

Smart Grid Voltage Regulations

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Abstract

The large-scale deployment of the Smart Grid paradigm will support the evolution of conventional electrical power systems toward active, flexible and self-healing web energy networks composed of distributed and cooperative energy resources. In a Smart Grid platform, the optimal coordination of distributed voltage controllers is one of the main issues to address. In this field, the application of traditional hierarchical control paradigms has some disadvantages that could hinder their application in Smart Grids where the constant growth of grid complexity and the need for massive pervasion of Distribution Generation Systems (DGS) require more scalable, more flexible control and regulation paradigms. To try and overcome these challenges, this paper proposes the concept of a decentralized non-hierarchical voltage regulation architecture based on intelligent and cooperative smart entities. The distributed voltage controllers employ traditional sensors to acquire local bus variables and mutually coupled oscillators to assess the main variables that characterize the operation of the global Smart Grid. These variables are then amalgamated by a Fuzzy Inference System in order to identify proper control actions aimed at improving the grid voltage profile and reducing power losses.

Keywords- Immune pathology; artificial immune system; negative selection algorithm; immunodeficiency; system Efficiency

I. INTRODUCTION

In order to respond to energy and climate challenges, an instrumented, intelligent smart electrical power grid has been proposed, and its related research and construction are in progress all over the world [1–14]. IBM has been contributing to the smart grid since 2004 when it proposed an Intelligent Utility Network [10, 14]. In China, Tsinghua University proposed the concept of a digital power system [12] in 1999 and initiated research work on a future electricity grid. In 2005, State Grid Corporation of China, the largest utility company in the world, started to study the framework for a digital electricity grid and digital substations and began to construct demonstration projects. In the same year, Tsinghua University was entrusted by China Southern Power Grid Corporation to carry out the research for a digital

China Southern Power Grid [13]. In 2009, State Grid Corporation of China proposed its aggressive plan to build a Strong Smart Grid. Undoubtedly, the smart grid has become the direction for future electricity grids. However, what does smart grid exactly mean? What are the essential differences between a smart grid and a traditional electricity grid? The smart grid should be evaluated with systematic indices so that it can be conveniently handled. In the United States, researchers studying smart grids have decided this through a series of regional conferences that took place from 2005 to 2010, together with some regional demonstrations such as the Pacific Northwest GridWise** Project carried out by IBM Research. Through these efforts, researchers have not only defined the core values of the U.S. smart grid that they expect but also determined how to evaluate the core values with corresponding indices. For instance, grid reliability can be evaluated by the average outage time, the outage frequency, and the instant outage and power supply quality. Although the conditions of each country could be different, the vision of a smart grid can always be expressed by core values, and each value can be evaluated with a series of indices.

In this paper, we synthesize the indices that a smart grid needs for operation and propose the multi-index and self-approximate-optimal operation architecture for a smart grid. First, we define the multi-index and self-approximate-optimal operation architecture and explain it in detail based on the analysis and synthesis of the exploration and practices on smart grids in the United States and Europe. Second, we show that multi-index and self-approximate-optimal operation could be achieved with two realistic and promising practices in China. Finally, conclusions are drawn, and future research is proposed.

A. Multi-index self approximate-optimal operation

The implementation of smart grids may involve new technologies, such as renewable energy generation, energy storage, superconducting transmission, ultrahigh-voltage transmission, intelligent metering, advanced sensing, communication, and information technology. There are many indices to be optimized. However, there is only theoretical meaning in performing an exact global optimization, whereas in engineering practice, it may be impossible and unnecessary to achieve the exact global optimal operation point. Moreover, the

electricity grid needs to have the ability to return to an approximate-optimal status from events such as disturbances through automated closed-loop control and regulation. Based on this, we define a smart grid as an electricity grid that has the ability of multi-index self-approximate-optimal operation. In this section, we first present the overall architecture. Then, we explain multi-index and self-approximate-optimal. Finally, the goal of multi-index self-approximate-optimal operation is summarized as an object function, and the solving algorithm is discussed. Architecture, each circle represents one of the three categories of indices for smart grid operation: stability, reliability, and economy. More circles can be drawn if more indices are needed or a new category can be created for several detailed indices. In fact, there are thousands of indices for grid operation; thus, there could be thousands of such circles. The BMulti-index[section describes the most important indices. Outside a circle means the corresponding index is not acceptable, but inside means the index is satisfied and the optimal value must fall into one location there. The central overlap region means all indices are satisfied, and therefore, this is where a smart grid should operate. A global optimal operation point must be within this overlap region, but it is practically impossible to find this single point due to the problem dimension, that is, thousands, as indicated above. Moreover, it is unnecessary and impossible to adjust the system in the optimal state because the power system is a complex and large-scale dynamic system in which load and operating conditions are changing all the time. It will also influence the service life and economic efficiency of the equipment if the adjustment is too frequent. As a result, the system can be considered to be operating in a satisfactory state if its state lies in a sufficiently small neighborhood near the optimal state, that is, within the overlap region, which is hereafter defined as the multiobjective approximate-optimal operation region (MOOR). In the MOOR, all the indices should be in the acceptable region. Each index has its own range. The abnormal condition occurs when the index is out of its range. For example, for one node, its voltage magnitude should be kept between 0.9 and 1.1 p.u. Whenever the voltage magnitude becomes 0.78, an abnormal situation is identified. If one or more indices are not within the MOOR, the control system will act to eliminate the unsatisfying states so that the operation point of the system returns to the MOOR. As is further explained, self-approximate-optimal means the smart grid can respond to events, including disturbances that happened in the grid, and return to the MOOR. In order to make this architecture better understood, we can use a common concept in the smart grid, namely, self-healing [15–20]. As pointed out in [15], BSelf-healing strategies are control options that are initiated to steer the power system to a more secure, less vulnerable, operating condition.[

This provides us with the following information: First, self-healing focuses on the operating condition of the power system, which is the focus of this paper; second, the operating condition should be secure and less vulnerable, which corresponds to Bmulti-index;[third, Bmore[means Bapproximate-optimal,[that is, not to guarantee the global optimal; and fourth, Bcontrol options[and Bsteer[match the Bself[term used in this paper to represent automated

II. A DECENTRALIZED NON-HIERARCHAL VOLTAGE REGULATION ARCHITECTURE

To address the voltage regulation problem in a Smart Grid, an innovative solution approach based on a decentralized /non-hierarchical architecture is proposed here. This architecture is based on a network of cooperative smart controllers, each regulating the voltage magnitude of a specific Smart Grid section. Each controller is equipped with five basic components:

1. a set of sensors measuring the available set of local electrical variables (i.e. voltage magnitude, active and reactive bus power);
2. a dynamical system, whose state is initialized by sensor measurements and evolves interactively with the states of nearby controllers according to a bio-inspired paradigm;
3. a radio interface ensuring the interaction among controllers by transmitting the state of the dynamical system and receiving the state transmitted by the other nodes.
4. a Fuzzy Inference System regulating the reactive power flows injected by the DGSs into the electrical grid.

The bio-inspired paradigm adopted for coupling the voltage controllers is based on a challenging idea, originating in papers [19, 21], that borrows the mathematics of populations of mutually coupled oscillators, where the self synchronization of the network is ensured without the need for a fusion center, but only with proper local coupling of nodes. As discussed in section III.A, the adoption of this paradigm allows the voltage controllers to assess, in a totally decentralized way, many important variables characterizing the operation of the global Smart Grid.

Thanks to this feature, each controller knows both the variables characterizing the monitored bus (sensed by in-built sensors) and the global variables describing the actual performances of the entire Smart Grid (assessed by checking the state of the dynamical system). Both local and global variables, if properly processed, allow each controller to identify the proper control actions aimed at improving the grid voltage profile and reducing power losses. This process is realized by a Fuzzy Inference System regulating the reactive

power flows injected by the DGSs into the electrical grid.

A. Approach

To address the voltage regulation problem in a Smart Grid, an innovative solution approach is proposed here. The reactive power flows injected by the DGSs into the electrical grid are regulated according to the following control strategy: The proposed algorithm first acquires the global variables characterizing the actual operation of the Smart Grid. These variables are then processed in order to estimate the current grid state. This could be classified on the basis of the mean grid voltage magnitude according to the following set of minimal states:

- 1) Grid Voltage Low: This state is characterized by high load demand. As a result, high power line loading and, consequently, high power losses and a very low value of the mean grid voltage magnitude are expected on the grid.
- 2) Grid Voltage High: This state is characterized by low load demand. As a result, low power line loading and, consequently, low power losses and a very high value of the mean grid voltage magnitude are expected on the grid.
- 3) Normal Operation: This state represents the nominal operating point of the electrical grid. Both the power losses and the value of the mean grid voltage magnitude lie within allowable intervals. Once the actual Smart Grid state has been assessed, the algorithm identifies proper voltage control strategies for each node. This is implemented according to the following control rules:

If the Smart Grid state is classified as Grid Voltage Low (respectively Grid Voltage High), then the control objective is to raise (respectively reduce) the mean grid voltage magnitude V_m by properly increasing (respectively decreasing) the reactive power injected by the DGSs into the grid. In order to obtain a uniform rise in voltage magnitude, the i th controller compares the local bus voltage magnitude (V_i) with the mean grid voltage magnitude (V_m)

We consider the following anycast field equations defined over an open bounded piece of network and /or feature space $\Omega \subset R^d$. They describe the dynamics of the mean anycast of each of p node populations.

$$\begin{cases} \left(\frac{d}{dt} + l_i \right) V_i(t, r) = \sum_{j=1}^p \int_{\Omega} J_{ij}(r, \bar{r}) S[(V_j(t - \tau_{ij}(r, \bar{r})) - h_{ij})] d\bar{r} \\ \quad + I_i^{ext}(r, t), \quad t \geq 0, 1 \leq i \leq p, \\ V_i(t, r) = \phi_i(t, r) \quad t \in [-T, 0] \end{cases} \quad (1)$$

We give an interpretation of the various parameters and functions that appear in (1), Ω is finite piece of nodes and/or feature space and is represented as an open bounded set of R^d . The vector r and \bar{r} represent points in Ω . The function $S : R \rightarrow (0, 1)$ is the normalized sigmoid function:

$$S(z) = \frac{1}{1 + e^{-z}} \quad (2)$$

It describes the relation between the input rate v_i of population i as a function of the packets potential, for example, $V_i = v_i = S[\sigma_i(V_i - h_i)]$. We note V the p -dimensional vector (V_1, \dots, V_p) . The p function $\phi_i, i = 1, \dots, p$, represent the initial conditions, see below. We note ϕ the p -dimensional vector (ϕ_1, \dots, ϕ_p) . The p function $I_i^{ext}, i = 1, \dots, p$, represent external factors from other network areas. We note I^{ext} the p -dimensional vector $(I_1^{ext}, \dots, I_p^{ext})$. The $p \times p$ matrix of functions $J = \{J_{ij}\}_{i,j=1,\dots,p}$ represents the connectivity between populations i and j , see below. The p real values $h_i, i = 1, \dots, p$, determine the threshold of activity for each population, that is, the value of the nodes potential corresponding to 50% of the maximal activity. The p real positive values $\sigma_i, i = 1, \dots, p$, determine the slopes of the sigmoids at the origin. Finally the p real positive values $l_i, i = 1, \dots, p$, determine the speed at which each anycast node potential decreases exponentially toward its real value. We also introduce the function $S : R^p \rightarrow R^p$, defined by $S(x) = [S(\sigma_1(x_1 - h_1)), \dots, S(\sigma_p(x_p - h_p))]$, and the diagonal $p \times p$ matrix $L_0 = \text{diag}(l_1, \dots, l_p)$. Is the intrinsic dynamics of the population given by the linear response of data transfer. $(\frac{d}{dt} + l_i)$ is replaced by $(\frac{d}{dt} + l_i)^2$ to use

the alpha function response. We use $(\frac{d}{dt} + l_i)$ for simplicity although our analysis applies to more general intrinsic dynamics. For the sake, of generality, the propagation delays are not assumed to be identical for all populations, hence they are described by a matrix $\tau(r, \bar{r})$ whose element $\tau_{ij}(r, \bar{r})$ is the propagation delay between

population j at \bar{r} and population i at r . The reason for this assumption is that it is still unclear from anycast if propagation delays are independent of the populations. We assume for technical reasons that τ is continuous, that is $\tau \in C^0(\bar{\Omega}^2, R_+^{p \times p})$. Moreover packet data indicate that τ is not a symmetric function i.e., $\tau_{ij}(r, \bar{r}) \neq \tau_{ij}(\bar{r}, r)$, thus no assumption is made about this symmetry unless otherwise stated. In order to compute the righthand side of (1), we need to know the node potential factor V on interval $[-T, 0]$. The value of T is obtained by considering the maximal delay:

$$\tau_m = \max_{i,j(r,r \in \Omega \times \bar{\Omega})} \tau_{i,j}(r, \bar{r}) \quad (3)$$

Hence we choose $T = \tau_m$

B. Mathematical Framework

A convenient functional setting for the non-delayed packet field equations is to use the space $F = L^2(\Omega, R^p)$ which is a Hilbert space endowed with the usual inner product:

$$\langle V, U \rangle_F = \sum_{i=1}^p \int_{\Omega} V_i(r) U_i(r) dr \quad (1)$$

To give a meaning to (1), we defined the history space $C = C^0([-\tau_m, 0], F)$ with $\|\phi\| = \sup_{t \in [-\tau_m, 0]} \|\phi(t)\|_F$, which is the Banach phase space associated with equation (3). Using the notation $V_t(\theta) = V(t + \theta)$, $\theta \in [-\tau_m, 0]$, we write (1) as

$$\begin{cases} V(t) = -L_0 V(t) + L_1 S(V_t) + I^{ext}(t), \\ V_0 = \phi \in C, \end{cases} \quad (2)$$

Where

$$\begin{cases} L_1 : C \rightarrow F, \\ \phi \rightarrow \int_{\Omega} J(\cdot, \bar{r}) \phi(\bar{r}, -\tau(\cdot, \bar{r})) d\bar{r} \end{cases}$$

Is the linear continuous operator satisfying $\|L_1\| \leq \|J\|_{L^2(\Omega^2, R^{p \times p})}$. Notice that most of the papers on this subject assume Ω infinite, hence requiring $\tau_m = \infty$.

Proposition 1.0 If the following assumptions are satisfied.

1. $J \in L^2(\Omega^2, R^{p \times p})$,
2. The external current $I^{ext} \in C^0(R, F)$,

$$3. \quad \tau \in C^0(\bar{\Omega}^2, R_+^{p \times p}), \sup_{\bar{\Omega}^2} \tau \leq \tau_m.$$

Then for any $\phi \in C$, there exists a unique solution $V \in C^1([0, \infty), F) \cap C^0([-\tau_m, \infty), F)$ to (3)

Notice that this result gives existence on R_+ , finite-time explosion is impossible for this delayed differential equation. Nevertheless, a particular solution could grow indefinitely, we now prove that this cannot happen.

C. Boundedness of Solutions

A valid model of neural networks should only feature bounded packet node potentials.

Theorem 1.0 All the trajectories are ultimately bounded by the same constant R if $I \equiv \max_{t \in R^+} \|I^{ext}(t)\|_F < \infty$.

Proof :Let us defined $f : R \times C \rightarrow R^+$ as

$$f(t, V_t) = \left\langle -L_0 V_t(0) + L_1 S(V_t) + I^{ext}(t), V(t) \right\rangle_F \stackrel{def}{=} \frac{1}{2} \frac{d \|V\|_F^2}{dt}$$

We note $l = \min_{i=1, \dots, p} l_i$

$$f(t, V_t) \leq -l \|V(t)\|_F^2 + (\sqrt{p} \|\Omega\| \|J\|_F + I) \|V(t)\|_F$$

Thus, if

$$\|V(t)\|_F \geq 2 \frac{\sqrt{p} \|\Omega\| \|J\|_F + I}{l} \stackrel{def}{=} R, f(t, V_t) \leq -\frac{l R^2}{2} \stackrel{def}{=} -\delta < 0$$

Let us show that the open route of F of center 0 and radius R, B_R , is stable under the dynamics of equation. We know that $V(t)$ is defined for all $t \geq 0$ s and that $f < 0$ on ∂B_R , the boundary of B_R . We consider three cases for the initial condition V_0 . If $\|V_0\|_C < R$ and set $T = \sup\{t \mid \forall s \in [0, t], V(s) \in \bar{B}_R\}$. Suppose that $T \in R$, then $V(T)$ is defined and belongs to \bar{B}_R , the closure of B_R , because \bar{B}_R is closed, in effect to ∂B_R , we also have

$$\frac{d}{dt} \|V\|_F^2 \Big|_{t=T} = f(T, V_T) \leq -\delta < 0 \quad \text{because}$$

$V(T) \in \partial B_R$. Thus we deduce that for $\varepsilon > 0$ and small enough, $V(T + \varepsilon) \in \bar{B}_R$ which contradicts the definition of T. Thus $T \notin R$ and \bar{B}_R is stable.

Because $f < 0$ on $\partial B_R, V(0) \in \partial B_R$ implies that $\forall t > 0, V(t) \in B_R$. Finally we consider the case $V(0) \in \overline{CB_R}$. Suppose that $\forall t > 0, V(t) \notin \overline{B_R}$, then $\forall t > 0, \frac{d}{dt} \|V\|_F^2 \leq -2\delta$, thus $\|V(t)\|_F$ is monotonically decreasing and reaches the value of R in finite time when $V(t)$ reaches ∂B_R . This contradicts our assumption. Thus $\exists T > 0 | V(T) \in B_R$.

