \left[ P_k' \ - \ P_k' H^T \left( H P_k' H^T \ + \ R \right)^{-1} H P_k' \right] \left[ P_k'^{-1} \ + \ H^T R^{-1} H \right] \\ = & I \ - \ P_k' H^T \left[ \left( H P_k' H^T \ + \ R \right)^{-1} \\ - & R^{-1} \ + \ \left( H P_k' H^T \ + \ R \right)^{-1} H P_k' H^T R^{-1} \right] H \end{split}$$

<sup>&</sup>lt;sup>2</sup>when the Kalman filter is built around the information matrix it is known as the *information filter* 

$$= I - P'_{k}H^{T} \left[ \left( HP'_{k}H^{T} + R \right)^{-1} \left( I + HP'_{k}H^{T}R^{-1} \right) - R^{-1} \right] H$$

$$= I - P'_{k}H^{T} \left[ R^{-1} - R^{-1} \right] H$$

$$= I$$
(52)

Thus the two update schemes for covariances are mathematically equivalent. Any differences in practical use will therefore be numerical, for example the standard update scheme requires intialisation of the covariance with large (arbitrary) values in order to avoid numerical problems, while the information form of the update process can be correctly iniatialised with the inverse covariance set to zero.

## 8 Alternative Kalman equations

Having shown that the covariance matrix can be updated via the previous equation it is possible to formulate an alternative Kalman gain from the standard gain equations 11.27. Inserting the identity matrix  $P_k \times P_k^{-1}$ ;

$$K_k = P_k P_k^{-1} P_k' H^T (H P_k' H^T + R)^{-1}$$

Replacing  $P_k^{-1}$  with equation 11.50 and rearranging;

$$= P_{k} (P_{k}^{\prime -1} + H^{T}R^{-1}H) P_{k}^{\prime}H^{T} (HP_{k}^{\prime}H^{T} + R)^{-1}$$

$$= P_{k} (I + H^{T}R^{-1}HP_{k}^{\prime}) H^{T} (HP_{k}^{\prime}H^{T} + R)^{-1}$$

$$= P_{k} (H^{T} + H^{T}R^{-1}HP_{k}^{\prime}H^{T}) (HP_{k}^{\prime}H^{T} + R)^{-1}$$

$$= P_{k}H^{T}R^{-1} (R + HP_{k}^{\prime}H^{T}) (HP_{k}^{\prime}H^{T} + R)^{-1}$$

$$= P_{k}H^{T}R^{-1}$$

$$= P_{k}H^{T}R^{-1}$$
(53)

Finally, replacing  $P_k$  with the inverse of equation 11.50;

$$K_k = \left[ H^T R^{-1} H + P_k^{\prime - 1} \right]^{-1} H^T R^{-1}$$
 (54)

Which is the same as the gain calculated from the chi-square equations, confirming that the gains are indeed equivalent.

#### 9 Conclusions

This tutorial has shown how the Kalman filter may be derived from the desire to minimise the mean squared error of a signal prediction. Several points in the derivation have been emphasised;

- The minimisation of the mean squared error is shown to be applicable when the expected errors on the signal are distribution as a Gaussian. Under such conditions the minimisation of the mean squared error between the data and the data prediction leads to the development of a maximum likelihood statistic
- It has been shown how the Kalman filter can be thought of in terms of a chi-squared minimiser by deriving an alternative form of the Kalman filter which highlights its statistical constructs including the processes of error propagation and data combination. This derivation leads to a common, alternative set of filter equations.

Although an alternative recursive algorithm has been developed the objective was to demonstrate the relationship between the Kalman filter and the chi-square statistic, showing how the Kalman filter embodies this statistic. The diagram of figure 11.2 shows how the alternative set of filter equations may be used to implement a Kalman filter. This form of the filter may be attractive due to the simplified gain calculation and some authors have been able to use this form of the filter in a distributed implementation [?]. The selection of which of these is most suitable for a particular task will depend upon the quantity of data measured (size of z) and the complexity of the state vector (size of x). For combination of single measurements the standard filter is simplest, thus the preferred implementation here is that given in figure 11.5. For large numbers of measurements, which will require a correspondingly large matrix inverse, the alternative approach 11.2 may be preffered.

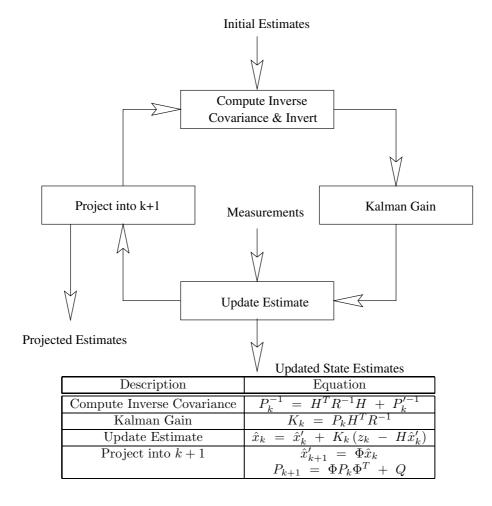


Figure 2: Alternative Kalman filter recursive algorithm

Put simply, the Kalman filter is fundamentally a recursive implementation of least squares fitting to a linear model. We can therefore not expect the method to work well for problems which we already know can not be tackled in this way, such as data contaminated with outliers. However, it does offer the possibility to test each new data point for consistency with the model parameters determined for the data included up to that time. Only by rejecting points which are statistically inconsistent with the model can we achieve something different to least squares fitting.

In addition, the statistical methods for data combination identified in this document, and in particular 11.38 can be used as the general approach to the combination of data when updating initial estimates of parameters with new measurements. On modern computers, optimisation can easily be performed iteratively and can even be modified to use a robust likelihood formulation, thereby eliminating the majority of problems caused by outliers. Often this is the simplest and most reliable approach and can even be done when the model z = f(x) is non-linear. This approach avoids many of the approximations inherrent in the "Extended Kalman Filter", which is the standard extension of the Kalman filter to non-linear models, and is therefore expected to be more reliable. It also generalises the use of likelihood to non-Gaussian processes in a rather straight forward manner, a problem recently tackled by the more complex technique referred to as "particle filtering". Such a powerful approach to the use of likelihood is sufficiently basic that it should be taught on introductory courses, as it is unlikely to be the subject of a paper in any major journal in the near future.

In summary, although the Kalman filter is optimal in the mean-squared error sense, it is limited, practically by the quality and accuracy of the model which is embedded within it and the data given to it. In particular, without an appropriate model the filter will be unable to perform the task for which it is designed. This is a particular problem when analysing data coming from an unknown process, such as feature tracking and motion analysis in computer vision. The following sections describe a new statistical approach to the solution of model selection.

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