Proposition 1.1 : Let s and t be measured simple functions on X . for $E \in \mathcal{M}$, define

$$\phi(E) = \int_E s d\mu \quad (1)$$

Then ϕ is a measure on M .

$$\int_X (s+t) d\mu = \int_X s d\mu + \int_X t d\mu \quad (2)$$

Proof : If s and if E_1, E_2, \dots are disjoint members of M whose union is E , the countable additivity of μ shows that

$$\begin{aligned} \phi(E) &= \sum_{i=1}^n \alpha_i \mu(A_i \cap E) = \sum_{i=1}^n \alpha_i \sum_{r=1}^{\infty} \mu(A_i \cap E_r) \\ &= \sum_{r=1}^{\infty} \sum_{i=1}^n \alpha_i \mu(A_i \cap E_r) = \sum_{r=1}^{\infty} \phi(E_r) \end{aligned}$$

Also, $\phi(\phi) = 0$, so that ϕ is not identically ∞ .

Next, let s be as before, let β_1, \dots, β_m be the distinct values of t , and let $B_j = \{x : t(x) = \beta_j\}$ If

$$E_{ij} = A_i \cap B_j, \quad \text{the}$$

$$\int_{E_{ij}} (s+t) d\mu = (\alpha_i + \beta_j) \mu(E_{ij})$$

$$\text{and } \int_{E_{ij}} s d\mu + \int_{E_{ij}} t d\mu = \alpha_i \mu(E_{ij}) + \beta_j \mu(E_{ij})$$

Thus (2) holds with E_{ij} in place of X . Since X is the disjoint union of the sets E_{ij} ($1 \leq i \leq n, 1 \leq j \leq m$), the first half of our proposition implies that (2) holds.

Theorem 1.1: If K is a compact set in the plane whose complement is connected, if f is a continuous complex function on K which is holomorphic in the interior of K , and if $\varepsilon > 0$, then there exists a polynomial P such that

$|f(z) - P(z)| < \varepsilon$ for all $z \in K$. If the interior of K is empty, then part of the hypothesis is vacuously satisfied, and the conclusion holds for every $f \in C(K)$. Note that K need to be connected.

Proof: By Tietze's theorem, f can be extended to a continuous function in the plane, with compact support. We fix one such extension and denote it again by f . For any $\delta > 0$, let $\omega(\delta)$ be the supremum of the numbers $|f(z_2) - f(z_1)|$ Where

z_1 and z_2 are subject to the condition $|z_2 - z_1| \leq \delta$. Since f is uniformly continuous, we have $\lim_{\delta \rightarrow 0} \omega(\delta) = 0$ (1) From now on,

δ will be fixed. We shall prove that there is a polynomial P such that

$$|f(z) - P(z)| < 10,000 \omega(\delta) \quad (z \in K) \quad (2)$$

By (1), this proves the theorem. Our first objective is the construction of a function $\Phi \in C_c'(R^2)$, such that for all z

$$|f(z) - \Phi(z)| \leq \omega(\delta), \quad (3)$$

$$|(\partial\Phi)(z)| < \frac{2\omega(\delta)}{\delta}, \quad (4)$$

And

$$\Phi(z) = -\frac{1}{\pi} \iint_X \frac{(\partial\Phi)(\zeta)}{\zeta - z} d\zeta d\eta \quad (\zeta = \xi + i\eta), \quad (5)$$

Where X is the set of all points in the support of Φ whose distance from the complement of K does not δ . (Thus X contains no point which is "far within" K .) We construct Φ as the convolution of f with a smoothing function A . Put $a(r) = 0$ if $r > \delta$, put

$$a(r) = \frac{3}{\pi\delta^2} \left(1 - \frac{r^2}{\delta^2}\right)^2 \quad (0 \leq r \leq \delta), \quad (6)$$

And define

$$A(z) = a(|z|) \quad (7)$$

For all complex z . It is clear that $A \in C_c'(R^2)$. We claim that

$$\iint_{R^2} A = 1, \quad (8)$$

$$\iint_{R^2} \partial A = 0, \quad (9)$$

$$\iint_{R^2} |\partial A| = \frac{24}{15\delta} < \frac{2}{\delta}, \quad (10)$$

The constants are so adjusted in (6) that (8) holds. (Compute the integral in polar coordinates), (9) holds simply because A has compact support. To compute (10), express ∂A in polar coordinates, and note that $\frac{\partial A}{\partial \theta} = 0$,

$$\frac{\partial A}{\partial r} = -a'$$

Now define

$$\Phi(z) = \iint_{R^2} f(z-\zeta) A d\xi d\eta = \iint_{R^2} A(z-\zeta) f(\zeta) d\xi d\eta \quad (11)$$

Since f and A have compact support, so does Φ . Since

$$\begin{aligned} & \Phi(z) - f(z) \\ &= \iint_{R^2} [f(z-\zeta) - f(z)] A(\xi) d\xi d\eta \quad (12) \end{aligned}$$

And $A(\zeta) = 0$ if $|\zeta| > \delta$, (3) follows from (8).

The difference quotients of A converge boundedly to the corresponding partial derivatives, since $A \in C_c^1(R^2)$. Hence the last expression in (11) may be differentiated under the integral sign, and we obtain

$$\begin{aligned} (\partial\Phi)(z) &= \iint_{R^2} (\partial A)(z-\zeta) f(\zeta) d\xi d\eta \\ &= \iint_{R^2} f(z-\zeta) (\partial A)(\zeta) d\xi d\eta \\ &= \iint_{R^2} [f(z-\zeta) - f(z)] (\partial A)(\zeta) d\xi d\eta \quad (13) \end{aligned}$$

The last equality depends on (9). Now (10) and (13) give (4). If we write (13) with Φ_x and Φ_y in place of $\partial\Phi$, we see that Φ has continuous partial derivatives, if we can show that $\partial\Phi = 0$ in G , where G is the set of all $z \in K$ whose distance from the complement of K exceeds δ . We shall do this by showing that

$$\Phi(z) = f(z) \quad (z \in G); \quad (14)$$

Note that $\partial f = 0$ in G , since f is holomorphic there. Now if $z \in G$, then $z - \zeta$ is in the interior of

K for all ζ with $|\zeta| < \delta$. The mean value property for harmonic functions therefore gives, by the first equation in (11),

$$\begin{aligned} \Phi(z) &= \int_0^\delta a(r) r dr \int_0^{2\pi} f(z - r e^{i\theta}) d\theta \\ &= 2\pi f(z) \int_0^\delta a(r) r dr = f(z) \iint_{R^2} A = f(z) \quad (15) \end{aligned}$$

For all $z \in G$, we have now proved (3), (4), and (5). The definition of X shows that X is compact and that X can be covered by finitely many open discs D_1, \dots, D_n , of radius 2δ , whose centers are not in K . Since $S^2 - K$ is connected, the center of each D_j can be joined to ∞ by a polygonal path in $S^2 - K$. It follows that each D_j contains a compact connected set E_j , of diameter at least 2δ , so that $S^2 - E_j$ is connected and so that $K \cap E_j = \emptyset$ with $r = 2\delta$. There are functions $g_j \in H(S^2 - E_j)$ and constants b_j so that the inequalities.

$$|Q_j(\zeta, z)| < \frac{50}{\delta}, \quad (16)$$

$$\left| Q_j(\zeta, z) - \frac{1}{z - \zeta} \right| < \frac{4,000\delta^2}{|z - \zeta|^2} \quad (17)$$

Hold for $z \notin E_j$ and $\zeta \in D_j$, if

$$Q_j(\zeta, z) = g_j(z) + (\zeta - b_j) g_j^2(z) \quad (18)$$

Let Ω be the complement of $E_1 \cup \dots \cup E_n$. Then

Ω is an open set which contains K . Put $X_1 = X \cap D_1$ and

$X_j = (X \cap D_j) - (X_1 \cup \dots \cup X_{j-1})$, for $2 \leq j \leq n$,

Define

$$R(\zeta, z) = Q_j(\zeta, z) \quad (\zeta \in X_j, z \in \Omega) \quad (19)$$

And

$$F(z) = \frac{1}{\pi} \iint_X (\partial\Phi)(\zeta) R(\zeta, z) d\xi d\eta \quad (20)$$

$(z \in \Omega)$

Since,

$$F(z) = \sum_{j=1}^n \frac{1}{\pi} \iint_{X_j} (\partial\Phi)(\zeta) Q_j(\zeta, z) d\xi d\eta, \quad (21)$$

(18) shows that F is a finite linear combination of the functions g_j and g_j^2 . Hence $F \in H(\Omega)$. By (20), (4), and (5) we have

$$|F(z) - \Phi(z)| < \frac{2\omega(\delta)}{\pi\delta} \iint_X |R(\zeta, z)| - \frac{1}{z - \zeta} |d\xi d\eta| \quad (z \in \Omega) \quad (22)$$

Observe that the inequalities (16) and (17) are valid with R in place of Q_j if $\zeta \in X$ and $z \in \Omega$.

Now fix $z \in \Omega$, put $\zeta = z + \rho e^{i\theta}$, and estimate the integrand in (22) by (16) if $\rho < 4\delta$, by (17) if $4\delta \leq \rho$. The integral in (22) is then seen to be less than the sum of

$$2\pi \int_0^{4\delta} \left(\frac{50}{\delta} + \frac{1}{\rho} \right) \rho d\rho = 808\pi\delta \quad (23)$$

And

$$2\pi \int_{4\delta}^{\infty} \frac{4,000\delta^2}{\rho^2} \rho d\rho = 2,000\pi\delta. \quad (24)$$

Hence (22) yields

$$|F(z) - \Phi(z)| < 6,000\omega(\delta) \quad (z \in \Omega) \quad (25)$$

Since $F \in H(\Omega)$, $K \subset \Omega$, and $S^2 - K$ is connected, Runge's theorem shows that F can be uniformly approximated on K by polynomials. Hence (3) and (25) show that (2) can be satisfied. This completes the proof.

Lemma 1.0 : Suppose $f \in C_c^1(\mathbb{R}^2)$, the space of all continuously differentiable functions in the plane, with compact support. Put

$$\partial = \frac{1}{2} \left(\frac{\partial}{\partial x} + i \frac{\partial}{\partial y} \right) \quad (1)$$

Then the following "Cauchy formula" holds:

$$f(z) = -\frac{1}{\pi} \iint_{\mathbb{R}^2} \frac{(\partial f)(\zeta)}{\zeta - z} d\xi d\eta \quad (\zeta = \xi + i\eta) \quad (2)$$

Proof: This may be deduced from Green's theorem. However, here is a simple direct proof:

Put $\varphi(r, \theta) = f(z + re^{i\theta})$, $r > 0$, θ real

If $\zeta = z + re^{i\theta}$, the chain rule gives

$$(\partial f)(\zeta) = \frac{1}{2} e^{i\theta} \left[\frac{\partial}{\partial r} + \frac{i}{r} \frac{\partial}{\partial \theta} \right] \varphi(r, \theta) \quad (3)$$

The right side of (2) is therefore equal to the limit, as $\varepsilon \rightarrow 0$, of

$$-\frac{1}{2} \int_{\varepsilon}^{\infty} \int_0^{2\pi} \left(\frac{\partial \varphi}{\partial r} + \frac{i}{r} \frac{\partial \varphi}{\partial \theta} \right) d\theta dr \quad (4)$$

For each $r > 0$, φ is periodic in θ , with period 2π . The integral of $\partial \varphi / \partial \theta$ is therefore 0, and (4) becomes

$$-\frac{1}{2\pi} \int_0^{2\pi} d\theta \int_{\varepsilon}^{\infty} \frac{\partial \varphi}{\partial r} dr = \frac{1}{2\pi} \int_0^{2\pi} \varphi(\varepsilon, \theta) d\theta \quad (5)$$

As $\varepsilon \rightarrow 0$, $\varphi(\varepsilon, \theta) \rightarrow f(z)$ uniformly. This gives (2)

If $X^\alpha \in a$ and $X^\beta \in k[X_1, \dots, X_n]$, then $X^\alpha X^\beta = X^{\alpha+\beta} \in a$, and so A satisfies the condition (*). Conversely,

$$\left(\sum_{\alpha \in A} c_\alpha X^\alpha \right) \left(\sum_{\beta \in \square^n} d_\beta X^\beta \right) = \sum_{\alpha, \beta} c_\alpha d_\beta X^{\alpha+\beta} \quad (\text{finite sums}),$$

and so if A satisfies (*), then the subspace generated by the monomials $X^\alpha, \alpha \in a$, is an ideal. The proposition gives a classification of the monomial ideals in $k[X_1, \dots, X_n]$: they are in one to one correspondence with the subsets A of \square^n satisfying (*). For example, the monomial ideals in $k[X]$ are exactly the ideals $(X^n), n \geq 1$, and the zero ideal (corresponding to the empty set A). We write $\langle X^\alpha \mid \alpha \in A \rangle$ for the ideal corresponding to A (subspace generated by the $X^\alpha, \alpha \in a$).

LEMMA 1.1. Let S be a subset of \square^n . The ideal a generated by $X^\alpha, \alpha \in S$ is the monomial ideal corresponding to

$$A \stackrel{df}{=} \{ \beta \in \square^n \mid \beta - \alpha \in \square^n, \text{ some } \alpha \in S \}$$

Thus, a monomial is in a if and only if it is divisible by one of the $X^\alpha, \alpha \in S$

PROOF. Clearly A satisfies (*), and $a \subset \langle X^\beta \mid \beta \in A \rangle$. Conversely, if $\beta \in A$, then $\beta - \alpha \in \square^n$ for some $\alpha \in S$, and $X^\beta = X^\alpha X^{\beta-\alpha} \in a$. The last statement follows from the fact that $X^\alpha \mid X^\beta \Leftrightarrow \beta - \alpha \in \square^n$. Let $A \subset \square^n$ satisfy (*). From the geometry of A , it

is clear that there is a finite set of elements $S = \{\alpha_1, \dots, \alpha_s\}$ of A such that

$$A = \left\{ \beta \in \square^n \mid \beta - \alpha_i \in \square^2, \text{ some } \alpha_i \in S \right\}$$

(The α_i 's are the corners of A) Moreover,

$$a \stackrel{df}{=} \left\langle X^\alpha \mid \alpha \in A \right\rangle \text{ is generated by the monomials } X^{\alpha_i}, \alpha_i \in S.$$

DEFINITION 1.0. For a nonzero ideal a in $k[X_1, \dots, X_n]$, we let $(LT(a))$ be the ideal generated by

$$\{LT(f) \mid f \in a\}$$

LEMMA 1.2 Let a be a nonzero ideal in $k[X_1, \dots, X_n]$; then $(LT(a))$ is a monomial ideal, and it equals $(LT(g_1), \dots, LT(g_n))$ for some $g_1, \dots, g_n \in a$.

PROOF. Since $(LT(a))$ can also be described as the ideal generated by the leading monomials (rather than the leading terms) of elements of a .

THEOREM 1.2. Every ideal a in $k[X_1, \dots, X_n]$ is finitely generated; more precisely, $a = (g_1, \dots, g_s)$ where g_1, \dots, g_s are any elements of a whose leading terms generate $LT(a)$

PROOF. Let $f \in a$. On applying the division algorithm, we find $f = a_1 g_1 + \dots + a_s g_s + r$, $a_i, r \in k[X_1, \dots, X_n]$, where either $r = 0$ or no monomial occurring in it is divisible by any $LT(g_i)$. But $r = f - \sum a_i g_i \in a$, and therefore $LT(r) \in LT(a) = (LT(g_1), \dots, LT(g_s))$, implies that every monomial occurring in r is divisible by one in $LT(g_i)$. Thus $r = 0$, and $g \in (g_1, \dots, g_s)$.

DEFINITION 1.1. A finite subset $S = \{g_1, \dots, g_s\}$ of an ideal a is a standard (Gröbner) bases for a if $(LT(g_1), \dots, LT(g_s)) = LT(a)$. In other words, S is a standard basis if the leading term of every

element of a is divisible by at least one of the leading terms of the g_i .

THEOREM 1.3 The ring $k[X_1, \dots, X_n]$ is Noetherian i.e., every ideal is finitely generated.

PROOF. For $n = 1$, $k[X]$ is a principal ideal domain, which means that every ideal is generated by single element. We shall prove the theorem by induction on n . Note that the obvious map $k[X_1, \dots, X_{n-1}][X_n] \rightarrow k[X_1, \dots, X_n]$ is an isomorphism – this simply says that every polynomial f in n variables X_1, \dots, X_n can be expressed uniquely as a polynomial in X_n with coefficients in $k[X_1, \dots, X_{n-1}]$:

$$f(X_1, \dots, X_n) = a_0(X_1, \dots, X_{n-1})X_n^r + \dots + a_r(X_1, \dots, X_{n-1})$$

Thus the next lemma will complete the proof

LEMMA 1.3. If A is Noetherian, then so also is $A[X]$

PROOF. For a polynomial

$$f(X) = a_0 X^r + a_1 X^{r-1} + \dots + a_r, \quad a_i \in A, \quad a_0 \neq 0,$$

r is called the degree of f , and a_0 is its leading coefficient. We call 0 the leading coefficient of the polynomial 0. Let a be an ideal in $A[X]$. The leading coefficients of the polynomials in a form an ideal a' in A , and since A is Noetherian, a' will be finitely generated. Let g_1, \dots, g_m be elements of a whose leading coefficients generate a' , and let r be the maximum degree of g_i . Now let $f \in a$, and suppose f has degree $s > r$, say, $f = aX^s + \dots$. Then $a \in a'$, and so we can write $a = \sum b_i a_i$, $b_i \in A$, $a_i = \text{leading coefficient of } g_i$

Now

$$f - \sum b_i g_i X^{s-r_i}, \quad r_i = \deg(g_i), \text{ has degree}$$

$< \deg(f)$. By continuing in this way, we find that $f \equiv f_t \pmod{(g_1, \dots, g_m)}$ With f_t a polynomial of degree $t < r$. For each $d < r$, let a_d be the subset of A consisting of 0 and the leading coefficients of all polynomials in a of degree d ; it is again an ideal in A . Let

$g_{d,1}, \dots, g_{d,m_d}$ be polynomials of degree d whose leading coefficients generate a_d . Then the same argument as above shows that any polynomial f_d in a of degree d can be written $f_d \equiv f_{d-1} \pmod{(g_{d,1}, \dots, g_{d,m_d})}$ With f_{d-1} of degree $\leq d-1$. On applying this remark repeatedly we find that $f_t \in (g_{r-1,1}, \dots, g_{r-1,m_{r-1}}, \dots, g_{0,1}, \dots, g_{0,m_0})$ Hence

$$f_t \in (g_1, \dots, g_m, g_{r-1,1}, \dots, g_{r-1,m_{r-1}}, \dots, g_{0,1}, \dots, g_{0,m_0})$$

and so the polynomials g_1, \dots, g_{0,m_0} generate a

One of the great successes of category theory in computer science has been the development of a “unified theory” of the constructions underlying denotational semantics. In the untyped λ -calculus, any term may appear in the function position of an application. This means that a model D of the λ -calculus must have the property that given a term t whose interpretation is $d \in D$, Also, the interpretation of a functional abstraction like $\lambda x. x$ is most conveniently defined as a function from D to D , which must then be regarded as an element of D . Let $\psi: [D \rightarrow D] \rightarrow D$ be the function that picks out elements of D to represent elements of $[D \rightarrow D]$ and $\phi: D \rightarrow [D \rightarrow D]$ be the function that maps elements of D to functions of D . Since $\psi(f)$ is intended to represent the function f as an element of D , it makes sense to require that $\phi(\psi(f)) = f$, that is, $\psi \circ \psi = id_{[D \rightarrow D]}$ Furthermore, we often want to view every element of D as representing some function from D to D and require that elements representing the same function be equal – that is $\psi(\phi(d)) = d$

or

$$\psi \circ \phi = id_D$$

The latter condition is called extensionality. These conditions together imply that ϕ and ψ are inverses--- that is, D is isomorphic to the space of functions from D to D that can be the interpretations of functional abstractions: $D \cong [D \rightarrow D]$. Let us suppose we are working with the untyped λ -calculus, we need a solution of the equation $D \cong A + [D \rightarrow D]$, where A is some

predetermined domain containing interpretations for elements of C . Each element of D corresponds to either an element of A or an element of $[D \rightarrow D]$, with a tag. This equation can be solved by finding least fixed points of the function $F(X) = A + [X \rightarrow X]$ from domains to domains --- that is, finding domains X such that $X \cong A + [X \rightarrow X]$, and such that for any domain Y also satisfying this equation, there is an embedding of X to Y --- a pair of maps

$$X \begin{matrix} \xrightarrow{f} \\ \square \\ \xleftarrow{f^R} \end{matrix} Y$$

Such that

$$f^R \circ f = id_X$$

$$f \circ f^R \subseteq id_Y$$

Where $f \subseteq g$ means that f approximates g in some ordering representing their information content. The key shift of perspective from the domain-theoretic to the more general category-theoretic approach lies in considering F not as a function on domains, but as a *functor* on a category of domains. Instead of a least fixed point of the function, F .

Definition 1.3: Let K be a category and $F: K \rightarrow K$ as a functor. A fixed point of F is a pair (A, a) , where A is a **K-object** and $a: F(A) \rightarrow A$ is an isomorphism. A prefixed point of F is a pair (A, a) , where A is a **K-object** and a is any arrow from $F(A)$ to A

Definition 1.4 : An ω -chain in a category K is a diagram of the following form:

$$\Delta = D_0 \xrightarrow{f_0} D_1 \xrightarrow{f_1} D_2 \xrightarrow{f_2} \dots$$

Recall that a cocone μ of an ω -chain Δ is a K -object X and a collection of K -arrows $\{\mu_i: D_i \rightarrow X \mid i \geq 0\}$ such that $\mu_i = \mu_{i+1} \circ f_i$ for all $i \geq 0$. We sometimes write $\mu: \Delta \rightarrow X$ as a reminder of the arrangement of μ 's components. Similarly, a colimit $\mu: \Delta \rightarrow X$ is a cocone with the property that if $\nu: \Delta \rightarrow X'$ is also a cocone then there exists a unique mediating arrow $k: X \rightarrow X'$ such that for all $i \geq 0$, $\nu_i = k \circ \mu_i$. Colimits of ω -chains are sometimes referred to as ω -colimits. Dually, an ω^{op} -chain in K is a diagram of the following form:

$$\Delta = D_0 \xleftarrow{f_0} D_1 \xleftarrow{f_1} D_2 \xleftarrow{f_2} \dots \quad \text{A cone}$$

$\mu: X \rightarrow \Delta$ of an ω^{op} -chain Δ is a \mathbf{K} -object X and a collection of \mathbf{K} -arrows $\{\mu_i: D_i \mid i \geq 0\}$ such that for all $i \geq 0$, $\mu_i = f_i \circ \mu_{i+1}$. An ω^{op} -limit of an ω^{op} -chain Δ is a cone $\mu: X \rightarrow \Delta$ with the property that if $\nu: X' \rightarrow \Delta$ is also a cone, then there exists a unique mediating arrow $k: X' \rightarrow X$ such that for all $i \geq 0$, $\mu_i \circ k = \nu_i$. We write \perp_k (or just \perp) for the distinguished initial object of \mathbf{K} , when it has one, and $\perp \rightarrow A$ for the unique arrow from \perp to each \mathbf{K} -object A . It is also convenient to write $\Delta^- = D_1 \xrightarrow{f_1} D_2 \xrightarrow{f_2} \dots$ to denote all of Δ except D_0 and f_0 . By analogy, μ^- is $\{\mu_i \mid i \geq 1\}$.

For the images of Δ and μ under F we write $F(\Delta) = F(D_0) \xrightarrow{F(f_0)} F(D_1) \xrightarrow{F(f_1)} F(D_2) \xrightarrow{F(f_2)} \dots$

$$\text{and } F(\mu) = \{F(\mu_i) \mid i \geq 0\}$$

We write F^i for the i -fold iterated composition of F that is, $F^0(f) = f, F^1(f) = F(f), F^2(f) = F(F(f))$, etc. With these definitions we can state that every monotonic function on a complete lattice has a least fixed point:

Lemma 1.4. Let \mathbf{K} be a category with initial object \perp and let $F: \mathbf{K} \rightarrow \mathbf{K}$ be a functor. Define the ω -chain Δ by

$$\Delta = \perp \xrightarrow{\perp \rightarrow F(\perp)} F(\perp) \xrightarrow{F(\perp \rightarrow F(\perp))} F^2(\perp) \xrightarrow{F^2(\perp \rightarrow F(\perp))} \dots$$

If both $\mu: \Delta \rightarrow D$ and $F(\mu): F(\Delta) \rightarrow F(D)$ are colimits, then (D, d) is an initial F -algebra, where $d: F(D) \rightarrow D$ is the mediating arrow from $F(\mu)$ to the cocone μ^-

Theorem 1.4 Let a DAG G given in which each node is a random variable, and let a discrete conditional probability distribution of each node given values of its parents in G be specified. Then the product of these conditional distributions yields a joint probability distribution P of the variables, and (G, P) satisfies the Markov condition.

Proof. Order the nodes according to an ancestral ordering. Let X_1, X_2, \dots, X_n be the resultant ordering. Next define.

$$P(x_1, x_2, \dots, x_n) = P(x_n | pa_n) P(x_{n-1} | pa_{n-1}) \dots P(x_2 | pa_2) P(x_1 | pa_1),$$

Where PA_i is the set of parents of X_i of in G and $P(x_i | pa_i)$ is the specified conditional probability distribution. First we show this does indeed yield a joint probability distribution. Clearly, $0 \leq P(x_1, x_2, \dots, x_n) \leq 1$ for all values of the variables. Therefore, to show we have a joint distribution, as the variables range through all their possible values, is equal to one. To that end, Specified conditional distributions are the conditional distributions they notationally represent in the joint distribution. Finally, we show the Markov condition is satisfied. To do this, we need show for $1 \leq k \leq n$ that

$$\text{whenever } P(pa_k) \neq 0, \text{ if } P(nd_k | pa_k) \neq 0 \text{ and } P(x_k | pa_k) \neq 0$$

$$\text{then } P(x_k | nd_k, pa_k) = P(x_k | pa_k),$$

Where ND_k is the set of nondescendants of X_k of in G . Since $PA_k \subseteq ND_k$, we need only show $P(x_k | nd_k) = P(x_k | pa_k)$. First for a given k , order the nodes so that all and only nondescendants of X_k precede X_k in the ordering. Note that this ordering depends on k , whereas the ordering in the first part of the proof does not. Clearly then

$$ND_k = \{X_1, X_2, \dots, X_{k-1}\}$$

Let

$$D_k = \{X_{k+1}, X_{k+2}, \dots, X_n\}$$

follows \sum_{d_k}

We define the m^{th} cyclotomic field to be the field $Q[x]/(\Phi_m(x))$ Where $\Phi_m(x)$ is the m^{th} cyclotomic polynomial. $Q[x]/(\Phi_m(x))$ $\Phi_m(x)$ has degree $\varphi(m)$ over Q since $\Phi_m(x)$ has degree $\varphi(m)$. The roots of $\Phi_m(x)$ are just the primitive m^{th} roots of unity, so the complex embeddings of $Q[x]/(\Phi_m(x))$ are simply the $\varphi(m)$ maps

$$\sigma_k: Q[x]/(\Phi_m(x)) \mapsto C,$$

$$1 \leq k < m, (k, m) = 1, \text{ where}$$

$$\sigma_k(x) = \xi_m^k,$$

ξ_m being our fixed choice of primitive m^{th} root of unity. Note that $\xi_m^k \in Q(\xi_m)$ for every k ; it follows that $Q(\xi_m) = Q(\xi_m^k)$ for all k relatively prime to m . In particular, the images of the σ_i coincide, so $Q[x]/(\Phi_m(x))$ is Galois over Q . This means that we can write $Q(\xi_m)$ for $Q[x]/(\Phi_m(x))$ without much fear of ambiguity; we will do so from now on, the identification being $\xi_m \mapsto x$. One advantage of this is that one can easily talk about cyclotomic fields being extensions of one another, or intersections or compositums; all of these things take place considering them as subfield of C . We now investigate some basic properties of cyclotomic fields. The first issue is whether or not they are all distinct; to determine this, we need to know which roots of unity lie in $Q(\xi_m)$. Note, for example, that if m is odd, then $-\xi_m$ is a $2m^{\text{th}}$ root of unity. We will show that this is the only way in which one can obtain any non- m^{th} roots of unity.

LEMMA 1.5 If m divides n , then $Q(\xi_m)$ is contained in $Q(\xi_n)$

PROOF. Since $\xi_m^{n/m} = \xi_m$, we have $\xi_m \in Q(\xi_n)$, so the result is clear

LEMMA 1.6 If m and n are relatively prime, then

$$Q(\xi_m, \xi_n) = Q(\xi_{mn})$$

and

$$Q(\xi_m) \cap Q(\xi_n) = Q$$

(Recall the $Q(\xi_m, \xi_n)$ is the compositum of $Q(\xi_m)$ and $Q(\xi_n)$)

PROOF. One checks easily that $\xi_m \xi_n$ is a primitive mn^{th} root of unity, so that

$$Q(\xi_{mn}) \subseteq Q(\xi_m, \xi_n)$$

$$\begin{aligned} [Q(\xi_m, \xi_n) : Q] &\leq [Q(\xi_m) : Q][Q(\xi_n) : Q] \\ &= \varphi(m)\varphi(n) = \varphi(mn); \end{aligned}$$

Since $[Q(\xi_{mn}) : Q] = \varphi(mn)$; this implies that

$Q(\xi_m, \xi_n) = Q(\xi_{mn})$ We know that $Q(\xi_m, \xi_n)$ has degree $\varphi(mn)$ over Q , so we must have

$$[Q(\xi_m, \xi_n) : Q(\xi_m)] = \varphi(n)$$

and

$$[Q(\xi_m, \xi_n) : Q(\xi_m)] = \varphi(n)$$

$$[Q(\xi_m) : Q(\xi_m) \cap Q(\xi_n)] \geq \varphi(m)$$

$$\text{And thus that } Q(\xi_m) \cap Q(\xi_n) = Q$$

PROPOSITION 1.2 For any m and n

$$Q(\xi_m, \xi_n) = Q(\xi_{[m,n]})$$

And

$$Q(\xi_m) \cap Q(\xi_n) = Q(\xi_{(m,n)});$$

here $[m, n]$ and (m, n) denote the least common multiple and the greatest common divisor of m and n , respectively.

PROOF. Write $m = p_1^{e_1} \dots p_k^{e_k}$ and $p_1^{f_1} \dots p_k^{f_k}$ where the p_i are distinct primes. (We allow e_i or f_i to be zero)

$$Q(\xi_m) = Q(\xi_{p_1^{e_1}}) Q(\xi_{p_2^{e_2}}) \dots Q(\xi_{p_k^{e_k}})$$

and

$$Q(\xi_n) = Q(\xi_{p_1^{f_1}}) Q(\xi_{p_2^{f_2}}) \dots Q(\xi_{p_k^{f_k}})$$

Thus

$$\begin{aligned} Q(\xi_m, \xi_n) &= Q(\xi_{p_1^{e_1}}) \dots Q(\xi_{p_2^{e_k}}) Q(\xi_{p_1^{f_1}}) \dots Q(\xi_{p_k^{f_k}}) \\ &= Q(\xi_{p_1^{e_1}}) Q(\xi_{p_1^{f_1}}) \dots Q(\xi_{p_k^{e_k}}) Q(\xi_{p_k^{f_k}}) \\ &= Q(\xi_{p_1^{\max(e_1, f_1)}}) \dots Q(\xi_{p_k^{\max(e_k, f_k)}}) \\ &= Q(\xi_{p_1^{\max(e_1, f_1)} \dots p_k^{\max(e_k, f_k)}}) \\ &= Q(\xi_{[m,n]}); \end{aligned}$$

An entirely similar computation shows that $Q(\xi_m) \cap Q(\xi_n) = Q(\xi_{(m,n)})$

Mutual information measures the information transferred when x_i is sent and y_i is received, and is defined as

$$I(x_i, y_i) = \log_2 \frac{P(x_i/y_i)}{P(x_i)} \text{ bits} \quad (1)$$

In a noise-free channel, each y_i is uniquely connected to the corresponding x_i , and so they constitute an input-output pair (x_i, y_i) for which

$$P(x_i/y_j)=1 \text{ and } I(x_i, y_j)=\log_2 \frac{1}{P(x_i)} \text{ bits};$$

that is, the transferred information is equal to the self-information that corresponds to the input x_i . In a very noisy channel, the output y_i and input x_i would be completely uncorrelated, and so $P(x_i/y_j)=P(x_i)$ and also $I(x_i, y_j)=0$; that is,

there is no transference of information. In general, a given channel will operate between these two extremes. The mutual information is defined between the input and the output of a given channel. An average of the calculation of the mutual information for all input-output pairs of a given channel is the average mutual information:

$$I(X, Y) = \sum_{i,j} P(x_i, y_j) I(x_i, y_j) = \sum_{i,j} P(x_i, y_j) \log_2 \left[\frac{P(x_i/y_j)}{P(x_i)} \right]$$

bits per symbol. This calculation is done over the input and output alphabets. The average mutual information. The following expressions are useful for modifying the mutual information expression:

$$P(x_i, y_j) = P(x_i/y_j)P(y_j) = P(y_j/x_i)P(x_i)$$

$$P(y_j) = \sum_i P(y_j/x_i)P(x_i)$$

$$P(x_i) = \sum_j P(x_i/y_j)P(y_j)$$

Then

$$I(X, Y) = \sum_{i,j} P(x_i, y_j) \log_2 \left[\frac{1}{P(x_i)} \right]$$

$$- \sum_{i,j} P(x_i, y_j) \log_2 \left[\frac{1}{P(x_i/y_j)} \right]$$

$$+ \sum_{i,j} P(x_i, y_j) \log_2 \left[\frac{1}{P(x_i)} \right]$$

$$= \sum_i \left[P(x_i/y_j)P(y_j) \right] \log_2 \frac{1}{P(x_i)}$$

$$\sum_i P(x_i) \log_2 \frac{1}{P(x_i)} = H(X)$$

$$I(X, Y) = H(X) - H(X/Y)$$

$$\text{Where } H(X/Y) = \sum_{i,j} P(x_i, y_j) \log_2 \frac{1}{P(x_i/y_j)}$$

is usually called the equivocation. In a sense, the equivocation can be seen as the information lost in the noisy channel, and is a function of the backward conditional probability. The observation of an output symbol y_j provides $H(X) - H(X/Y)$ bits of information. This difference is the mutual information of the channel. *Mutual Information: Properties* Since

$$P(x_i/y_j)P(y_j) = P(y_j/x_i)P(x_i)$$

The mutual information fits the condition

$$I(X, Y) = I(Y, X)$$

And by interchanging input and output it is also true that

$$I(X, Y) = H(Y) - H(Y/X)$$

Where

$$H(Y) = \sum_j P(y_j) \log_2 \frac{1}{P(y_j)}$$

This last entropy is usually called the noise entropy. Thus, the information transferred through the channel is the difference between the output entropy and the noise entropy. Alternatively, it can be said that the channel mutual information is the difference between the number of bits needed for determining a given input symbol before knowing the corresponding output symbol, and the number of bits needed for determining a given input symbol after knowing the corresponding output symbol

$$I(X, Y) = H(X) - H(X/Y)$$

As the channel mutual information expression is a difference between two quantities, it seems that this parameter can adopt negative values. However, and in spite of the fact that for some y_j , $H(X/y_j)$

can be larger than $H(X)$, this is not possible for the average value calculated over all the outputs:

$$\sum_{i,j} P(x_i, y_j) \log_2 \frac{P(x_i/y_j)}{P(x_i)} = \sum_{i,j} P(x_i, y_j) \log_2 \frac{P(x_i, y_j)}{P(x_i)P(y_j)}$$

Then

$$-I(X, Y) = \sum_{i,j} P(x_i, y_j) \frac{P(x_i)P(y_j)}{P(x_i, y_j)} \leq 0$$

Because this expression is of the form

$$\sum_{i=1}^M P_i \log_2 \left(\frac{Q_i}{P_i} \right) \leq 0$$

The above expression can be applied due to the factor $P(x_i)P(y_j)$, which is the product of two probabilities, so that it behaves as the quantity Q_i , which in this expression is a dummy variable that fits the condition $\sum_i Q_i \leq 1$. It can be concluded that the average mutual information is a non-negative number. It can also be equal to zero, when the input and the output are independent of each other. A related entropy called the joint entropy is defined as

$$H(X,Y) = \sum_{i,j} P(x_i, y_j) \log_2 \frac{1}{P(x_i, y_j)}$$

$$= \sum_{i,j} P(x_i, y_j) \log_2 \frac{P(x_i)P(y_j)}{P(x_i, y_j)}$$

$$+ \sum_{i,j} P(x_i, y_j) \log_2 \frac{1}{P(x_i)P(y_j)}$$

Theorem 1.5: Entropies of the binary erasure channel (BEC) The BEC is defined with an alphabet of two inputs and three outputs, with symbol probabilities.

$P(x_1) = \alpha$ and $P(x_2) = 1 - \alpha$, and transition probabilities

$$P(y_3/x_2) = 1 - p \text{ and } P(y_2/x_1) = 0,$$

$$\text{and } P(y_3/x_1) = 0$$

$$\text{and } P(y_1/x_2) = p$$

$$\text{and } P(y_2/x_2) = 1 - p$$

Lemma 1.7. Given an arbitrary restricted time-discrete, amplitude-continuous channel whose restrictions are determined by sets F_n and whose density functions exhibit no dependence on the state s , let n be a fixed positive integer, and $p(x)$ an arbitrary probability density function on Euclidean n -space. $p(y|x)$ for the density $p_n(y_1, \dots, y_n | x_1, \dots, x_n)$ and F for F_n . For any real number a , let

$$A = \left\{ (x, y) : \log \frac{p(y|x)}{p(y)} > a \right\} \quad (1)$$

Then for each positive integer u , there is a code (u, n, λ) such that

$$\lambda \leq ue^{-a} + P\{(X, Y) \notin A\} + P\{X \notin F\}$$

Where

$$P\{(X, Y) \in A\} = \int_A \dots \int p(x, y) dx dy, \quad p(x, y) = p(x)p(y|x)$$

and

$$P\{X \in F\} = \int_F \dots \int p(x) dx$$

Proof: A sequence $x^{(1)} \in F$ such that

$$P\{Y \in A_{x^{(1)}} | X = x^{(1)}\} \geq 1 - \varepsilon$$

where $A_x = \{y : (x, y) \in A\}$;

Choose the decoding set B_1 to be $A_{x^{(1)}}$. Having

chosen $x^{(1)}, \dots, x^{(k-1)}$ and B_1, \dots, B_{k-1} , select $x^k \in F$ such that

$$P\left\{Y \in A_{x^{(k)}} - \bigcup_{i=1}^{k-1} B_i \mid X = x^{(k)}\right\} \geq 1 - \varepsilon;$$

Set $B_k = A_{x^{(k)}} - \bigcup_{i=1}^{k-1} B_i$. If the process does not

terminate in a finite number of steps, then the sequences $x^{(i)}$ and decoding sets $B_i, i = 1, 2, \dots, u$, form the desired code. Thus assume that the process terminates after t steps. (Conceivably $t = 0$). We will show $t \geq u$ by showing that $\varepsilon \leq te^{-a} + P\{(X, Y) \notin A\} + P\{X \notin F\}$. We proceed as follows.

Let

$$B = \bigcup_{j=1}^t B_j. \text{ (If } t = 0, \text{ take } B = \phi). \text{ Then}$$

$$P\{(X, Y) \in A\} = \int_{(x,y) \in A} p(x, y) dx dy$$

$$= \int_x p(x) \int_{y \in A_x} p(y|x) dy dx$$

$$= \int_x p(x) \int_{y \in B \cap A_x} p(y|x) dy dx + \int_x p(x)$$

D. Algorithms

Ideals. Let A be a ring. Recall that an ideal a in A is a subset such that a is a subgroup of A regarded as a group under addition;

$$a \in a, r \in A \Rightarrow ra \in a$$

The ideal generated by a subset S of A is the intersection of all ideals A containing S ----- it is easy to verify that this is in fact an ideal, and that it consist of all finite sums of the form $\sum r_i s_i$ with

$$(2) r_i \in A, s_i \in S. \text{ When } S = \{s_1, \dots, s_m\}, \text{ we shall}$$

write (s_1, \dots, s_m) for the ideal it generates.

Let a and b be ideals in A . The set $\{a+b | a \in a, b \in b\}$ is an ideal, denoted by

$a + b$. The ideal generated by $\{ab \mid a \in a, b \in b\}$ is denoted by ab . Note that $ab \subset a \cap b$. Clearly ab consists of all finite sums $\sum a_i b_i$ with $a_i \in a$ and $b_i \in b$, and if $a = (a_1, \dots, a_m)$ and $b = (b_1, \dots, b_n)$, then $ab = (a_1 b_1, \dots, a_i b_j, \dots, a_m b_n)$. Let a be an ideal of A . The set of cosets of a in A forms a ring A/a , and $a \mapsto a+a$ is a homomorphism $\phi: A \mapsto A/a$. The map $b \mapsto \phi^{-1}(b)$ is a one to one correspondence between the ideals of A/a and the ideals of A containing a . An ideal p is prime if $p \neq A$ and $ab \in p \Rightarrow a \in p$ or $b \in p$. Thus p is prime if and only if A/p is nonzero and has the property that $ab = 0, b \neq 0 \Rightarrow a = 0$, i.e., A/p is an integral domain. An ideal m is maximal if $m \neq A$ and there does not exist an ideal n contained strictly between m and A . Thus m is maximal if and only if A/m has no proper nonzero ideals, and so is a field. Note that m maximal $\Rightarrow m$ prime. The ideals of $A \times B$ are all of the form $a \times b$, with a and b ideals in A and B . To see this, note that if c is an ideal in $A \times B$ and $(a, b) \in c$, then $(a, 0) = (a, b)(1, 0) \in c$ and $(0, b) = (a, b)(0, 1) \in c$. This shows that $c = a \times b$ with

$$a = \{a \mid (a, b) \in c \text{ some } b \in b\}$$

and

$$b = \{b \mid (a, b) \in c \text{ some } a \in a\}$$

Let A be a ring. An A -algebra is a ring B together with a homomorphism $i_B: A \rightarrow B$. A homomorphism of A -algebra $B \rightarrow C$ is a homomorphism of rings $\phi: B \rightarrow C$ such that $\phi(i_B(a)) = i_C(a)$ for all $a \in A$. An A -algebra B is said to be *finitely generated* (or of *finite-type* over A) if there exist elements $x_1, \dots, x_n \in B$ such that every element of B can be expressed as a polynomial in the x_i with coefficients in $i(A)$, i.e., such that the homomorphism $A[X_1, \dots, X_n] \rightarrow B$ sending X_i to x_i is surjective. A ring homomorphism $A \rightarrow B$ is *finite*, and B is finitely generated as an A -module. Let k be a field, and let A be a k -algebra. If $1 \neq 0$ in A , then the map

$k \rightarrow A$ is injective, we can identify k with its image, i.e., we can regard k as a subring of A . If $1=0$ in a ring R , the R is the zero ring, i.e., $R = \{0\}$.

Polynomial rings. Let k be a field. A *monomial* in X_1, \dots, X_n is an expression of the form $X_1^{a_1} \dots X_n^{a_n}$, $a_j \in \mathbb{N}$. The *total degree* of the monomial is $\sum a_i$. We sometimes abbreviate it by X^α , $\alpha = (a_1, \dots, a_n) \in \mathbb{N}^n$. The elements of the polynomial ring $k[X_1, \dots, X_n]$ are finite sums $\sum c_{a_1, \dots, a_n} X_1^{a_1} \dots X_n^{a_n}$, $c_{a_1, \dots, a_n} \in k$, $a_j \in \mathbb{N}$. With the obvious notions of equality, addition and multiplication. Thus the monomials form a basis for $k[X_1, \dots, X_n]$ as a k -vector space. The ring $k[X_1, \dots, X_n]$ is an integral domain, and the only units in it are the nonzero constant polynomials. A polynomial $f(X_1, \dots, X_n)$ is *irreducible* if it is nonconstant and has only the obvious factorizations, i.e., $f = gh \Rightarrow g$ or h is constant. **Division in $k[X]$.** The division algorithm allows us to divide a nonzero polynomial into another: let f and g be polynomials in $k[X]$ with $g \neq 0$; then there exist unique polynomials $q, r \in k[X]$ such that $f = qg + r$ with either $r = 0$ or $\deg r < \deg g$. Moreover, there is an algorithm for deciding whether $f \in (g)$, namely, find r and check whether it is zero. Moreover, the Euclidean algorithm allows to pass from finite set of generators for an ideal in $k[X]$ to a single generator by successively replacing each pair of generators with their greatest common divisor.

(Pure) **lexicographic ordering (lex).** Here monomials are ordered by lexicographic (dictionary) order. More precisely, let $\alpha = (a_1, \dots, a_n)$ and $\beta = (b_1, \dots, b_n)$ be two elements of \mathbb{N}^n ; then $\alpha > \beta$ and $X^\alpha > X^\beta$ (lexicographic ordering) if, in the vector difference $\alpha - \beta \in \mathbb{N}^n$, the left most nonzero entry is positive. For example, $XY^2 > Y^3Z^4$; $X^3Y^2Z^4 > X^3Y^2Z$. Note that this isn't quite how the dictionary would order them: it would put $XXXYYZZZZ$ after $XXXYYZ$.

Graded reverse lexicographic order (grevlex). Here monomials are ordered by total degree, with ties broken by reverse lexicographic ordering. Thus,

$\alpha > \beta$ if $\sum a_i > \sum b_i$, or $\sum a_i = \sum b_i$ and in $\alpha - \beta$ the right most nonzero entry is negative. For example:

$$X^4Y^4Z^7 > X^5Y^5Z^4 \text{ (total degree greater)}$$

$$XY^5Z^2 > X^4YZ^3, \quad X^5YZ > X^4YZ^2$$

Orderings on $k[X_1, \dots, X_n]$. Fix an ordering on the monomials in $k[X_1, \dots, X_n]$. Then we can write an element f of $k[X_1, \dots, X_n]$ in a canonical fashion, by re-ordering its elements in decreasing order. For example, we would write

$$f = 4XY^2Z + 4Z^2 - 5X^3 + 7X^2Z^2$$

as

$$f = -5X^3 + 7X^2Z^2 + 4XY^2Z + 4Z^2 \text{ (lex)}$$

or

$$f = 4XY^2Z + 7X^2Z^2 - 5X^3 + 4Z^2 \text{ (grevlex)}$$

Let $\sum a_\alpha X^\alpha \in k[X_1, \dots, X_n]$, in decreasing order:

$$f = a_{\alpha_0} X^{\alpha_0} + a_{\alpha_1} X^{\alpha_1} + \dots, \quad \alpha_0 > \alpha_1 > \dots, \quad \alpha_0 \neq 0$$

Then we define.

- The *multidegree* of f to be $\text{multdeg}(f) = \alpha_0$;
- The *leading coefficient* of f to be $LC(f) = a_{\alpha_0}$;
- The *leading monomial* of f to be $LM(f) = X^{\alpha_0}$;
- The *leading term* of f to be $LT(f) = a_{\alpha_0} X^{\alpha_0}$

For the polynomial $f = 4XY^2Z + \dots$, the multidegree is (1,2,1), the leading coefficient is 4, the leading monomial is XY^2Z , and the leading term is $4XY^2Z$. **The division algorithm in $k[X_1, \dots, X_n]$.** Fix a monomial ordering in \square^2 .

Suppose given a polynomial f and an ordered set (g_1, \dots, g_s) of polynomials; the division algorithm then constructs polynomials a_1, \dots, a_s and r such that $f = a_1g_1 + \dots + a_s g_s + r$ Where either $r = 0$ or no monomial in r is divisible by any of $LT(g_1), \dots, LT(g_s)$ **Step 1:** If

$LT(g_1) | LT(f)$, divide g_1 into f to get

$$f = a_1g_1 + h, \quad a_1 = \frac{LT(f)}{LT(g_1)} \in k[X_1, \dots, X_n]$$

If $LT(g_1) | LT(h)$, repeat the process until $f = a_1g_1 + f_1$ (different a_1) with $LT(f_1)$ not divisible by $LT(g_1)$. Now divide g_2 into f_1 , and so on, until $f = a_1g_1 + \dots + a_s g_s + r_1$ With $LT(r_1)$ not divisible by any $LT(g_1), \dots, LT(g_s)$

Step 2: Rewrite $r_1 = LT(r_1) + r_2$, and repeat Step 1

with r_2 for f :

$$f = a_1g_1 + \dots + a_s g_s + LT(r_1) + r_3 \text{ (different } a_i \text{'s)}$$

Monomial ideals. In general, an ideal a will contain a polynomial without containing the individual terms of the polynomial; for example, the ideal $a = (Y^2 - X^3)$ contains $Y^2 - X^3$ but not Y^2 or X^3 .

DEFINITION 1.5. An ideal a is *monomial* if

$$\sum c_\alpha X^\alpha \in a \Rightarrow X^\alpha \in a \text{ all } \alpha \text{ with } c_\alpha \neq 0.$$

PROPOSITION 1.3. Let a be a *monomial ideal*, and let $A = \{\alpha | X^\alpha \in a\}$. Then A satisfies the

$$\text{condition } \alpha \in A, \beta \in \square^n \Rightarrow \alpha + \beta \in A \quad (*)$$

And a is the k -subspace of $k[X_1, \dots, X_n]$ generated by the $X^\alpha, \alpha \in A$. Conversely, if A is a subset of \square^n satisfying $(*)$, then the k -subspace a of $k[X_1, \dots, X_n]$ generated by $\{X^\alpha | \alpha \in A\}$ is a monomial ideal.

PROOF. It is clear from its definition that a monomial ideal a is the k -subspace of $k[X_1, \dots, X_n]$ generated by the set of monomials it contains. If $X^\alpha \in a$ and $X^\beta \in k[X_1, \dots, X_n]$.

If a permutation is chosen uniformly and at random from the $n!$ possible permutations in S_n , then the counts $C_j^{(n)}$ of cycles of length j are dependent random variables. The joint distribution of $C^{(n)} = (C_1^{(n)}, \dots, C_n^{(n)})$ follows from Cauchy's formula, and is given by

$$P[C^{(n)} = c] = \frac{1}{n!} N(n, c) = 1 \left\{ \sum_{j=1}^n j c_j = n \right\} \prod_{j=1}^n \left(\frac{1}{j} \right)^{c_j} \frac{1}{c_j!},$$

for $c \in \square_{+}^n$.

Lemma 1.7 For nonnegative integers m_1, \dots, m_n ,

$$E \left(\prod_{j=1}^n (C_j^{(n)})^{m_j} \right) = \left(\prod_{j=1}^n \left(\frac{1}{j} \right)^{m_j} \right) 1 \left\{ \sum_{j=1}^n j m_j \leq n \right\} \quad (1.4)$$

Proof. This can be established directly by exploiting cancellation of the form $c_j^{[m_j]} / c_j! = 1 / (c_j - m_j)!$ when $c_j \geq m_j$, which occurs between the ingredients in Cauchy's formula and the falling factorials in the moments. Write $m = \sum j m_j$. Then, with the first sum indexed by $c = (c_1, \dots, c_n) \in \square_{+}^n$ and the last sum indexed by $d = (d_1, \dots, d_n) \in \square_{+}^n$ via the correspondence $d_j = c_j - m_j$, we have

$$\begin{aligned} E \left(\prod_{j=1}^n (C_j^{(n)})^{m_j} \right) &= \sum_c P[C^{(n)} = c] \prod_{j=1}^n (c_j)^{m_j} \\ &= \sum_{c: c_j \geq m_j \text{ for all } j} 1 \left\{ \sum_{j=1}^n j c_j = n \right\} \prod_{j=1}^n \frac{(c_j)^{m_j}}{j^{c_j} c_j!} \\ &= \prod_{j=1}^n \frac{1}{j^{m_j}} \sum_d 1 \left\{ \sum_{j=1}^n j d_j = n - m \right\} \prod_{j=1}^n \frac{1}{j^{d_j} (d_j)!} \end{aligned}$$

This last sum simplifies to the indicator $1(m \leq n)$, corresponding to the fact that if $n - m \geq 0$, then $d_j = 0$ for $j > n - m$, and a random permutation in S_{n-m} must have some cycle structure (d_1, \dots, d_{n-m}) . The moments of $C_j^{(n)}$ follow immediately as

$$E(C_j^{(n)})^{[r]} = j^{-r} 1\{jr \leq n\} \quad (1.2)$$

We note for future reference that (1.4) can also be written in the form

$$E \left(\prod_{j=1}^n (C_j^{(n)})^{m_j} \right) = E \left(\prod_{j=1}^n Z_j^{m_j} \right) 1 \left\{ \sum_{j=1}^n j m_j \leq n \right\}, \quad (1.3)$$

Where the Z_j are independent Poisson-distribution random variables that satisfy $E(Z_j) = 1/j$

The marginal distribution of cycle counts provides a formula for the joint distribution of the cycle counts C_j^n , we find the distribution of C_j^n using a

combinatorial approach combined with the inclusion-exclusion formula.

Lemma 1.8. For $1 \leq j \leq n$,

$$P[C_j^{(n)} = k] = \frac{j^{-k}}{k!} \sum_{l=0}^{[n/j]-k} (-1)^l \frac{j^{-l}}{l!} \quad (1.1)$$

Proof. Consider the set I of all possible cycles of length j , formed with elements chosen from $\{1, 2, \dots, n\}$, so that $|I| = n^{[j]/j}$. For each $\alpha \in I$, consider the "property" G_α of having α ; that is,

G_α is the set of permutations $\pi \in S_n$ such that α is one of the cycles of π . We then have $|G_\alpha| = (n - j)!$, since the elements of $\{1, 2, \dots, n\}$ not in α must be permuted among themselves. To use the inclusion-exclusion formula we need to calculate the term S_r , which is the sum of the probabilities of the r -fold intersection of properties, summing over all sets of r distinct properties. There are two cases to consider. If the r properties are indexed by r cycles having no elements in common, then the intersection specifies how rj elements are moved by the permutation, and there are $(n - rj)!(rj \leq n)$ permutations in the intersection.

There are $n^{[rj]} / (j^r r!)$ such intersections. For the other case, some two distinct properties name some element in common, so no permutation can have both these properties, and the r -fold intersection is empty. Thus

$$\begin{aligned} S_r &= (n - rj)!(rj \leq n) \\ &\times \frac{n^{[rj]} 1}{j^r r! n!} = 1(rj \leq n) \frac{1}{j^r r!} \end{aligned}$$

Finally, the inclusion-exclusion series for the number of permutations having exactly k properties is

$$\sum_{l \geq 0} (-1)^l \binom{k+l}{l} S_{k+l},$$

Which simplifies to (1.1) Returning to the original hat-check problem, we substitute $j=1$ in (1.1) to obtain the distribution of the number of fixed points of a random permutation. For $k = 0, 1, \dots, n$,

$$P[C_1^{(n)} = k] = \frac{1}{k!} \sum_{l=0}^{n-k} (-1)^l \frac{1}{l!}, \quad (1.2)$$

and the moments of $C_1^{(n)}$ follow from (1.2) with $j=1$. In particular, for $n \geq 2$, the mean and variance of $C_1^{(n)}$ are both equal to 1. The joint distribution of $(C_1^{(n)}, \dots, C_b^{(n)})$ for any $1 \leq b \leq n$ has an expression similar to (1.7); this too can be

derived by inclusion-exclusion. For any $c = (c_1, \dots, c_b) \in \square_+^b$ with $m = \sum c_i$,

$$P[(C_1^{(n)}, \dots, C_b^{(n)}) = c] = \left\{ \prod_{i=1}^b \left(\frac{1}{i} \right)^{c_i} \frac{1}{c_i!} \right\} \sum_{\substack{l \geq 0 \text{ with} \\ \sum l_i \leq n-m}} (-1)^{l_1 + \dots + l_b} \prod_{i=1}^b \left(\frac{1}{i} \right)^{l_i} \frac{1}{l_i!}$$

The joint moments of the first b counts $C_1^{(n)}, \dots, C_b^{(n)}$ can be obtained directly from (1.2) and (1.3) by setting $m_{b+1} = \dots = m_n = 0$

The limit distribution of cycle counts

It follows immediately from Lemma 1.2 that for each fixed j , as $n \rightarrow \infty$,

$$P[C_j^{(n)} = k] \rightarrow \frac{j^{-k}}{k!} e^{-1/j}, \quad k = 0, 1, 2, \dots,$$

So that $C_j^{(n)}$ converges in distribution to a random variable Z_j having a Poisson distribution with mean $1/j$; we use the notation $C_j^{(n)} \rightarrow_d Z_j$ where $Z_j \square P_o(1/j)$ to describe this. Infact, the limit random variables are independent.

Theorem 1.6 The process of cycle counts converges in distribution to a Poisson process of \square with intensity j^{-1} . That is, as $n \rightarrow \infty$,

$$(C_1^{(n)}, C_2^{(n)}, \dots) \rightarrow_d (Z_1, Z_2, \dots) \quad (1.1)$$

Where the $Z_j, j = 1, 2, \dots$ are independent Poisson-distributed random variables with $E(Z_j) = \frac{1}{j}$

Proof. To establish the converges in distribution one shows that for each fixed $b \geq 1$, as $n \rightarrow \infty$,

$$P[(C_1^{(n)}, \dots, C_b^{(n)}) = c] \rightarrow P[(Z_1, \dots, Z_b) = c]$$

Error rates

The proof of Theorem says nothing about the rate of convergence. Elementary analysis can be used to estimate this rate when $b = 1$. Using properties of alternating series with decreasing terms, for $k = 0, 1, \dots, n$,

$$\frac{1}{k!} \left(\frac{1}{(n-k+1)!} - \frac{1}{(n-k+2)!} \right) \leq |P[C_1^{(n)} = k] - P[Z_1 = k]| \leq \frac{1}{k!(n-k+1)!} \quad (1.3)$$

It follows that

$$\frac{2^{n+1}}{(n+1)!} \frac{n}{n+2} \leq \sum_{k=0}^n |P[C_1^{(n)} = k] - P[Z_1 = k]| \leq \frac{2^{n+1} - 1}{(n+1)!} \quad (1.11)$$

Since

$$P[Z_1 > n] = \frac{e^{-1}}{(n+1)!} \left(1 + \frac{1}{n+2} + \frac{1}{(n+2)(n+3)} + \dots \right) < \frac{1}{(n+1)!}$$

We see from (1.11) that the total variation distance between the distribution $L(C_1^{(n)})$ of $C_1^{(n)}$ and the distribution $L(Z_1)$ of Z_1

Establish the asymptotics of $P[A_n(C^{(n)})]$ under conditions (A_0) and (B_{01}) , where

$$A_n(C^{(n)}) = \bigcap_{1 \leq i \leq n} \bigcap_{r_i + 1 \leq j \leq r_i} \{C_{ij}^{(n)} = 0\},$$

and $\zeta_i = (r_i' / r_{id}') - 1 = O(i^{-g'})$ as $i \rightarrow \infty$, for some $g' > 0$. We start with the expression

$$P[A_n(C^{(n)})] = \frac{P[T_{0m}(Z') = n]}{P[T_{0m}(Z) = n]}$$

$$\prod_{\substack{1 \leq i \leq n \\ r_i + 1 \leq j \leq r_i}} \left\{ 1 - \frac{\theta}{ir_i} (1 + E_{i0}) \right\} \quad (1.1)$$

$$P[T_{0n}(Z') = n] = \frac{\theta d}{n} \exp \left\{ \sum_{i \geq 1} [\log(1 + i^{-1} \theta d) - i^{-1} \theta d] \right\}$$

$$\left\{ 1 + O(n^{-1} \phi_{\{1,2,7\}}'(n)) \right\} \quad (1.2)$$

and

$$P[T_{0n}(Z) = n] = \frac{\theta d}{n} \exp \left\{ \sum_{i \geq 1} [\log(1 + i^{-1} \theta d) - i^{-1} \theta d] \right\}$$

$$\left\{ 1 + O(n^{-1} \phi_{\{1,2,7\}}'(n)) \right\} \quad (1.3)$$

Where $\phi_{\{1,2,7\}}'(n)$ refers to the quantity derived from Z' . It thus follows that $P[A_n(C^{(n)})] \square Kn^{-\theta(1-d)}$ for a constant K , depending on Z and the r_i' and computable

explicitly from (1.1) – (1.3), if Conditions (A_0) and (B_{01}) are satisfied and if $\zeta_i^* = O(i^{-g'})$ from some $g' > 0$, since, under these circumstances, both $n^{-1}\phi_{\{1,2,7\}}^*(n)$ and $n^{-1}\phi_{\{1,2,7\}}(n)$ tend to zero as $n \rightarrow \infty$. In particular, for polynomials and square free polynomials, the relative error in this asymptotic approximation is of order n^{-1} if $g' > 1$.

$$\begin{aligned} & \text{For } 0 \leq b \leq n/8 \text{ and } n \geq n_0, \text{ with } n_0 \\ & d_{TV}(L(C[1,b]), L(Z[1,b])) \\ & \leq d_{TV}(L(C[1,b]), L(Z[1,b])) \\ & \leq \varepsilon_{\{7,7\}}(n,b), \end{aligned}$$

Where $\varepsilon_{\{7,7\}}(n,b) = O(b/n)$ under Conditions $(A_0), (D_1)$ and (B_{11}) . Since, by the Conditioning Relation,

$$L(C[1,b] | T_{0b}(C) = l) = L(Z[1,b] | T_{0b}(Z) = l),$$

It follows by direct calculation that

$$\begin{aligned} & d_{TV}(L(C[1,b]), L(Z[1,b])) \\ & = d_{TV}(L(T_{0b}(C)), L(T_{0b}(Z))) \\ & = \max_A \sum_{r \in A} P[T_{0b}(Z) = r] \\ & \left\{ 1 - \frac{P[T_{bn}(Z) = n-r]}{P[T_{0n}(Z) = n]} \right\} \quad (1.4) \end{aligned}$$

Suppressing the argument Z from now on, we thus obtain

$$\begin{aligned} & d_{TV}(L(C[1,b]), L(Z[1,b])) \\ & = \sum_{r \geq 0} P[T_{0b} = r] \left\{ 1 - \frac{P[T_{bn} = n-r]}{P[T_{0n} = n]} \right\} \\ & \leq \sum_{r > n/2} P[T_{0b} = r] + \sum_{r=0}^{[n/2]} \frac{P[T_{0b} = r]}{P[T_{0b} = n]} \\ & \times \left\{ \sum_{s=0}^n P[T_{0b} = s] (P[T_{bn} = n-s] - P[T_{bn} = n-r]) \right\} \\ & \leq \sum_{r > n/2} P[T_{0b} = r] + \sum_{r=0}^{[n/2]} P[T_{0b} = r] \end{aligned}$$

$$\begin{aligned} & \times \sum_{s=0}^{[n/2]} P[T_{0b} = s] \frac{\{P[T_{bn} = n-s] - P[T_{bn} = n-r]\}}{P[T_{0n} = n]} \\ & + \sum_{s=0}^{[n/2]} P[T_{0b} = r] \sum_{s=[n/2]+1}^n P[T = s] P[T_{bn} = n-s] / P[T_{0n} = n] \end{aligned}$$

The first sum is at most $2n^{-1}ET_{0b}$; the third is bound by

$$\begin{aligned} & (\max_{n/2 < s \leq n} P[T_{0b} = s]) / P[T_{0n} = n] \\ & \leq \frac{2\varepsilon_{\{10.5(1)\}}(n/2, b)}{n} \frac{3n}{\theta P_\theta[0,1]}, \\ & \frac{3n}{\theta P_\theta[0,1]} 4n^{-2} \phi_{\{10.8\}}^*(n) \sum_{r=0}^{[n/2]} P[T_{0b} = r] \sum_{s=0}^{[n/2]} P[T_{0b} = s] \frac{1}{2} |r-s| \\ & \leq \frac{12\phi_{\{10.8\}}^*(n)}{\theta P_\theta[0,1]} \frac{ET_{0b}}{n} \end{aligned}$$

Hence we may take

$$\begin{aligned} \varepsilon_{\{7,7\}}(n,b) & = 2n^{-1}ET_{0b}(Z) \left\{ 1 + \frac{6\phi_{\{10.8\}}^*(n)}{\theta P_\theta[0,1]} \right\} P \\ & + \frac{6}{\theta P_\theta[0,1]} \varepsilon_{\{10.5(1)\}}(n/2, b) \quad (1.5) \end{aligned}$$

Required order under Conditions $(A_0), (D_1)$ and (B_{11}) , if $S(\infty) < \infty$. If not, $\phi_{\{10.8\}}^*(n)$ can be replaced by $\phi_{\{10.11\}}^*(n)$ in the above, which has the required order, without the restriction on the r_i implied by $S(\infty) < \infty$. Examining the Conditions $(A_0), (D_1)$ and (B_{11}) , it is perhaps surprising to find that (B_{11}) is required instead of just (B_{01}) ; that is, that we should need $\sum_{l \geq 2} l\varepsilon_{il} = O(i^{-a_1})$ to hold for some $a_1 > 1$. A first observation is that a similar problem arises with the rate of decay of ε_{i1} as well. For this reason, n_1 is replaced by n_1 . This makes it possible to replace condition (A_1) by the weaker pair of conditions (A_0) and (D_1) in the eventual assumptions needed for $\varepsilon_{\{7,7\}}(n,b)$ to be of order $O(b/n)$; the decay rate requirement of order $i^{-1-\gamma}$ is shifted from ε_{i1} itself to its first difference. This is needed to obtain the right approximation error for the random mappings example. However, since all the classical applications make far more stringent assumptions

about the $\varepsilon_{i1}, l \geq 2$, than are made in (B_{11}) . The critical point of the proof is seen where the initial estimate of the difference $P[T_{bn}^{(m)} = s] - P[T_{bn}^{(m)} = s + 1]$. The factor $\varepsilon_{\{10.10\}}(n)$, which should be small, contains a far tail element from n_1 of the form $\phi_1^\theta(n) + u_1^*(n)$, which is only small if $a_1 > 1$, being otherwise of order $O(n^{1-a_1+\delta})$ for any $\delta > 0$, since $a_2 > 1$ is in any case assumed. For $s \geq n/2$, this gives rise to a contribution of order $O(n^{1-a_1+\delta})$ in the estimate of the difference $P[T_{bn} = s] - P[T_{bn} = s + 1]$, which, in the remainder of the proof, is translated into a contribution of order $O(n^{1-a_1+\delta})$ for differences of the form $P[T_{bn} = s] - P[T_{bn} = s + 1]$, finally leading to a contribution of order $bn^{-a_1+\delta}$ for any $\delta > 0$ in $\varepsilon_{\{7.7\}}(n, b)$. Some improvement would seem to be possible, defining the function g by $g(w) = 1_{\{w=s\}} - 1_{\{w=s+t\}}$, differences that are of the form $P[T_{bn} = s] - P[T_{bn} = s + t]$ can be directly estimated, at a cost of only a single contribution of the form $\phi_1^\theta(n) + u_1^*(n)$. Then, iterating the cycle, in which one estimate of a difference in point probabilities is improved to an estimate of smaller order, a bound of the form $|P[T_{bn} = s] - P[T_{bn} = s + t]| = O(n^{-2}t + n^{1-a_1+\delta})$ for any $\delta > 0$ could perhaps be attained, leading to a final error estimate in order $O(bn^{-1} + n^{1-a_1+\delta})$ for any $\delta > 0$, to replace $\varepsilon_{\{7.7\}}(n, b)$. This would be of the ideal order $O(b/n)$ for large enough b , but would still be coarser for small b .

With b and n as in the previous section, we wish to show that

$$\left| d_{TV}(L(C[1, b]), L(Z[1, b])) - \frac{1}{2}(n+1)^{-1} |1 - \theta| E|T_{0b} - ET_{0b}| \right| \leq \varepsilon_{\{7.8\}}(n, b),$$

Where $\varepsilon_{\{7.8\}}(n, b) = O(n^{-1}b[n^{-1}b + n^{-\beta_{12}+\delta}])$ for any $\delta > 0$ under Conditions $(A_0), (D_1)$ and (B_{12}) , with β_{12} . The proof uses sharper estimates. As before, we begin with the formula

$$d_{TV}(L(C[1, b]), L(Z[1, b])) = \sum_{r \geq 0} P[T_{0b} = r] \left\{ 1 - \frac{P[T_{bn} = n - r]}{P[T_{0n} = n]} \right\}_+$$

Now we observe that

$$\left| \sum_{r \geq 0} P[T_{0b} = r] \left\{ 1 - \frac{P[T_{bn} = n - r]}{P[T_{0n} = n]} \right\}_+ - \sum_{r=0}^{\lfloor n/2 \rfloor} \frac{P[T_{0b} = r]}{P[T_{0n} = n]} \right| \times \left| \sum_{s=\lfloor n/2 \rfloor+1}^n P[T_{0b} = s] (P[T_{bn} = n - s] - P[T_{bn} = n - r]) \right| \leq 4n^{-2} ET_{0b}^2 + (\max_{n/2 < s \leq n} P[T_{0b} = s]) / P[T_{0n} = n] + P[T_{0b} > n/2] \leq 8n^{-2} ET_{0b}^2 + \frac{3\varepsilon_{\{10.5(2)\}}(n/2, b)}{\theta P_\theta[0, 1]}, \quad (1.1)$$

We have

$$\left| \sum_{r=0}^{\lfloor n/2 \rfloor} \frac{P[T_{0b} = r]}{P[T_{0n} = n]} \times \left\{ \sum_{s=0}^{\lfloor n/2 \rfloor} P[T_{0b} = s] (P[T_{bn} = n - s] - P[T_{bn} = n - r]) \right\}_+ - \left\{ \sum_{s=0}^{\lfloor n/2 \rfloor} P[T_{0b} = s] \frac{(s-r)(1-\theta)}{n+1} P[T_{0n} = n] \right\}_+ \right| \leq \frac{1}{n^2 P[T_{0n} = n]} \sum_{r \geq 0} P[T_{0b} = r] \sum_{s \geq 0} P[T_{0b} = s] |s - r| \times \left\{ \varepsilon_{\{10.14\}}(n, b) + 2(r \vee s) |1 - \theta| n^{-1} \{K_0 \theta + 4\phi_{\{10.8\}}^*(n)\} \right\} \leq \frac{6}{\theta n P_\theta[0, 1]} ET_{0b} \varepsilon_{\{10.14\}}(n, b) + 4 |1 - \theta| n^{-2} ET_{0b}^2 \{K_0 \theta + 4\phi_{\{10.8\}}^*(n)\} \left(\frac{3}{\theta n P_\theta[0, 1]} \right), \quad (1.2)$$

The approximation in (1.2) is further simplified by noting that

$$\sum_{r=0}^{\lfloor n/2 \rfloor} P[T_{0b} = r] \left\{ \sum_{s=0}^{\lfloor n/2 \rfloor} P[T_{0b} = s] \frac{(s-r)(1-\theta)}{n+1} \right\}_+ - \left\{ \sum_{s=0}^{\lfloor n/2 \rfloor} P[T_{0b} = s] \frac{(s-r)(1-\theta)}{n+1} \right\}_+$$

$$\begin{aligned} &\leq \sum_{r=0}^{\lfloor n/2 \rfloor} P[T_{ob} = r] \sum_{s>\lfloor n/2 \rfloor} P[T_{ob} = s] \frac{(s-r)|1-\theta|}{n+1} \\ &\leq |1-\theta| n^{-1} E(T_{ob} 1\{T_{ob} > n/2\}) \leq 2|1-\theta| n^{-2} ET_{ob}^2, \end{aligned} \quad (1.3)$$

and then by observing that

$$\begin{aligned} &\sum_{r>\lfloor n/2 \rfloor} P[T_{ob} = r] \left\{ \sum_{s \geq 0} P[T_{ob} = s] \frac{(s-r)(1-\theta)}{n+1} \right\} \\ &\leq n^{-1} |1-\theta| (ET_{ob} P[T_{ob} > n/2] + E(T_{ob} 1\{T_{ob} > n/2\})) \\ &\leq 4|1-\theta| n^{-2} ET_{ob}^2 \end{aligned} \quad (1.4)$$

Combining the contributions of (1.2) –(1.3), we thus find

$$\begin{aligned} &|d_{TV}(L(C[1,b]), L(Z[1,b])) \\ &-(n+1)^{-1} \sum_{r \geq 0} P[T_{ob} = r] \left\{ \sum_{s \geq 0} P[T_{ob} = s] (s-r)(1-\theta) \right\}_+ \\ &\leq \varepsilon_{\{7.8\}}(n,b) \\ &= \frac{3}{\theta P_\theta[0,1]} \left\{ \varepsilon_{\{10.5(2)\}}(n/2,b) + 2n^{-1} ET_{ob} \varepsilon_{\{10.14\}}(n,b) \right\} \\ &+ 2n^{-2} ET_{ob}^2 \left\{ 4+3|1-\theta| + \frac{24|1-\theta| \phi_{\{10.8\}}^*(n)}{\theta P_\theta[0,1]} \right\} \end{aligned} \quad (1.5)$$

The quantity $\varepsilon_{\{7.8\}}(n,b)$ is seen to be of the order claimed under Conditions $(A_0), (D_1)$ and (B_{12}) , provided that $S(\infty) < \infty$; this supplementary condition can be removed if $\phi_{\{10.8\}}^*(n)$ is replaced by $\phi_{\{10.11\}}^*(n)$ in the definition of $\varepsilon_{\{7.8\}}(n,b)$, has the required order without the restriction on the r_i implied by assuming that $S(\infty) < \infty$. Finally, a direct calculation now shows that

$$\begin{aligned} &\sum_{r \geq 0} P[T_{ob} = r] \left\{ \sum_{s \geq 0} P[T_{ob} = s] (s-r)(1-\theta) \right\}_+ \\ &= \frac{1}{2} |1-\theta| E|T_{ob} - ET_{ob}| \end{aligned}$$

Example 1.0. Consider the point $O = (0, \dots, 0) \in \square^n$. For an arbitrary vector r , the coordinates of the point $x = O + r$ are equal to the respective coordinates of the vector $r : x = (x^1, \dots, x^n)$ and $r = (x^1, \dots, x^n)$. The vector r such as in the example is called the position vector or the radius vector of the point x . (Or, in greater detail: r is the radius-vector of x w.r.t an origin O). Points are frequently specified by their radius-

vectors. This presupposes the choice of O as the “standard origin”. Let us summarize. We have

considered \square^n and interpreted its elements in two ways: as points and as vectors. Hence we may say that we leading with the two copies of $\square^n : \square^n =$

$\{\text{points}\}, \square^n = \{\text{vectors}\}$
 Operations with vectors: multiplication by a number, addition. Operations with points and vectors: adding a vector to a point (giving a point), subtracting two points (giving a vector). \square^n treated in this way is called an *n-dimensional affine space*. (An “abstract” affine space is a pair of sets, the set of points and the set of vectors so that the operations as above are defined axiomatically). Notice that vectors in an affine space are also known as “free vectors”. Intuitively, they are not fixed at points and “float freely” in space. From \square^n considered as an affine space we can precede in two opposite directions: \square^n as an Euclidean space $\Leftarrow \square^n$ as an affine space $\Rightarrow \square^n$ as a manifold. Going to the left means introducing some extra structure which will make the geometry richer. Going to the right means forgetting about part of the affine structure; going further in this direction will lead us to the so-called “smooth (or differentiable) manifolds”. The theory of differential forms does not require any extra geometry. So our natural direction is to the right. The Euclidean structure, however, is useful for examples and applications. So let us say a few words about it:

Remark 1.0. *Euclidean geometry.* In \square^n considered as an affine space we can already do a good deal of geometry. For example, we can consider lines and planes, and quadric surfaces like an ellipsoid. However, we cannot discuss such things as “lengths”, “angles” or “areas” and “volumes”. To be able to do so, we have to introduce some more definitions, making \square^n a Euclidean space. Namely, we define the length of a vector $a = (a^1, \dots, a^n)$ to be

$$|a| := \sqrt{(a^1)^2 + \dots + (a^n)^2} \quad (1)$$

After that we can also define distances between points as follows:

$$d(A, B) := |\overline{AB}| \quad (2)$$

One can check that the distance so defined possesses natural properties that we expect: is it always non-negative and equals zero only for coinciding points; the distance from A to B is the same as that from B to A (symmetry); also, for three points, A, B and C, we have $d(A, B) \leq d(A, C) + d(C, B)$ (the “triangle inequality”). To define angles, we first introduce the scalar product of two vectors

$$(a, b) := a^1 b^1 + \dots + a^n b^n \quad (3)$$

Thus $|a| = \sqrt{(a, a)}$. The scalar product is also denote by dot: $a \cdot b = (a, b)$, and hence is often referred to as the “dot product”. Now, for nonzero vectors, we define the angle between them by the equality

$$\cos \alpha := \frac{(a, b)}{|a||b|} \quad (4)$$

The angle itself is defined up to an integral multiple of 2π . For this definition to be consistent we have to ensure that the r.h.s. of (4) does not exceed 1 by the absolute value. This follows from the inequality

$$(a, b)^2 \leq |a|^2 |b|^2 \quad (5)$$

known as the Cauchy–Bunyakovsky–Schwarz inequality (various combinations of these three names are applied in different books). One of the ways of proving (5) is to consider the scalar square of the linear combination $a + tb$, where $t \in \mathbb{R}$. As $(a + tb, a + tb) \geq 0$ is a quadratic polynomial in t which is never negative, its discriminant must be less or equal zero. Writing this explicitly yields (5). The triangle inequality for distances also follows from the inequality (5).

Example 1.1. Consider the function $f(x) = x^i$ (the i -th coordinate). The linear function dx^i (the differential of x^i) applied to an arbitrary vector h is simply h^i . From these examples follows that we can rewrite df as

$$df = \frac{\partial f}{\partial x^1} dx^1 + \dots + \frac{\partial f}{\partial x^n} dx^n, \quad (1)$$

which is the standard form. Once again: the partial derivatives in (1) are just the coefficients (depending on x); dx^1, dx^2, \dots are linear functions giving on an arbitrary vector h its coordinates h^1, h^2, \dots , respectively. Hence

$$df(x)(h) = \partial_{hf(x)} = \frac{\partial f}{\partial x^1} h^1 + \dots + \frac{\partial f}{\partial x^n} h^n, \quad (2)$$

Theorem 1.7. Suppose we have a parametrized curve $t \mapsto x(t)$ passing through $x_0 \in \mathbb{R}^n$ at $t = t_0$ and with the velocity vector $x(t_0) = v$. Then

$$\frac{df(x(t))}{dt}(t_0) = \partial_v f(x_0) = df(x_0)(v) \quad (1)$$

Proof. Indeed, consider a small increment of the parameter $t : t_0 \mapsto t_0 + \Delta t$, Where $\Delta t \mapsto 0$. On the other hand, we have $f(x_0 + h) - f(x_0) = df(x_0)(h) + \beta(h)|h|$ for an arbitrary vector h , where $\beta(h) \rightarrow 0$ when $h \rightarrow 0$. Combining it together, for the increment of $f(x(t))$ we obtain

$$\begin{aligned} f(x(t_0 + \Delta t)) - f(x_0) &= df(x_0)(v \cdot \Delta t + \alpha(\Delta t) \Delta t) \\ &+ \beta(v \cdot \Delta t + \alpha(\Delta t) \Delta t) \cdot |v \Delta t + \alpha(\Delta t) \Delta t| \\ &= df(x_0)(v) \cdot \Delta t + \gamma(\Delta t) \Delta t \end{aligned}$$

For a certain $\gamma(\Delta t)$ such that $\gamma(\Delta t) \rightarrow 0$ when $\Delta t \rightarrow 0$ (we used the linearity of $df(x_0)$). By the definition, this means that the derivative of $f(x(t))$ at $t = t_0$ is exactly $df(x_0)(v)$. The statement of the theorem can be expressed by a simple formula:

$$\frac{df(x(t))}{dt} = \frac{\partial f}{\partial x^1} x^1 + \dots + \frac{\partial f}{\partial x^n} x^n \quad (2)$$

To calculate the value Of df at a point x_0 on a given vector v one can take an arbitrary curve passing Through x_0 at t_0 with v as the velocity vector at t_0 and calculate the usual derivative of $f(x(t))$ at $t = t_0$.

Theorem 1.8. For functions $f, g : U \rightarrow \mathbb{R}$, $U \subset \mathbb{R}^n$,

$$d(f + g) = df + dg \quad (1)$$

$$d(fg) = df \cdot g + f \cdot dg \quad (2)$$

Proof. Consider an arbitrary point x_0 and an arbitrary vector v stretching from it. Let a curve $x(t)$ be such that $x(t_0) = x_0$ and $x'(t_0) = v$. Hence

$$d(f + g)(x_0)(v) = \frac{d}{dt}(f(x(t)) + g(x(t)))$$

at $t = t_0$ and

$$d(fg)(x_0)(v) = \frac{d}{dt}(f(x(t))g(x(t)))$$

at $t = t_0$ Formulae (1) and (2) then immediately follow from the corresponding formulae for the

usual derivative Now, almost without change the theory generalizes to functions taking values in \mathbb{R}^m instead of \mathbb{R} . The only difference is that now the differential of a map $F:U \rightarrow \mathbb{R}^m$ at a point x will be a linear function taking vectors in \mathbb{R}^n to vectors in \mathbb{R}^m (instead of \mathbb{R}). For an arbitrary vector $h \in \mathbb{R}^n$,

$$F(x+h) = F(x) + dF(x)(h) + \beta(h)|h| \quad (3)$$

Where $\beta(h) \rightarrow 0$ when $h \rightarrow 0$. We have

$$dF = (dF^1, \dots, dF^m) \text{ and}$$

$$dF = \frac{\partial F}{\partial x^1} dx^1 + \dots + \frac{\partial F}{\partial x^n} dx^n$$

$$= \begin{pmatrix} \frac{\partial F^1}{\partial x^1} & \dots & \frac{\partial F^1}{\partial x^n} \\ \dots & \dots & \dots \\ \frac{\partial F^m}{\partial x^1} & \dots & \frac{\partial F^m}{\partial x^n} \end{pmatrix} \begin{pmatrix} dx^1 \\ \dots \\ dx^n \end{pmatrix} \quad (4)$$

In this matrix notation we have to write vectors as vector-columns.

Theorem 1.9. For an arbitrary parametrized curve $x(t)$ in \mathbb{R}^n , the differential of a map $F:U \rightarrow \mathbb{R}^m$ (where $U \subset \mathbb{R}^n$) maps the velocity vector $\dot{x}(t)$ to the velocity vector of the curve $F(x(t))$ in \mathbb{R}^m :

$$\frac{dF(x(t))}{dt} = dF(x(t))(\dot{x}(t)) \quad (1)$$

Proof. By the definition of the velocity vector,

$$x(t+\Delta t) = x(t) + \dot{x}(t)\Delta t + \alpha(\Delta t)\Delta t \quad (2)$$

Where $\alpha(\Delta t) \rightarrow 0$ when $\Delta t \rightarrow 0$. By the definition of the differential,

$$F(x+h) = F(x) + dF(x)(h) + \beta(h)|h| \quad (3)$$

Where $\beta(h) \rightarrow 0$ when $h \rightarrow 0$. we obtain

$$F(x(t+\Delta t)) = F(x(t) + \underbrace{\dot{x}(t)\Delta t + \alpha(\Delta t)\Delta t}_h)$$

$$= F(x) + dF(x)(\dot{x}(t)\Delta t + \alpha(\Delta t)\Delta t) + \beta(\dot{x}(t)\Delta t + \alpha(\Delta t)\Delta t)|\dot{x}(t)\Delta t + \alpha(\Delta t)\Delta t|$$

$$= F(x) + dF(x)(\dot{x}(t)\Delta t + \gamma(\Delta t)\Delta t)$$

For some $\gamma(\Delta t) \rightarrow 0$ when $\Delta t \rightarrow 0$. This precisely means that $dF(x)\dot{x}(t)$ is the velocity vector of $F(x)$. As every vector attached to a point can be viewed as the velocity vector of some curve passing through this point, this theorem gives a clear geometric picture of dF as a linear map on vectors.

Theorem 1.10 Suppose we have two maps $F:U \rightarrow V$ and $G:V \rightarrow W$, where $U \subset \mathbb{R}^n, V \subset \mathbb{R}^m, W \subset \mathbb{R}^p$ (open domains). Let $F:x \mapsto y = F(x)$. Then the differential of the composite map $GoF:U \rightarrow W$ is the composition of the differentials of F and G :

$$d(GoF)(x) = dG(y)odF(x) \quad (4)$$

Proof. We can use the description of the differential. Consider a curve $x(t)$ in \mathbb{R}^n with the velocity vector \dot{x} . Basically, we need to know to which vector in \mathbb{R}^p it is taken by $d(GoF)$. the curve $(GoF)(x(t)) = G(F(x(t)))$. By the same theorem, it equals the image under dG of the Anycast Flow vector to the curve $F(x(t))$ in \mathbb{R}^m . Applying the theorem once again, we see that the velocity vector to the curve $F(x(t))$ is the image under dF of the vector $\dot{x}(t)$. Hence $d(GoF)(x) = dG(dF(x))$ for an arbitrary vector \dot{x} .

Corollary 1.0. If we denote coordinates in \mathbb{R}^n by (x^1, \dots, x^n) and in \mathbb{R}^m by (y^1, \dots, y^m) , and write

$$dF = \frac{\partial F}{\partial x^1} dx^1 + \dots + \frac{\partial F}{\partial x^n} dx^n \quad (1)$$

$$dG = \frac{\partial G}{\partial y^1} dy^1 + \dots + \frac{\partial G}{\partial y^m} dy^m, \quad (2)$$

Then the chain rule can be expressed as follows:

$$d(GoF) = \frac{\partial G}{\partial y^1} dF^1 + \dots + \frac{\partial G}{\partial y^m} dF^m, \quad (3)$$

Where dF^i are taken from (1). In other words, to get $d(GoF)$ we have to substitute into (2) the expression for $dy^i = dF^i$ from (3). This can also be expressed by the following matrix formula:

$$d(GoF) = \begin{pmatrix} \frac{\partial G^1}{\partial y^1} & \dots & \frac{\partial G^1}{\partial y^m} \\ \dots & \dots & \dots \\ \frac{\partial G^p}{\partial y^1} & \dots & \frac{\partial G^p}{\partial y^m} \end{pmatrix} \begin{pmatrix} \frac{\partial F^1}{\partial x^1} & \dots & \frac{\partial F^1}{\partial x^n} \\ \dots & \dots & \dots \\ \frac{\partial F^m}{\partial x^1} & \dots & \frac{\partial F^m}{\partial x^n} \end{pmatrix} \begin{pmatrix} dx^1 \\ \dots \\ dx^n \end{pmatrix} \quad (4)$$

i.e., if dG and dF are expressed by matrices of partial derivatives, then $d(GoF)$ is expressed by the product of these matrices. This is often written as

$$\begin{pmatrix} \frac{\partial z^1}{\partial x^1} & \dots & \frac{\partial z^1}{\partial x^n} \\ \dots & \dots & \dots \\ \frac{\partial z^p}{\partial x^1} & \dots & \frac{\partial z^p}{\partial x^n} \end{pmatrix} = \begin{pmatrix} \frac{\partial z^1}{\partial y^1} & \dots & \frac{\partial z^1}{\partial y^m} \\ \dots & \dots & \dots \\ \frac{\partial z^p}{\partial y^1} & \dots & \frac{\partial z^p}{\partial y^m} \end{pmatrix} \begin{pmatrix} \frac{\partial y^1}{\partial x^1} & \dots & \frac{\partial y^1}{\partial x^n} \\ \dots & \dots & \dots \\ \frac{\partial y^m}{\partial x^1} & \dots & \frac{\partial y^m}{\partial x^n} \end{pmatrix}, \quad (5)$$

Or

$$\frac{\partial z^\mu}{\partial x^a} = \sum_{i=1}^m \frac{\partial z^\mu}{\partial y^i} \frac{\partial y^i}{\partial x^a}, \quad (6)$$

Where it is assumed that the dependence of $y \in \square^m$ on $x \in \square^n$ is given by the map F , the dependence of $z \in \square^p$ on $y \in \square^m$ is given by the map G , and the dependence of $z \in \square^p$ on $x \in \square^n$ is given by the composition GoF .

Definition 1.6. Consider an open domain $U \subset \square^n$. Consider also another copy of \square^n , denoted for distinction \square_y^n , with the standard coordinates $(y^1 \dots y^n)$. A system of coordinates in the open domain U is given by a map $F: V \rightarrow U$, where $V \subset \square_y^n$ is an open domain of \square_y^n , such that the following three conditions are satisfied :

- (1) F is smooth;
- (2) F is invertible;
- (3) $F^{-1}: U \rightarrow V$ is also smooth

The coordinates of a point $x \in U$ in this system are the standard coordinates of $F^{-1}(x) \in \square_y^n$

In other words,

$$F: (y^1 \dots, y^n) \mapsto x = x(y^1 \dots, y^n) \quad (1)$$

Here the variables $(y^1 \dots, y^n)$ are the “new” coordinates of the point x

Example 1.2. Consider a curve in \square^2 specified in polar coordinates as

$$x(t) : r = r(t), \varphi = \varphi(t) \quad (1)$$

We can simply use the chain rule. The map $t \mapsto x(t)$ can be considered as the composition of the maps $t \mapsto (r(t), \varphi(t)), (r, \varphi) \mapsto x(r, \varphi)$.

Then, by the chain rule, we have

$$\dot{x} = \frac{dx}{dt} = \frac{\partial x}{\partial r} \frac{dr}{dt} + \frac{\partial x}{\partial \varphi} \frac{d\varphi}{dt} = \frac{\partial x}{\partial r} \dot{r} + \frac{\partial x}{\partial \varphi} \dot{\varphi} \quad (2)$$

Here r and φ are scalar coefficients depending on t , whence the partial derivatives $\frac{\partial x}{\partial r}, \frac{\partial x}{\partial \varphi}$ are vectors depending on point in \square^2 . We can compare this with the formula in the “standard” coordinates:

$\dot{x} = e_1 \dot{x} + e_2 \dot{y}$. Consider the vectors $\frac{\partial x}{\partial r}, \frac{\partial x}{\partial \varphi}$. Explicitly we have

$$\frac{\partial x}{\partial r} = (\cos \varphi, \sin \varphi) \quad (3)$$

$$\frac{\partial x}{\partial \varphi} = (-r \sin \varphi, r \cos \varphi) \quad (4)$$

From where it follows that these vectors make a basis at all points except for the origin (where $r = 0$). It is instructive to sketch a picture, drawing vectors corresponding to a point as starting from that point. Notice that $\frac{\partial x}{\partial r}, \frac{\partial x}{\partial \varphi}$ are, respectively,

the velocity vectors for the curves $r \mapsto x(r, \varphi)$ ($\varphi = \varphi_0$ fixed) and

$\varphi \mapsto x(r, \varphi)$ ($r = r_0$ fixed). We can conclude that for an arbitrary curve given in polar coordinates

the velocity vector will have components $(\dot{r}, \dot{\varphi})$ if as a basis we take $e_r := \frac{\partial x}{\partial r}, e_\varphi := \frac{\partial x}{\partial \varphi}$:

$$\dot{x} = e_r \dot{r} + e_\varphi \dot{\varphi} \quad (5)$$

A characteristic feature of the basis e_r, e_φ is that it is not “constant” but depends on point. Vectors “stuck to points” when we consider curvilinear coordinates.

Proposition 1.3. The velocity vector has the same appearance in all coordinate systems.

Proof. Follows directly from the chain rule and the transformation law for the basis e_i . In particular,

the elements of the basis $e_i = \frac{\partial x}{\partial x^i}$ (originally, a formal notation) can be understood directly as the velocity vectors of the coordinate lines $x^i \mapsto x(x^1, \dots, x^n)$ (all coordinates but x^i are fixed). Since we now know how to handle velocities in arbitrary coordinates, the best way to treat the differential of a map $F: \square^n \rightarrow \square^m$ is by its action on the velocity vectors. By definition, we set

$$dF(x_0): \frac{dx(t)}{dt}(t_0) \mapsto \frac{dF(x(t))}{dt}(t_0) \quad (1)$$

Now $dF(x_0)$ is a linear map that takes vectors attached to a point $x_0 \in \square^n$ to vectors attached to the point $F(x) \in \square^m$

$$dF = \frac{\partial F}{\partial x^1} dx^1 + \dots + \frac{\partial F}{\partial x^n} dx^n$$

$$(e_1, \dots, e_m) \begin{pmatrix} \frac{\partial F^1}{\partial x^1} & \dots & \frac{\partial F^1}{\partial x^n} \\ \dots & \dots & \dots \\ \frac{\partial F^m}{\partial x^1} & \dots & \frac{\partial F^m}{\partial x^n} \end{pmatrix} \begin{pmatrix} dx^1 \\ \dots \\ dx^n \end{pmatrix}, \quad (2)$$

In particular, for the differential of a function we always have

$$df = \frac{\partial f}{\partial x^1} dx^1 + \dots + \frac{\partial f}{\partial x^n} dx^n, \quad (3)$$

Where x^i are arbitrary coordinates. The form of the differential does not change when we perform a change of coordinates.

Example 1.3 Consider a 1-form in \square^2 given in the standard coordinates:

$A = -ydx + xdy$ In the polar coordinates we will have $x = r \cos \varphi, y = r \sin \varphi$, hence

$$dx = \cos \varphi dr - r \sin \varphi d\varphi$$

$$dy = \sin \varphi dr + r \cos \varphi d\varphi$$

Substituting into A , we get

$$A = -r \sin \varphi (\cos \varphi dr - r \sin \varphi d\varphi)$$

$$+ r \cos \varphi (\sin \varphi dr + r \cos \varphi d\varphi)$$

$$= r^2 (\sin^2 \varphi + \cos^2 \varphi) d\varphi = r^2 d\varphi$$

Hence $A = r^2 d\varphi$ is the formula for A in the polar coordinates. In particular, we see that this is again a 1-form, a linear combination of the differentials of coordinates with functions as coefficients. Secondly, in a more conceptual way, we can define a 1-form in a domain U as a linear function on vectors at every point of U :

$$\omega(v) = \omega_1 v^1 + \dots + \omega_n v^n, \quad (1)$$

If $v = \sum e_i v^i$, where $e_i = \frac{\partial x}{\partial x^i}$. Recall that the differentials of functions were defined as linear functions on vectors (at every point), and

$$dx^i(e_j) = dx^i \left(\frac{\partial x}{\partial x^j} \right) = \delta_j^i \quad (2) \quad \text{at}$$

every point x .

Theorem 1.9. For arbitrary 1-form ω and path γ , the integral $\int_\gamma \omega$ does not change if we change parametrization of γ provide the orientation remains the same.

Proof: Consider $\left\langle \omega(x(t)), \frac{dx}{dt} \right\rangle$ and

$\left\langle \omega(x(t(t'))), \frac{dx}{dt} \right\rangle$ As

$$\left\langle \omega(x(t(t'))), \frac{dx}{dt} \right\rangle = \left\langle \omega(x(t(t'))), \frac{dx}{dt} \right\rangle \cdot \frac{dt}{dt},$$

Let p be a rational prime and let $K = \square(\zeta_p)$. We write ζ for ζ_p or this section. Recall that K has degree $\varphi(p) = p-1$ over \square . We wish to show that $O_K = \square[\zeta]$. Note that ζ is a root of $x^p - 1$, and thus is an algebraic integer; since O_K is a ring we have that $\square[\zeta] \subseteq O_K$. We give a proof without assuming unique factorization of ideals. We begin

with some norm and trace computations. Let j be an integer. If j is not divisible by p , then ζ^j is a primitive p^{th} root of unity, and thus its conjugates are $\zeta, \zeta^2, \dots, \zeta^{p-1}$. Therefore

$$\text{Tr}_{K/\mathbb{Q}}(\zeta^j) = \zeta + \zeta^2 + \dots + \zeta^{p-1} = \Phi_p(\zeta) - 1 = -1$$

If p does divide j , then $\zeta^j = 1$, so it has only the one conjugate 1, and $\text{Tr}_{K/\mathbb{Q}}(\zeta^j) = p - 1$. By linearity of the trace, we find that

$$\begin{aligned} \text{Tr}_{K/\mathbb{Q}}(1 - \zeta) &= \text{Tr}_{K/\mathbb{Q}}(1 - \zeta^2) = \dots \\ &= \text{Tr}_{K/\mathbb{Q}}(1 - \zeta^{p-1}) = p \end{aligned}$$

We also need to compute the norm of $1 - \zeta$. For this, we use the factorization

$$\begin{aligned} x^{p-1} + x^{p-2} + \dots + 1 &= \Phi_p(x) \\ &= (x - \zeta)(x - \zeta^2) \dots (x - \zeta^{p-1}); \end{aligned}$$

Plugging in $x = 1$ shows that

$$p = (1 - \zeta)(1 - \zeta^2) \dots (1 - \zeta^{p-1})$$

Since the $(1 - \zeta^j)$ are the conjugates of $(1 - \zeta)$, this shows that $N_{K/\mathbb{Q}}(1 - \zeta) = p$. The key result for determining the ring of integers O_K is the following.

LEMMA 1.9

$$(1 - \zeta)O_K \cap \mathbb{Z} = p\mathbb{Z}$$

Proof. We saw above that p is a multiple of $(1 - \zeta)$ in O_K , so the inclusion $(1 - \zeta)O_K \cap \mathbb{Z} \supseteq p\mathbb{Z}$ is immediate. Suppose now that the inclusion is strict. Since $(1 - \zeta)O_K \cap \mathbb{Z}$ is an ideal of \mathbb{Z} containing $p\mathbb{Z}$ and $p\mathbb{Z}$ is a maximal ideal of \mathbb{Z} , we must have $(1 - \zeta)O_K \cap \mathbb{Z} = p\mathbb{Z}$. Thus we can write

$$1 = \alpha(1 - \zeta)$$

For some $\alpha \in O_K$. That is, $1 - \zeta$ is a unit in O_K .

COROLLARY 1.1 For any $\alpha \in O_K$,

$$\text{Tr}_{K/\mathbb{Q}}((1 - \zeta)\alpha) \in p\mathbb{Z}$$

PROOF. We have

$$\begin{aligned} \text{Tr}_{K/\mathbb{Q}}((1 - \zeta)\alpha) &= \sigma_1((1 - \zeta)\alpha) + \dots + \sigma_{p-1}((1 - \zeta)\alpha) \\ &= \sigma_1(1 - \zeta)\sigma_1(\alpha) + \dots + \sigma_{p-1}(1 - \zeta)\sigma_{p-1}(\alpha) \\ &= (1 - \zeta)\sigma_1(\alpha) + \dots + (1 - \zeta^{p-1})\sigma_{p-1}(\alpha) \end{aligned}$$

Where the σ_i are the complex embeddings of K (which we are really viewing as automorphisms of K) with the usual ordering. Furthermore, $1 - \zeta^j$ is a multiple of $1 - \zeta$ in O_K for every $j \neq 0$. Thus $\text{Tr}_{K/\mathbb{Q}}(\alpha(1 - \zeta)) \in (1 - \zeta)O_K$. Since the trace is also a rational integer.

PROPOSITION 1.4 Let p be a prime number and let $K = \mathbb{Q}(\zeta_p)$ be the p^{th} cyclotomic field. Then $O_K = \mathbb{Z}[\zeta_p] \cong \mathbb{Z}[x]/(\Phi_p(x))$; Thus $1, \zeta_p, \dots, \zeta_p^{p-2}$ is an integral basis for O_K .

PROOF. Let $\alpha \in O_K$ and write

$$\alpha = a_0 + a_1\zeta + \dots + a_{p-2}\zeta^{p-2} \quad \text{With } a_i \in \mathbb{Z}.$$

Then

$$\begin{aligned} \alpha(1 - \zeta) &= a_0(1 - \zeta) + a_1(\zeta - \zeta^2) + \dots \\ &+ a_{p-2}(\zeta^{p-2} - \zeta^{p-1}) \end{aligned}$$

By the linearity of the trace and our above calculations we find that $\text{Tr}_{K/\mathbb{Q}}(\alpha(1 - \zeta)) = pa_0$

We also have

$\text{Tr}_{K/\mathbb{Q}}(\alpha(1 - \zeta)) \in p\mathbb{Z}$, so $a_0 \in \mathbb{Z}$. Next consider the algebraic integer

$(\alpha - a_0)\zeta^{-1} = a_1 + a_2\zeta + \dots + a_{p-2}\zeta^{p-3}$; This is an algebraic integer since $\zeta^{-1} = \zeta^{p-1}$ is. The same argument as above shows that $a_1 \in \mathbb{Z}$, and continuing in this way we find that all of the a_i are in \mathbb{Z} . This completes the proof.

Example 1.4 Let $K = \mathbb{Q}$, then the local ring $\mathbb{Z}_{(p)}$ is simply the subring of \mathbb{Q} of rational numbers with denominator relatively prime to p . Note that this ring $\mathbb{Z}_{(p)}$ is not the ring \mathbb{Z}_p of p -adic integers; to get \mathbb{Z}_p one must complete $\mathbb{Z}_{(p)}$. The usefulness of $O_{K,p}$ comes from the fact that it has a particularly simple ideal structure. Let a be any proper ideal of $O_{K,p}$ and consider the ideal $a \cap O_K$ of O_K . We claim that $a = (a \cap O_K)O_{K,p}$; That is, that a is generated by the elements of a in $a \cap O_K$. It is clear from the definition of an ideal that $a \supseteq (a \cap O_K)O_{K,p}$. To prove the other inclusion, let α be any element of a . Then we can write $\alpha = \beta/\gamma$ where $\beta \in O_K$ and $\gamma \notin p$. In particular, $\beta \in a$ (since $\beta/\gamma \in a$ and a is an

ideal), so $\beta \in O_K$ and $\gamma \notin p$. so $\beta \in a \cap O_K$.
 Since $1/\gamma \in O_{K,p}$, this implies that
 $\alpha = \beta/\gamma \in (a \cap O_K)O_{K,p}$, as claimed. We can
 use this fact to determine all of the ideals of $O_{K,p}$.
 Let a be any ideal of $O_{K,p}$ and consider the ideal
 factorization of $a \cap O_K$ in O_K . write it as
 $a \cap O_K = p^n b$ For some n and some ideal b ,
 relatively prime to p . we claim first that
 $bO_{K,p} = O_{K,p}$. We now find that
 $a = (a \cap O_K)O_{K,p} = p^n bO_{K,p} = p^n O_{K,p}$
 Since $bO_{K,p} = O_{K,p}$. Thus every ideal of $O_{K,p}$ has the
 form $p^n O_{K,p}$ for some n ; it follows immediately
 that $O_{K,p}$ is noetherian. It is also now clear that
 $p^n O_{K,p}$ is the unique non-zero prime ideal in $O_{K,p}$.
 Furthermore, the inclusion $O_K \mapsto O_{K,p} / pO_{K,p}$
 Since $pO_{K,p} \cap O_K = p$, this map is also
 surjection, since the residue class of $\alpha/\beta \in O_{K,p}$
 (with $\alpha \in O_K$ and $\beta \notin p$) is the image of $\alpha\beta^{-1}$
 in $O_{K/p}$, which makes sense since β is invertible
 in $O_{K/p}$. Thus the map is an isomorphism. In
 particular, it is now abundantly clear that every non-
 zero prime ideal of $O_{K,p}$ is maximal. To

show that $O_{K,p}$ is a Dedekind domain, it remains to
 show that it is integrally closed in K . So let $\gamma \in K$
 be a root of a polynomial with coefficients in
 $O_{K,p}$; write this polynomial as
 $x^m + \frac{\alpha_{m-1}}{\beta_{m-1}} x^{m-1} + \dots + \frac{\alpha_0}{\beta_0}$ With $\alpha_i \in O_K$ and
 $\beta_i \in O_{K-p}$. Set $\beta = \beta_0 \beta_1 \dots \beta_{m-1}$. Multiplying by
 β^m we find that $\beta\gamma$ is the root of a monic
 polynomial with coefficients in O_K . Thus
 $\beta\gamma \in O_K$; since $\beta \notin p$, we have
 $\beta\gamma/\beta = \gamma \in O_{K,p}$. Thus $O_{K,p}$ is integrally close
 in K .

COROLLARY 1.2. Let K be a number field of
 degree n and let α be in O_K then
 $N'_{K/\mathbb{Q}}(\alpha O_K) = |N_{K/\mathbb{Q}}(\alpha)|$

PROOF. We assume a bit more Galois theory than
 usual for this proof. Assume first that K/\mathbb{Q} is
 Galois. Let σ be an element of $Gal(K/\mathbb{Q})$. It is
 clear that $\sigma(O_K)/\sigma(\alpha) \cong O_{K/\alpha}$; since
 $\sigma(O_K) = O_K$, this shows that
 $N'_{K/\mathbb{Q}}(\sigma(\alpha)O_K) = N'_{K/\mathbb{Q}}(\alpha O_K)$. Taking the
 product over all $\sigma \in Gal(K/\mathbb{Q})$, we have
 $N'_{K/\mathbb{Q}}(N_{K/\mathbb{Q}}(\alpha)O_K) = N'_{K/\mathbb{Q}}(\alpha O_K)^n$ Since
 $N_{K/\mathbb{Q}}(\alpha)$ is a rational integer and O_K is a free \mathbb{Q} -
 module of rank n ,
 $O_K/N_{K/\mathbb{Q}}(\alpha)O_K$ Will have order $N_{K/\mathbb{Q}}(\alpha)^n$;
 therefore

$$N'_{K/\mathbb{Q}}(N_{K/\mathbb{Q}}(\alpha)O_K) = N_{K/\mathbb{Q}}(\alpha O_K)^n$$

This completes the proof. In the general case, let L
 be the Galois closure of K and set $[L:K] = m$.

III. TIME AUTOMATED BASED FUZZY CONTROLLERS

This section is devoted to describe the
 Timed Automata based Fuzzy Control. This novel
 fuzzy inference engine improves classic inference
 methods in terms of dynamism and ability to model
 variable structure systems that are characterized
 from a discontinuous nonlinear behaviour. As
 mentioned, this kind of behaviour strongly
 characterizes power systems. Consequently, a
 standard control system could not be sufficient to
 manage a so complex framework and an alternative
 scenario is necessary. This scenario is provided by
 Timed Automata based Fuzzy Controllers. Before
 formally defining TAFCS, it is necessary to
 introduce timed fuzzy controllers, i.e, a novel fuzzy
 inference engine that extends fuzzy control idea with
 three additional concepts: control configuration,
 control era and control time. In detail, a timed fuzzy
 controller uses the control time as a clock moving
 the controlled systems through several time
 intervals, known as control eras. Each control era is
 characterized by a specific control configuration
 which is formed by: 1) the number and typology of
 fuzzy variables and 2) the number and structure of
 relationships among variables.

A. Building a business case for Smart Grids

In its most general terms, a business case
 provides the basic rationale for investment in
 projects for business change. In the smart grids
 arena, the entities looking into building business
 cases are primarily network operators and possibly
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 electricity retailers and now emerging players like
 generation and demand aggregators-see, for

example, [2], [3]. All these players from the supply-side of the business may see benefits in terms of primary plant (i. e. copper, steel and concrete) investment avoidance or deferral, increased transfer capacities, new markets, and so on. On the other hand, end-consumers (at least currently and for the vast majority of them) see no rational business case for smart grids. This is where the problem arises. The major investments in smart grid technologies are vied not to provide nothing much new to consumers in the short term-aside possibly from more information about consumption and bills. The announced smart grids revolution has nothing revolutionary like the advent of the Internet, mobile communications and electrification which were all radically new products and services. The best example to date here is the Telegestore project of ENEL Distribuzione S.p.A. which installed over 30 million advanced meters in Italy over the last decade. Most, if not all, of the benefits from this project were accrued by the utility-through reduction in non-technical losses, streamlined automated meter reading, remote-controlled consumer disconnection, etc. The only apparent benefit for the consumers at the moment is the possibility to adopt different pricing structures (e.g. time of use pricing), which many consumers are not keen to adopt in the first place. At the same time, it is not clear whether all the savings generated by the installation of those smart meters are being passed on to the consumers. In the short term, smart grids will not provide the means to consume "premium electrons". Smart grid technologies will benefit consumers only in the long run by enabling the decarbonization of the electricity sector. Hence, we are facing a severe imbalance of costs and benefits over time. Consumers will have to bear the cost of the infrastructure in the short term while it is not clear when they will start benefitting from it. This contrasts with the supply-side who may see benefits quite early on.

B. The accounting challenges

The activities of the Power System Economics Subcommittee are at their infancy when considering the cost-benefit accounting of the smart grid. Cost accounting remains the easy part here. Significant work remains to be done especially in the area of benefit quantification, however. Quantifying benefits in the electricity business is not a slim task. It is relatively straightforward in theory to quantify the benefits and possible externalities caused by the deployment of a specific smart grid technology. This can be achieved using computational general equilibrium (CGE) models which can be used to predict the effects of investment and policy decisions on entire segments of the economy. The main problem with CGE models is that they are as good as their underlying assumptions and their supporting econometric databases. An externality is a negative side-effect caused to one or more

economic agents by the actions of another agent. The prime example here is pollution (externality) as a side-effect of fossil-fuel electricity generation. CGE models may well lack the robustness required to justify the massive investments in smart grid technologies. In fact, this problem is clearly exacerbated by the other exogenous factors which will evolve over the lifetime of the technologies (be it government policies and targets, poor reliability of the technology, fuel and CO₂ emission credits price shocks, consumer behavior, etc.) The challenge is therefore to steer efforts to encounter more robust and realistic ways to assess benefits, especially when consumers need to be modeled explicitly. Agent-based simulation appears to have become the de facto standard for assessing strategic behavior in electricity markets in the past two years in replacement of equilibrium models. There is scope to study whether the successes of agent-based simulation could be translated into the benefit evaluation arena. The field is wide open at the moment to develop radically new ways of estimating the future benefits of smart grids. The Power System Economics Subcommittee will be at the forefront to coordinate those efforts.

IV. ALLOCATING THE COSTS AND BENEFITS: CHALLENGES FOR UTILITY REGULATION

The seeming imbalance of costs and benefits between the supply- and the demand-side of the electricity business represents a major challenge for utility regulators. Ultimately in most cases, the regulators are responsible for the giving the go/no-go signal for smart grid investments, especially for network operators. They have assess how best to structure the rate base of the utilities which will allow for beneficial and sustainable investments. In this, the regulators have to conduct a balancing act trying to allocate the costs and the benefits in a socially acceptable manner over time. Finding such win-win situations should prove difficult in general.

A. A giving the right kick

Nonetheless, the utility industry, because of its institutional risk aversion, remains reluctant to embark onto significant changes in its technologies and its processes. It is also the role of the regulators to give the proper incentives to utilities to take up and get accustomed some of the smart grid technologies. It is without a doubt that these will make the possible the operation of networks with close to 50% of energy coming from renewables and while bearing increased loading from the electrification of the road transportation sector. A good example of such proactive behavior by a regulator is the recent announcement of a £6.5 billion proposed investment over the period 2010-2015 by the British regulator Of gem [4]. Out of this envelope, £500 million will go to demonstration

projects for "low carbon networks", i.e. smart grids. The questions for the Power System Economics Subcommittee here are really about by how much should the regulators be pushing the utilities and under what kind of incentive mechanisms. Likewise, we have to be aware of the value proposition this is entailing for the consumers, which still is begging the question about how consumers will end up benefiting against what they are paying-in the British case, the estimated price tag per consumer is expected to be less than £4 per year [4].

V. COMMUNICATION CHALLENGES IN SMART GRID

The Smart Grid will be characterized by tight integration of a flexible and secure communications network with novel energy management techniques requiring a very large number of sensor and actuator nodes. Therefore, out of many different technology provisions that are required for the realization of the Smart Grid, communication networks become the key enabler for the Smart Grid [10, 12]. The communications network will not only facilitate advanced control and monitoring but also support extension of participation of generation, transmission, marketing and service provision to new interested parties. However, the greatest challenge for communications technology is to provide robust, secure and interoperable networks.

- Interoperability [12]: The Smart Grid will connect a large number of components from disparate distribution and transmission networks, generating sources and consumers. Further, the Smart Grid will consist of heterogeneous network architectures, technologies and standards. For instance, short range wireless (e.g. Bluetooth, UWB, Wi-Fi, ZigBee) and wireline (e.g. PLC, Ethernet) could be used for interfacing devices in the local area (e.g. to provide communications for home automation, smart metering, substation automation or power generation control systems) while cellular (e.g. GPRS), 4G technologies (e.g. 802.16m and LTE) or wired broadband (e.g. xDSL, HFC, FTTH) could be used for wide area networking. Achieving the interoperability of communication systems and architectures supporting the Smart Grids would require agreement on the usage, an interpretation of interfaces and messages that can seamlessly bridge different standards and technologies.
- Security and Privacy [11, 12]: In Smart Grid, grid operations are integrated with ICT to facilitate efficient monitoring, control and management of systems over bi-directional communication links (e.g. in smart metering, building energy management, and load balancing applications). Smart Grid security risks are not only mainly arising from physical vulnerabilities as in the case of traditional grid, but also associated with the communications systems. It

presents potential attackers to manipulate services to homes and business for instance by gaining the control of smart meters and disrupting load balancing by sudden increase or decrease of power demand. Hence, the communication networks of the Smart Grid require substantial security precautions. Another important aspect is privacy issues associated in particular with the consumer energy usage data (i.e. collected by smart meters). While sensor and meter data could enormously benefit grid operations for the improved efficiency of energy consumption, adequate data protection mechanisms are required to safeguard the privacy and commercial value of energy related data. Since still it is not clear how the Smart Grid will evolve in the future, the major challenge for developing communications solutions would be to understand the future Smart Grid system architecture/s and its implications to communications solutions. One way forward would be to analyze various architectural options for future energy grid and to identify communications requirements between key building blocks such as generation, transmission, distribution, and consumers (including residential, commercial and industrial). Some of the generic engineering and research challenges in the communications technology area are summarized below.

- Novel flexible networking architectures: Each smart grid application will have different expectations from the communications networks, in terms of reliability, Quality of Service (QoS) and capacity. Further, overall network and systems management requirements for the Smart Grid networking infrastructure needs to be identified to realize efficient and scalable networks; this would entail the investigation of a plethora of different heterogeneous wireless and wireline networks (IPv6 over IEEE 802.15.4, HomePlug, Cellular networks and connectivity to the Internet).
- QoS provisioning: State of the art QoS concepts considered by most communications networks and application developers will not be sufficient in the smart grid networks. For instance, networks will have to meet not only guaranteed delivery but also strict deadlines on message level delivery times. This raises need for defining new QoS metrics and developing solutions considering application, service and infrastructure requirements of the Smart Grid. Techniques to integrate and implement these novel QoS requirements will also need to be developed.
- Techniques to facilitate reliability and fault-tolerance: The Smart Grid applications will require end-to-end connectivity between energy providers, energy consumers, and their respective equipment. Depending on the application at hand smart energy messages envisioned to flow between the above entities using a number of different networks.

Traditional transport layer protocols, such as TCP, are known to have several limitations especially in the cases where the end-to-end path has multiple wireless links. Hence, for the reliable exchange of smart energy messages optimized networking solutions need to be investigated.

- Self organizing and optimizing networking: Creating a real-time energy market via the use of the Smart Grid communication infrastructure requires networking management techniques that in an autonomous way react and gracefully adapt to changing network conditions. That would be an important feature of the Smart Grid communication network to allow minimum operational cost while at the same time be able to provide a flexible and automated flow of smart energy messages.

- Security and Privacy [12]: Security of smart grid networks needs to be achieved not only by technology but also through policies and processes dictated by deployment models. Bi-directional communication

A. Generalities on SVR and TVR

The basic concepts of SVR are summarized here to permit the understanding of the proposed control system's structure, performance and advantages [1,2,3]:

- In modern power systems it is critical to have automatic real time voltage control at hundreds of transmission busses. Such a control is in fact too complex, unreliable and uneconomical;
- The reactive power available at generating units is a simple and low cost means that could be used to control the voltages;
- A realistic simple voltage control system should consider the dominant "strongest" buses only (a small amount called Pilot Nodes (PN)), thus allowing a suboptimal solution based on difficult to control PN but greatly affecting the voltages of the other busses around them. PN are selected based on short circuit powers and the sensitivity matrix computation criteria.
- The busses whose voltages are greatly affected by the PN voltage are electrically coupled to it and form a Control Area (CA). Per definition the bus voltages in a given CA would be similar – i.e. bus voltages in CA move in unison;
- The control structure, based on the grid subdivision into CA with electrically decoupled PN, automatically and, as much as possible, independently regulates each area pilot node voltage;
- The reactive power used to achieve the above voltage control is sourced from the largest units in the CA that have the greatest effect on the PN voltage. The need to increase the operating security and efficiency of the transmission grid using

centralized coordination of the decentralized SVR structure is the basic idea behind TVR. This coordination happens as follows:

- The pilot nodes voltage set points must be adequately updated and coordinated with dynamics slower than SVR, by considering the real time condition of the overall grid and avoiding useless and conflicting inter-area control efforts;
- The pilot nodes voltage set points have to be computed and updated simultaneously in real-time, considering the global control system structure and its real-time measurements;
- The pilot nodes voltage set points have to be optimized to minimize grid losses while still preserving control margins.

An example of a control system based on the above concepts is the one defined for the South Africa power system grid subdivision into electric areas around the pilot nodes (buses selected from the strongest ones through an automatic procedure based on shortcircuit powers and sensitivity matrices). The SVR areas are controlled by signals of "reactive power level (q_{lev})", one for each area, provided by the Regional Voltage Regulator (RVR), which maintains the pilot nodes voltages in the Region areas at the desired values, through controlling in real-time the reactive powers of those generators which most influence the pilot nodes buses. A Reactive Power Regulator (PQR) that acts directly on the Automatic Voltage Regulator (AVR) set point ensures that q_{lev} is achieved at each CG in CA(i). In the TVR framework the National Voltage Regulator (NVR), situated at country/utility level, controls in closed loop the RVR pilot node voltage set-points for a secure and economic operation and hence, establishes the grid "voltage profile" according to both the actual network state and the long or short-term forecasting of optimal voltages and reactive power levels given by the Loss Minimization Control (LMC). The above hierarchical structure is sufficient to ensure that the grid could be operated at high voltages (close to the maximum continuous voltage) while network security is maintained. The former and latter is achieved via real time control of the main generators that, under SVR control, seldom if ever operate at their reactive limits. The voltage at the PN_i represents the voltage profile of the CA_i busses the operator controls by simply choosing the PN_i voltage. Therefore the CA_i voltages change in unison, in front of local loads variation or network perturbations, according to the trend of the area PN_i voltage. The SVR reactive power level $q_j(t)$ of the j-area, represents at the "t" instant the control effort under-way at the j-area, that is the real-time reactive power load for the j-area control units. More precisely the $q_j(t)$ value stands for the reactive power percentage of the j-area units with respect to the corresponding under or over-excitation limits: in particular when $q_j(t)$ reaches +1 the j-area voltage

regulation is over-saturated, because the operating points of all the j-area control generators/SVCs are fixed by their over-excitation limits. Therefore, increasing the load at a given area, the local pilot node voltage is regulated by SVR to the desired value, unless all the area control units reach their over-excitation limits. Under TVR, this extreme operating condition approaching is certainly due to the achievement of the area voltage instability limit [7]. Before that critical state, the RVR also operates, via high-speed telecommunications, the turning on/off of reactor banks and shunt capacitor, the up/down switching of the OLTCs and FACTS controller set points, till to command the area OLTCs blocking. According to these real time functionalities and because the voltage degradation does usually take some minutes for moving from the initial instability to the irreversible collapse, it appears reasonable, simple and effective the direct computing, inside the SVR, of a real-time and online indicator of the j-area proximity to voltage instability, mainly based on the actual value of the area reactive power level $qj(t)$ [7 – 8].

B. Authors and Affiliations

Dr Akash Singh is working with IBM Corporation as an IT Architect and has been designing Mission Critical System and Service Solutions; He has published papers in IEEE and other International Conferences and Journals.

He joined IBM in Jul 2003 as a IT Architect which conducts research and design of High Performance Smart Grid Services and Systems and design mission critical architecture for High Performance Computing Platform and Computational Intelligence and High Speed Communication systems. He is a Senior Member of IEEE (Institute for Electrical and Electronics Engineers), the AAAI (Association for the Advancement of Artificial Intelligence) and the AACR (American Association for Cancer Research). He is the recipient of numerous awards from World Congress in Computer Science, Computer Engineering and Applied Computing 2010, 2011, and IP Multimedia System 2008 and Billing and Roaming 2008. He is active research in the field of Artificial Intelligence and advancement in Medical Systems. He is in Industry for 18 Years where he performed various role to provide the Leadership in Information Technology and Cutting edge Technology.

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